Introduction

The ACL 2018 Workshop on Representation Learning for NLP (RepL4NLP) takes place on Friday, July 20, 2018 in Melbourne, Australia, immediately following the 56th Annual Meeting of the Association for Computational Linguistics (ACL). The workshop is generously sponsored by Facebook, Salesforce, ASAPP, DeepMind, Microsoft Research, and Naver.

Repl4NLP is organised by Isabelle Augenstein, Kris Cao, He He, Felix Hill, Spandana Gella, Jamie Kiros, Hongyuan Mei and Dipendra Misra, and advised by Kyunghyun Cho, Edward Grefenstette, Karl Moritz Hermann and Laura Rimell.

The 3rd Workshop on Representation Learning for NLP aims to continue the success of the 1st Workshop on Representation Learning for NLP, which received about 50 submissions and over 250 attendees and was the second most attended collocated event at ACL 2016 in Berlin, Germany after WMT; and the 2nd Workshop on Representation Learning for NLP at ACL 2017 in Vancouver, Canada.

The workshop has a focus on vector space models of meaning, compositionality, and the application of deep neural networks and spectral methods to NLP. It provides a forum for discussing recent advances on these topics, as well as future research directions in linguistically motivated vector-based models in NLP.
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Chris Quirk, Microsoft Research
Lyle Ungar, University of Pennsylvania
Eva Maria Vecchi, University of Cambridge
Dirk Weissenborn, German Research Center for AI
Tsung-Hsien Wen, University of Cambridge
Yi Yang, Bloomberg LP
Helen Yannakoudakis, University of Cambridge

Invited Speaker:

Yejin Choi, University of Washington
Trevor Cohn, University of Melbourne
Margaret Mitchell, Google Research
Yoav Goldberg, Bar Ilan University
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09:45–14:45  Keynote Session

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Yejin Choi

10:30–11:00  Coffee Break

11:00–11:45  Invited Talk 2
Trevor Cohn

11:45–12:30  Invited Talk 3
Margaret Mitchell

12:30–14:00  Lunch

14:00–14:45  Invited Talk 4
Yoav Goldberg

14:45–15:00  Outstanding Papers Spotlight Presentations
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16:30–17:30  Panel Discussion

17:30–17:40  Closing Remarks + Best Paper Awards Announcement
Corpus specificity in LSA and word2vec: the role of out-of-domain documents

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Abstract

Despite the popularity of word embeddings, the precise way by which they acquire semantic relations between words remain unclear. In the present article, we investigate whether LSA and word2vec capacity to identify relevant semantic relations increases with corpus size. One intuitive hypothesis is that the capacity to identify relevant associations should increase as the amount of data increases. However, if corpus size grows in topics which are not specific to the domain of interest, signal to noise ratio may weaken. Here we investigate the effect of corpus specificity and size in word-embeddings, and for this, we study two ways for progressive elimination of documents: the elimination of random documents vs. the elimination of documents unrelated to a specific task. We show that word2vec can take advantage of all the documents, obtaining its best performance when it is trained with the whole corpus. On the contrary, the specialization (removal of out-of-domain documents) of the training corpus, accompanied by a decrease of dimensionality, can increase LSA word-representation quality while speeding up the processing time. From a cognitive-modeling point of view, we point out that LSA’s word-knowledge acquisitions may not be efficiently exploiting higher-order co-occurrences and global relations, whereas word2vec does.

1 Introduction

The main idea behind corpus-based semantic representation is that words with similar meanings tend to occur in similar contexts (Harris, 1954). This proposition is called distributional hypothesis and provides a practical framework to understand and compute the semantic relationship between words. Based in the distributional hypothesis, Latent Semantic Analysis (LSA) (Deerwester et al., 1990; Landauer and Dumais, 1997; Hu et al., 2007) and word2vec (Mikolov et al., 2013a,b), are one of the most important methods for word meaning representation, which describes each word in a vectorial space, where words with similar meanings are located close to each other.

Word embeddings have been applied in a wide variety of areas such as information retrieval (Deerwester et al., 1990), psychiatry (Altszyler et al., 2018; Carrillo et al., 2018), treatment optimization (Corcoran et al., 2018), literature (Diuk et al., 2012) and cognitive sciences (Landauer and Dumais, 1997; Denhière and Lemaire, 2004; Lemaire and Denhi, 2004; Diuk et al., 2012).

LSA takes as input a training Corpus formed by a collection of documents. Then a word by document co-occurrence matrix is constructed, which contains the distribution of occurrence of the different words along the documents. Then, usually, a mathematical transformation is applied to reduce the weight of uninformative high-frequency words in the words-documents matrix (Dumais, 1991). Finally, a linear dimensionality reduction is implemented by a truncated Singular Value Decomposition, SVD, which projects every word in a subspace of a predefined number of dimensions, $k$. The success of LSA in capturing the latent meaning of words comes from this low-dimensional mapping. This representation improvement can be explained as a consequence of the elimination of the noisiest dimensions (Turney and Pantel, 2010).

Word2vec consists of two neural network models, Continuous Bag of Words (CBOW) and Skip-gram. To train the models, a sliding window is
moved along the corpus. In the CBOW scheme, in each step, the neural network is trained to predict the center word (the word in the center of the window based) given the context words (the other words in the window). While in the skip-gram scheme, the model is trained to predict the context words based on the central word. In the present paper, we use the skip-gram, which has produced better performance in Mikolov et al. (2013b).

Despite the development of new word representation methods, LSA is still intensively used and has been shown that produce better performances than word2vec methods in small to medium size training corpus (Altszyler et al., 2017).

1.1 Training Corpus Size and Specificity in Word-embeddings

Over the last years, great effort has been devoted to understanding how to choose the right parameter settings for different tasks (Quesada, 2011; Dumais, 2003; Landauer and Dumais, 1997; Lapesa and Evert, 2014; Bradford, 2008; Nakov et al., 2003; Baroni et al., 2014). However, considerably lesser attention has been given to study how different corpus used as input for training may affect the performance. Here we ask a simple question on the property of the corpus: is there a monotonic relation between corpus size and the performance? More precisely, what happens if the topic of additional documents differs from the topics in the specific task? Previous studies have surprisingly shown some contradictory results on this simple question.

On the one hand, in the foundational work, Landauer and Dumais (1997) compare the word-knowledge acquisition between LSA and that of children’s. This acquisition process may be produced by 1) direct learning, enhancing the incorporation of new words by reading texts that explicitly contain them; or 2) indirect learning, enhancing the incorporation of new words by reading texts that do not contain them. To do that, they evaluate LSA semantic representation trained with different size corpus in multiple-choice synonym questions extracted from the TOEFL exam. This test consists of 80 multiple-choice questions, in which its requested to identify the synonym of a word between 4 options. In order to train the LSA, Landauer and Dumais used the TASA corpus (Zeno et al., 1995).

Landauer and Dumais (1997) randomly replaced exam-words in the corpus with non-sense words and varied the number of corpus’ documents selecting nested sub-samples of the total corpus. They concluded that LSA improves its performance on the exam both when training with documents with exam-words and without them. However, as could be expected, they observed a greater effect when training with exam-words. It is worth mentioning that the replacement of exam-words with non-sense words may create incorrect documents, thus, making the algorithm acquire word-knowledge from documents which should have an exam-word but do not. In the Results section, we will study this indirect word acquisition in the TOEFL test without using non-sense words.

Along the same line, Lemaire and Denhiere (2006) studied the effect of high-order co-occurrences in LSA semantic similarity, which goes further in the study of Landauer’s indirect word acquisition.

In their work, Lemaire and Denhiere (2006) measure how the similarity between 28 pairs of words (such as bee/honey and buy/shop) changes when a 400-dimensions LSA is trained with a growing number of paragraphs. Furthermore, they identify for this task the marginal contribution of the first, second and third order of co-occurrence as the number of paragraphs is increased. In this experiment, they found that not only does the first order of co-occurrence contribute to the semantic closeness of the word pairs, but also the second and the third order promote an increment on pairs similarity. It is worth noting that Landauer’s indirect word acquisition can be understood in terms of paragraphs without either of the words in a pair, and containing a third or more order co-occurrence link.

So, the conclusion from Lemaire and Denhiere (2006) and Landauer and Dumais (1997) studies suggest that increasing corpus size results in a gain, even if this increase is in topics which are unrelated for the relevant semantic directions which are pertinent for the task.

However, a different conclusion seems to result from other sets of studies. Stone et al. (2006) have studied the effect of Corpus size and specificity in a document similarity rating task. They found that training LSA with smaller subcorpus selected for the specific task domain maintains or even improves LSA performance. This corresponds to the intuition of noise filtering, when removing infor-
mation from irrelevant dimensions results in improvements of performance.

In addition, Olde et al. (2002) have studied the effect of selecting specific subcorpus in an automatic exam evaluation task. They created several subcorpora from a Physics corpus, progressively discarding documents unrelated to the specific questions. Their results showed small differences in the performance between the LSA trained with original corpus and the LSA trained with the more specific subcorpus.

It is well known that the number of LSA dimensions ($k$) is a key parameter to be duly adjusted in order to eliminate the noisiest dimensions (Landauer and Dumais, 1997; Turney and Pantel, 2010). Excessively high $k$ values may not eliminate enough noisy dimensions, while excessively low $k$ values may not have enough dimensions to generate a proper representation. In this context, we hypothesize that when out-of-domain documents are discarded, the number of dimensions needed to represent the data should be lower, thus, $k$ must be decreased.

Regarding word2vec, Cardellino and Alemany (2017) and Dusserre and Padró (2017) have shown that word2vec trained with a specific corpus can produce better performance in semantic tasks than when it is trained with a bigger and general corpus. Despite these works point out the relevance of domain-specific corpora, they do not study the specificity in isolation, as they compare corpus from different sources.

In this article, we set to investigate the effect of the specificity and size of training corpus in word-embeddings, and how this interacts with the number of dimensions. To measure the semantic representations quality we have used two different tasks: the TOEFL exam, and a categorization test. The corpus evaluation method consists in the comparison between two ways of progressive elimination of documents: the elimination of random documents vs the elimination of out-of-domain documents (unrelated to the specific task). In addition, we have varied $k$ within a wide range of values.

As we show, LSA’s dimensionality plays a key role in the LSA representation when the corpus analysis is made. In particular, we observe that both, discarding out-of-domain documents and decreasing the number of dimensions produces an increase in the algorithm performance. In one of the two tasks, discarding out-of-domain docu-

ments without the decrease of $k$ results in the complete opposite behavior, showing a strong performance reduction. On the other hand, word2vec shows in all cases a performance reduction when discarding out-of-domain, which suggests an exploitation of higher-order word co-occurrences.

Our contribution in understanding the effect of out-of-domain documents in word-embeddings knowledge acquisitions is valuable from two different perspectives:

- From an operational point of view: we show that LSA’s performance can be enhanced when: (1) its training corpus is clean from out-of-domain documents, and (2) a reduction of LSA’s dimensions number is applied. Furthermore, the reduction of both the corpus size and the number of dimensions tend to speed up the processing time. On the other hand, word2vec can take advantage of all the documents, obtaining its best performance when it is trained with the whole corpus.

- From a cognitive modeling point of view: we point out that LSA’s word-knowledge acquisition does not take advantage of indirect learning, while word2vec does. This throws light upon models capabilities and limitations in modeling human cognitive tasks, such as: human word-learning (Landauer and Dumais, 1997; Lemaire and Denhière, 2006; Landauer, 2007), semantic memory (Denhière and Lemaire, 2004; Kintsch and Mangalath, 2011; Landauer, 2007) and words classification (Laham, 1997).

2 Methods

We used TASA corpus (Zeno et al., 1995) in all experiments. TASA is a commonly used linguistic corpus consisting of more than 37 thousand educational texts from USA K12 curriculum. We word-tokenized each document, discarding punctuation marks, numbers, and symbols. Then, we transformed each word to lowercase and eliminated stopwords, using the stoplist in NLTK Python package (Bird et al., 2009). TASA corpus contains more than 5 million words in its cleaned version.

In each experiment, the training corpus size was changed by discarding documents in two different ways:

- Random documents discarding: The desired number of documents ($n$) contained in the
subcorpus is preselected. Then, documents are randomly eliminated from the original corpus until there are exactly \( n \) documents. If any of the test words (i.e. words that appear in the specific task) do not appear at least once in the remaining corpus, one document is randomly replaced with one of the discarded documents that contains the missing word.

- **Out-of-domain documents discarding**: The desired number of documents \( (n) \) contained in the subcorpus is preselected. Then, only documents with no test words are eliminated from the original corpus until there are exactly \( n \) documents. Here, \( n \) must be greater than or equal to the number of documents that contain at least one of the test words.

Both, LSA and Skip-gram word-embeddings were generated with Gensim Python library (Rehůrek and Sojka, 2010). In LSA implementation, a Log-Entropy transformation was applied before the truncated Singular Value Decomposition. In Skip-gram implementation, we discarded tokens with frequency higher than \( 10^{-3} \), and we set the window size and negative sampling parameters to 15 (which were found to be maximal in two semantic tasks over TASA corpus (Altszyler et al., 2017)). In all cases, word-embeddings dimensions values were varied to study its dependency.

The semantic similarity \( (S) \) of two words was calculated using the cosine similarity measure between their respective vectorial representation \( (v_1, v_2) \).

\[
S(v_1, v_2) = \cos(v_1, v_2) = \frac{v_1 \cdot v_2}{\|v_1\| \cdot \|v_2\|}
\]  

(1)

The semantic distances between two words \( d(v_1, v_2) \) is calculated as 1 minus the semantic similarity \( (d(v_1, v_2) = 1 - S(v_1, v_2)) \).

Word-embeddings knowledge acquisition was tested in two different tasks: a semantic categorization test and the TOEFL test.

### 2.1 Semantic categorization test

In this test we measured the capabilities of the model to represent the semantic categories used by Patel et al. (1997) (such as drinks, countries, tools and clothes). The test is composed of 53 categories with 10 words each. In order to measure how well the word \( i \) is grouped vis-à-vis the other words in its semantic category we used the Silhouette Coefficients, \( s(i) \) (Rousseeuw, 1987),

\[
s(i) = \frac{b(i) - a(i)}{\max\{a(i), b(i)\}}
\]

(2)

where \( a(i) \) is the mean distance of word \( i \) with all other words within the same category, and \( b(i) \) is the minimum mean distance of word \( i \) to any words within another category (i.e. the mean distance to the neighboring category). In other words, Silhouette Coefficients measure how close is a word to its own category words compared to the closeness to neighboring words. The Silhouette Score is computed as the mean value of all Silhouette Coefficients. The score takes values between -1 and 1, higher values reporting localized categories with larger distances between categories, representing better clustering.

The high number of test words (530) and the high frequency of some of them leaves only a few documents with no test words. This makes varied corpus size range in the out-of-domain documents discarding very small. To avoid this, we tested only on the 10 least frequent categories. The frequency of a question is measured as the number of documents in which at least one word from this category appears.

### 2.2 TOEFL test

The TOEFL test was introduced by Landauer and Dumais (1997) to evaluate the quality of semantic representations. This test consists of 80 multiple-choice questions, in which it is requested to identify the synonym of a target word between 4 options. For example: select the most semantically similar to “enormously” between this words: “tremendously”, “appropriately”, “uniquely” and “decidedly”. The performance of this test was measured by the percentage of correct responses.

Again, The high number of test words (400) and the high frequency of some of them leaves few documents with no test words. So we performed the test only on the 20 least frequent questions in order to have out-of-domain documents to discard.

### 3 Results

#### 3.1 Semantic categorization Test

In Figure 1 we show the LSA (top panel) and word2vec (bottom panel) categorization performance with both documents discarding methods.
For each corpus size and document discarding method, we took 10 subcorpus samples (in total we consider 90 subcorpus + the complete corpus). In each corpus/subcorpus, we trained LSA and word2vec with a wide range of dimension values, using in each case the dimension that produces the best mean performance.

In both cases, performance decreases when documents are randomly discarded (orange dashed lines). However, LSA and word2vec have different behavior in the out-of-domain document discarding method (blue solid lines). While LSA produces better scores with increasing specificity, the word2vec performance decreases in the same situation.

LSA’s maximum performance is obtained using 20 dimensions and removing all out-of-domain documents in the training corpus. While, when all the corpus is used the best number of dimensions is 100. These results show that performance for a specific task may be increased by “cleaning” the training corpus of out-of-domain documents. But, in order to enhance the performance, the elimination of out-of-domain documents should be accompanied by a decrease of the number of LSA dimensions. For example, fixing the number of dimensions to 100 the performance result in a reduction of 55%. We also point out that this technical subtlety has not been taken into account in previous results that reported the presence of indirect learning in LSA (Landauer and Dumais, 1997; Lemaire and Denhiere, 2006).

Figure 2 shows the results disaggregated by number of dimensions. It can be seen that in all cases the performance decreases when documents are randomly discarded (bottom panels). However, in the case of LSA, the dependency with the number of out-of-domain documents varied with the number of dimensions (top left panel). In the cases of 300, 500 and 1000 dimensions, the performance decreases when out-of-domain documents are eliminated. In contrast, we obtain the opposite behavior in the cases of 5, 10, 20 dimensions, in which the elimination of out-of-domain documents increases LSA’s categorization performance.

Consider the case when $k$ is fixed in the value that maximizes the performance with the entire corpus (around $k = 100$). When the corpus is “cleaned” of out-of-domain documents, the remaining corpus will have not only fewer documents, but also less topic diversity among texts. Thus, the number of dimensions ($k$) needed to generate a proper semantic representation should be reduced. As $k$ is fixed in high values, LSA may not eliminate enough noisy dimensions, leading to a decrease in the performance. This effect becomes
Figure 2: Semantic categorization test analysis disaggregated by number of dimensions. Categorization performance (Silhouette Score) vs corpus size, by number of dimensions. Both document variation methods are shown: out-of-domain documents discarding (top panels) and random document discarding (bottom panels). The shown scores values and their error bars are, respectively, the mean values and the standard error of the mean of 10 samples.

larger when the selected $k$ is high, as it can be seen for $k = 300, 500, 1000$. On the other hand, consider the case when $k$ is fixed in the value that maximizes the performance with the “cleaned” corpus (around $k = 20$). The presence of out-of-domain documents in the complete corpus increase the topic diversity. As $k$ is fixed in low values, the LSA will not have enough dimensions to represent all the intrinsic complexity of the whole corpus. So, when the corpus is “cleaned” of out-of-domain documents, the performance should increase.

On the other hand, in the case of word2vec, the performance decrease, with almost all dimension values, when out-of-domain documents are eliminated. Moreover, the discarding of out-of-domain documents do not require a considerable decrease of the number of dimensions. These findings supports the idea that individual dimensions of word2vec do not encode latent semantic domains, however, more analysis must be done in these direction (see Baroni et al. (2014) discussion).

3.2 TOEFL Test

In Figure 3 we show the TOEFL correct answer fraction vs the corpus size. We varied the corpus size by both methods: the out-of-domain documents discarding and the Random document discarding. As in the categorization test procedure, a wide range of dimension values where tested, using in each case the dimension that produces the best mean performance.

In both models, performance decreases when documents are randomly discarded (orange dashed lines in figure 3). For LSA, the elimination of out-of-domain documents does not produce a significant performance variation, which shows that LSA can not take advantage of out-of-domain document. These results are in contradiction with Landauer and Dumais (1997) observation of indirect
learning. We believe that this difference is due to the lack of adjustment in the number of dimensions. On the other hand, word2vec has the same behaviour as in the categorization test. The performance when the out-of-domain documents are discarded show a small downward trend (not significant, with \( p\text{-val}=0.31 \) in a two-sided Kolmogorov-Smirnov test), but not as pronounced as in random document discarding (orange dashed lines) and the out-of-domain documents discarding (blue solid lines). The shown Silhouette Score values and their error bars are, respectively, the mean values and the standard error of the mean of 10 samples. The dimension was varied among \{5, 10, 20, 50, 100, 300, 500, 1000\} for LSA and among \{5, 10, 20, 50, 100, 300, 500\} for word2vec. Due to the high computational effort, in the case of word2vec we avoid using 1000 dimensions.

4 Conclusion and Discussion

Despite the popularity of word-embeddings in several semantic representation task, the way by which they acquire new semantic relations between words is unclear. In particular, for the case of LSA there are two opposite visions about the effect of incorporating out-of-domain documents. From one point of view, training LSA with a specific subcorpus, cleaned of documents unrelated to the specific task increases the performance (Stone et al., 2006). From the other point of view, the presence of unrelated documents improves the representations. The second view point is supported by the conception that the SVD in LSA can capture high-order co-occurrence words relations (Landauer and Dumais, 1997; Lemaire and Denhiere, 2006; Turney and Pantel, 2010). Based on this, LSA is used as a plausible model of human semantic memory given that it can capture indirect relations (high-order word co-occurrences).

In the present article we studied the effect of out-of-domain documents in LSA and word2vec semantic representations construction. We compared two ways of progressive elimination of documents: the elimination of random documents vs the elimination of out-of-domain documents. The semantic representations quality was measured in two different tasks: a semantic categorization test and a TOEFL exam. Additionally, we have varied a large range of word-embedding dimensions (\( k \)). We have shown that word2vec can take advantage of all the documents, obtaining its best performance when it is trained with the whole corpus. On the contrary, LSA’s word-representation quality increases with a specialization of the training corpus (removal of out-of-domain document) accompanied by a decrease of \( k \). Furthermore, we have shown that the specialization without the decrease of \( k \) can produce a strong performance reduction. Thus, we point out the need to vary \( k \) when the corpus size dependency is studied. From a cognitive modeling point of view, we
Figure 4: TOEFL test analysis disaggregated by number of dimensions. Correct answer percentage vs corpus size, by number of dimensions. Both document variation methods are shown: out-of-domain documents discarding (top panels) and random document discarding (bottom panels). The shown scores values and their error bars are, respectively, the mean values and the standard error of the mean of 10 samples.

point out that LSA’s word-knowledge acquisitions does not take advantage of indirect learning (high-order word co-occurrences), while word2vec does. This throws light upon word-embeddings capabilities and limitations in modeling human cognitive tasks, such as: human word-learning (Landauer and Dumais, 1997; Lemaire and Denhardi, 2006; Landauer, 2007), semantic memory (Denhière and Lemaire, 2004; Kintsch and Mangalath, 2011; Landauer, 2007) and words classification (Laham, 1997).

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Hierarchical Convolutional Attention Networks for Text Classification

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Abstract

Recent work in machine translation has demonstrated that self-attention mechanisms can be used in place of recurrent neural networks to increase training speed without sacrificing model accuracy. We propose combining this approach with the benefits of convolutional filters and a hierarchical structure to create a document classification model that is both highly accurate and fast to train – we name our method Hierarchical Convolutional Attention Networks. We demonstrate the effectiveness of this architecture by surpassing the accuracy of the current state-of-the-art on several classification tasks while being twice as fast to train.

1 Introduction

Text classification is an important research area in natural language processing (NLP). Traditional text classification approaches utilize features generated from vector space models such as bag-of-words or term frequency-inverse document frequency (TF-IDF) (Sebastiani, 2005). More recently, deep learning approaches have been shown to outperform traditional approaches based on vector space models (Zhang et al., 2015; Tang et al., 2014). These newer deep learning approaches typically rely on architectures based off convolutional neural networks (CNNs) or recurrent neural networks (RNNs) (Young et al., 2017).

RNNs, which are designed to learn patterns over sequential data, have been successfully applied towards various NLP tasks (Liu et al., 2016; Irsoy and Cardie, 2014; Cho et al., 2014). In NLP, RNNs typically process one word at a time and learn features based on complex sequences of words. While RNNs are capable of capturing linguistic patterns useful for NLP tasks, especially over long segments of text, they can be slow to train compared to other deep learning architectures – in order to calculate the gradients associated with any given word in a sequence, an RNN must backpropagate through all previous words in that sequence, resulting in backpropogation functions far more complex than those in feedforward or convolutional architectures.

CNNs, traditionally used for computer vision, have also been applied to NLP tasks with notable success (Zeng et al., 2014; Dos Santos and Gatti, 2014; Wang et al., 2012). Unlike RNNs, which learn patterns across an entire sequence of text, CNNs use a sliding window that examines only a few words/characters at a time. Thus, CNNs learn features based on the most salient combinations of X words/characters where X is determined by the window size used; unlike RNNs, CNNs are less capable of capturing linguistic features across long distances. Despite this shortcoming, CNNs can often be as effective as RNNs in many basic NLP tasks (Yin et al., 2017). Furthermore, CNNs are generally faster to train than RNNs.

In this paper, we introduce Hierarchical Convolutional Attention Networks (HCANs), an architecture based off self-attention that can capture linguistic relationships over long sequences like RNNs while still being fast to train like CNNs. HCANs can achieve accuracy that surpasses the current state-of-the-art on several classification tasks while being twice as fast to train.

2 Related Work

In 2014, Kim (2014) proposed one of the first CNNs for text classification. Kim’s CNN used three parallel convolutional layers; these process a sentence using a sliding window that examines three, four, and five words at a time. The three
convolutions then feed into a maxpool across the entire sentence, which selects the most potent features in each convolution and concatenates them into a single feature vector. Finally, the selected features are fed into a dense softmax layer for classification. Due to its simplicity and strong performance in many tasks, Kim’s CNN architecture is still commonly used today in many text classification tasks (Qiu et al., 2017; Gehrmann et al., 2017).

One shortcoming of Kim’s CNN approach is that it cannot find linguistic patterns beyond a fixed window size, which may harm performance for complex NLP tasks. Attempts have been made to mitigate this issue by increasing the CNN depth and using local maxpooling to increase the receptive field (Conneau et al., 2017). However, Le et al. (2017) showed that increasing CNN depth helps the performance of character-level CNNs but not word-level CNNs. They further demonstrated that a shallow word-level CNN similar to Kim’s proposed structure can outperform much deeper and more complex CNN architectures on a wide range of text classification tasks.

The current state-of-the-art in text classification are Hierarchical Attention Networks (HANs), developed by Yang et al. (2016). Whereas the previous approaches mentioned are all based on CNNs, HANs utilize RNNs. HANs use a hierarchical structure in which each hierarchy uses the same architecture – a bidirectional RNN with gated recurrent units (GRUs) (Chung et al., 2014), followed by an attention mechanism that creates a weighted sum of the RNN outputs at each timestep. The HAN processes documents by first breaking a long document into its sentence components, then processing each sentence individually before processing the entire document. By breaking a document into smaller, more manageable chunks, the HAN can better locate and extract critical information useful for classification. This approach surpassed the performance of all previous approaches across several text classification tasks. However, compared to CNN-based approaches, HANs are much slower to train because they utilize RNNs.

In 2017, Vaswani et al. (2017) created a deep learning model for machine translation based entirely on self-attention mechanisms (Cheng et al., 2016; Lin et al., 2017; Paulus et al., 2017). Traditionally, CNN or RNN layers are used to extract meaningful features from words or images; attention is applied afterwards to the output of the CNN or RNN layers to help the network focus on features that are most salient (Luong et al., 2015; Xu et al., 2015; Hermann et al., 2015). However, Vaswani showed that self-attention could be applied directly on raw word embeddings to extract important relations and apply meaningful transformations on words. Like RNNs, this attention-based approach can capture relationships over long distances; unlike RNNs, this approach utilizes a feedforward architecture and is much faster to train. Vaswani achieved state-of-the-art results in machine translation using 10x-100x fewer trainable parameters than previous state-of-the-art models.

We hypothesize that a similar self-attention-based architecture can achieve both fast and accurate performance in text classification tasks. In the following section, we show how we adapt the attention-based architecture developed by Vaswani for machine translation into an effective approach for text classification.

3 Hierarchical Convolutional Attention Network

The overall structure of our HCAN is shown in Figure 1. The components and structure of our HCAN are described in greater detail in the following subsections.

3.1 Scaled Dot Product Attention

Suppose we have a sequence of word embeddings $E^{\text{input}} \in \mathbb{R}^{l \times d}$, where $l$ is the length of the sequence, $d$ is the embedding dimension, and $e_i^{\text{input}}$ is the $i$-th word embedding in the sequence.

Self-attention, sometimes referred to as intra-attention, compares each entry $e_i^{\text{input}}$ to every entry $e_j^{\text{input}}$ in that same sequence; this allows for the discovery of relationships between entries in the sequence. Self-attention outputs a new sequence $E^{\text{output}} \in \mathbb{R}^{l \times d}$ in which each entry $e_i^{\text{output}}$ is a weighted average of all entries $e_j^{\text{input}}$ in the input sequence. Each entry $e_i^{\text{output}}$ should contain within it the most pertinent information to that entry from all entries in the input sequence $e_j^{\text{input}}$.

$$\text{Attention}(Q, K, V) = \text{softmax}(\frac{QK^T}{\sqrt{d}})V$$ (1)

Scaled-dot-product attention (Figure 2, Equation 1) is a type of self-attention developed by Vaswani et al. that was shown to work well
in machine translation. Scaled-dot-product attention utilizes three word embedding matrices – the ‘query’ embeddings $Q \in \mathbb{R}^{l \times d}$, the ‘key’ embeddings $K \in \mathbb{R}^{l \times d}$, and the ‘value’ embeddings $V \in \mathbb{R}^{l \times d}$.

In the most basic implementation of self-attention, $Q$, $K$, and $V$ can all be substituted by the same sequence of word embeddings $E^{\text{input}} \in \mathbb{R}^{l \times d}$. $Q$ and $K$ are used to create a weight matrix $QK^T$ based on the similarity of entries in the sequence. Vaswani et al. found that scaling this weight matrix by a factor of $\sqrt{d}$ yields better performance for higher-dimensional word embeddings. Once this weight matrix is scaled and normalized, it is multiplied with $V$ to create a new output sequence $E^{\text{output}} \in \mathbb{R}^{l \times d}$ in which each entry $e_i^{\text{output}}$ is a weighted average of all entries $e_i^{\text{input}}$ in the input sequence.

### 3.2 Convolutional Feature Extraction

Rather than use the same $E^{\text{input}}$ for $Q$, $K$, and $V$, we can use a function to extract different features from $E^{\text{input}}$ for each of the $Q$, $K$, and $V$ embeddings. This allows for more expressive comparison between entries in a sequence; for example, certain features may be useful when comparing $Q$ and $K$ but may not be necessary when creating the output sequence from $V$. For our feature extractor function, we use a 1D convolution with $d$ filter maps and a window size of three words, which provides more context for each center word when extracting important features (Equation 2).

$$Q = \text{ELU}(\text{Conv1D}(E, W^q) + b^q)$$

$$K = \text{ELU}(\text{Conv1D}(E, W^k) + b^k)$$

$$V = \text{ELU}(\text{Conv1D}(E, W^v) + b^v)$$

In the equation above, Conv1D$(A, B)$ is a 1D convolution operation with $A$ as the input as $B$ as the filter, $\{Q, K, V, E\} \in \mathbb{R}^{l \times d}$, $\{W^q, W^k, W^v\} \in \mathbb{R}^{d \times d}$, and $\{b^q, b^k, b^v\} \in \mathbb{R}^d$.

We found exponential linear units (ELUs) (Clevert et al., 2016) to perform better than rectified linear units (ReLUs) and other activation functions. Unlike ReLUs, ELUs can output negative values, which allows for more complex interactions between the $Q$ and $K$ embeddings when calculating word weights – each word can be assigned a large range of both positive and negative values before being sent into the softmax function.

### 3.3 Convolutional Multihead Self-Attention

For each entry in the output sequence, scaled dot product attention calculates a set of weights that is used to create a weighted average; the same weights are applied across all $d$ dimensions of the $V$ embeddings. To expand the capabilities of scaled dot product attention, Vaswani et al. introduced multihead attention. Rather than using a single attention function across all $d$ dimensions of the embeddings, multihead attention uses $h$ parallel attention functions, each of which attends to a different portion of the embedding dimension. This allows different portions of the embeddings to be combined using different weights so that the final output sequence can be constructed from a more expressive combination. Vaswani demonstrated that multihead attention performs better than regular scaled dot product attention for machine translation.

$$\text{MultiHead}(Q, K, V) = [\text{head}_1, ..., \text{head}_h]$$

where $\text{head}_i = \text{Attention}(Q_i, K_i, V_i)$ (3)
Our implementation convolutional multihead self-attention (Figure 2, Equation 3) is based on the multihead attention developed by Vaswani. After using convolution to generate the \( Q, K, \) and \( V \) embeddings, we split each of the \( Q, K, \) and \( V \) embeddings into \( h \) sub-embeddings such that \( \{Q_i, K_i, V_i\} \in \mathbb{R}^{l \times d/h} \). Each triplet of sub-embeddings is then fed into their own scaled dot product attention function. The final output is the concatenation of the outputs \( head_i \in \mathbb{R}^{l \times d/h} \) from the individual scaled dot product attention functions to form an output sequence \( E^{output} \in \mathbb{R}^{l \times d} \).

### 3.4 Capturing Complex Word Relationships

In general, attention mechanisms are designed to produce a weighted average of an input sequence. Unfortunately, when trying to capture the overall content within a linguistic sequence, a weighted average may not be sufficient. Two examples of this are negation and scaling. In negation, a word sequence such as ‘was not the case’ may reverse the meaning of words in another part the sequence (i.e. multiply those embeddings by \(-1\)). Similarly in scaling, a word sequence such as ‘to a high degree’ may increase the polarity of another part of the sequence (i.e. multiply those embeddings by some positive value). Attention mechanisms, which only create weighted averages, are not designed to capture these interactions.

To better capture these types of linguistic interactions, we test the effectiveness of using two convolutional multihead self-attentions in parallel and performing elementwise multiplication on the outputs (Figure 1, Equation 4). This allows the network to capture more complex interactions between elements in the sequence and expands the expressiveness of the final output beyond that of a simple weighted average.

\[
\text{Parallel}(E) = \text{MultiHead}(Q^a, K^a, V^a) \\
\circ \text{MultiHead}(Q^b, K^b, V^b)
\]

where \( Q^a = \text{ELU} (\text{Conv1D}(E, W^{qa}) + b^{qa}) \)  
\( K^a = \text{ELU} (\text{Conv1D}(E, W^{ka}) + b^{ka}) \)  
\( V^a = \text{ELU} (\text{Conv1D}(E, W^{va}) + b^{va}) \)  
\( Q^b = \text{ELU} (\text{Conv1D}(E, W^{qb}) + b^{qb}) \)  
\( K^b = \text{ELU} (\text{Conv1D}(E, W^{kb}) + b^{kb}) \)  
\( V^b = \tanh (\text{Conv1D}(E, W^{vb}) + b^{vb}) \)  

(4)

Because \( \tanh \) outputs a value between -1 and 1, it is used to generate the \( V \) embeddings for the second self-attention to prevent the final output from becoming too small or large after multiplying the outputs from the two self-attention mechanisms.

### 3.5 Convolutional Multihead Target-Attention

The output of our convolutional multihead self-attention is a new output sequence \( E^{output} \in \mathbb{R}^{l \times d} \) in which \( l \) is based on the length of the input sequence. For classification purposes, we require that each sequence regardless of its length be represented by a single fixed-length vector \( V \in \mathbb{R}^{1 \times d} \). We therefore introduce convolutional multihead target-attention, which utilizes the concepts from multihead convolutional self-attention but...
operates like the traditional attention mechanism that is used on the outputs of a RNN.

\[
\text{Target}(E) = \text{MultiHead}(T, K, V)
\]

where \[ K = \text{ELU}(\text{Conv1D}(E, W^k) + b^k) \] (5)

\[ V = \text{ELU}(\text{Conv1D}(E, W^v) + b^v) \]

In convolutional multihed target-attention (Figure 2, Equation 5), instead of comparing the entries in a sequence \[ E^{\text{input}} \in \mathbb{R}^{1 \times d} \] to each other, we compare them to a learnable target vector \[ T \in \mathbb{R}^{1 \times d} \] that represents the most critical information to look for given a specific task – the content in this vector is learned through backpropagation based on the task at hand. The output is a single weighted average \[ E^{\text{output}} \in \mathbb{R}^{1 \times d} \] that captures the most critical content across the sequence. This final output vector may then be fed into a softmax and used for classification purposes.

3.6 Positional Embeddings

RNN-based approaches for text processing can inherently account for word order when extracting features. However, feedforward and convolution-based approaches such as our implementation of convolutional multihed self-attention do not have this capability. One way to address this problem is by adding positional embeddings \[ P \in \mathbb{R}^{1 \times d} \] (Gehring et al., 2017; dos Santos et al., 2015). Positional embeddings are vector representations of the absolute position of an entry in a sequence. These are added directly to each word/sentence embedding in the input sequence before the sequence is fed into the convolutional multihed self-attention. We use randomly initialized embeddings that are learned during training; we found that these provide a slight boost toward classification accuracy.

3.7 Hierarchical Structure

In their work on HANs, Yang et al. attained state-of-the-art performance by utilizing a hierarchical structure that first breaks up documents into sentences. The lower hierarchy reads in word embeddings from a given sentence and outputs a sentence embedding representing the content within that sentence, and the upper hierarchy reads the sentences embeddings created from the lower hierarchy and outputs a document embedding representing the content of the entire document; this document embedding is then used for classification. In our experiments, we test the effectiveness of our HCAN with and without this hierarchical structure. We expect that, like with RNNs, self-attention has difficulty capturing meaningful semantic relationships over very long sequences with too many entries; using a hierarchical structure to break down long sequences into more manageable chunks mitigates this issue.

Each hierarchy in our HCAN consists of two parallel convolutional multihed self-attentions followed by a convolutional multihed target attention (Figure 1). Positional embeddings are added to the inputs of each hierarchy to allow the network to identify relationships based on word/sentence positions. We tried increasing the depth within each hierarchy by using multiple layers of self-attentions but found that this did not improve model accuracy.

3.8 Regularization

To regularize our network, we apply dropout of 0.1 on the normalized attention weights (produced by scaling \[ QK^T \] by \[ \sqrt{d} \] and then applying softmax) within every scaled dot product attention. Furthermore, we apply dropout of 0.1 on the word and sentence embeddings after the positional embeddings have been added, which has been shown to be effective in other NLP applications (Peng et al., 2015).

We also apply layer normalization (Ba et al., 2016) after the elementwise multiplication of the two parallel convolutional multihed self-attentions (Figure 1). This not only applies a regularization effect, but also speeds up the rate of convergence. Layer normalization is used instead of batch normalization because layer normalization is still effective with very small batch sizes.

4 Experiments

4.1 Datasets

We evaluate the performance of the HCAN on four classification tasks using three datasets (Table 1).

The Yelp reviews dataset \(^1\) consists of over 4.7 million Yelp reviews of various businesses collected over 12 metropolitan areas. For our task, we use only reviews from 2016 (approximately 1 million reviews) and try to predict the rating 1-5.

The Amazon reviews dataset (McAuley and Leskovec, 2013) consists of 83.68 million Amazon product reviews from different product categories spanning May 1996 to July 2014. For

\(^1\)https://www.yelp.com/dataset
Table 1: Dataset Descriptions

<table>
<thead>
<tr>
<th>Dataset</th>
<th>Classes</th>
<th>Documents</th>
<th>Vocabulary</th>
<th>Task Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>Yelp Reviews 2016</td>
<td>5</td>
<td>1,033,037</td>
<td>72,880</td>
<td>Sentiment Analysis</td>
</tr>
<tr>
<td>Amazon Reviews Sentiment</td>
<td>5</td>
<td>500,000</td>
<td>67,802</td>
<td>Sentiment Analysis</td>
</tr>
<tr>
<td>Amazon Reviews Category</td>
<td>10</td>
<td>500,000</td>
<td>67,802</td>
<td>Topic Classification</td>
</tr>
<tr>
<td>Pubmed</td>
<td>8</td>
<td>800,000</td>
<td>182,167</td>
<td>Topic Classification</td>
</tr>
</tbody>
</table>

our evaluation, we selected 10 popular categories and extracted 50,000 randomly selected reviews from each: books, electronics, clothing, home and kitchen, sports and outdoors, health, video games, tools, pet supplies, and food. We use this dataset for two separate classification tasks – sentiment analysis (1-5) and product classification.

The Pubmed dataset \(^2\) consists of more than 26 million citations, abstracts, and other metadata from biomedical literature dating back to 1964. For our experiments, we use Pubmed abstracts associated with 8 common medical subject heading (MeSH) labels: metabolism, physiology, genetics, chemistry, pathology, surgery, psychology, and diagnosis. We only use abstracts that are associated with a single label, yielding a final selection of 800,000 abstracts, 100,000 for each label.

### 4.2 Baselines and Hyperparameters

As a baseline, we test the performance of two traditional machine learning classifiers that do not utilize deep learning: Naive Bayes (NB) and logistic regression (LR). For logistic regression, we use L1 regularization with a penalty strength of 1.0.

We also compare the performance of our HCAN to that of two other deep learning models. First, we test a word-level shallow-and-wide CNN using an architecture similar to that developed by Kim (2014) for sentence classification. We use three parallel convolution layers with 3-, 4-, and 5-word windows, all with 100 feature maps. These feed into a temporal maxpool across the entire document and the result is concatenated. We apply 50% dropout on the concatenated vector and feed this vector into a softmax classifier. This simple architecture has been shown to outperform many deeper and more complex CNN-based models (Le et al., 2017).

We also test the performance of HANs (Yang et al., 2016), which are the current state-of-the-art. For our HAN, we use the same optimized hyperparameters as those used by Yang – each hierarchy is composed of a bi-directional GRU with 50 units and an attention mechanism with a hidden layer of 200 neurons.

For the HCAN, we tuned the hyperparameters on the Yelp 2013 dataset. We tuned the attention embedding size \(d\) and the number of attention heads \(h\) used in our scaled dot-product attention. We use embedding size 512 and 8 heads for our final implementation.

### 4.3 Setup Details

For each dataset, we lowercase all characters and remove non-alphanumerics other than periods, exclamation marks, and questions marks (used to split documents into sentences). For the traditional machine learning approaches that utilize TFIDF features, we generate unigrams and bigrams with a minimum document frequency of 5. For deep learning models that utilize word embeddings, we train Word2Vec embeddings using a minimum word frequency of 5 and a dimension size of 512.

The deep learning models are all trained on a single document at a time with no batching. This configuration allows for variable length input so that long documents do not need to be cut short and short documents do not need to be padded. All models are trained using the Adam optimizer (Kingma and Ba, 2015) with learning rate 2E-5, beta1 0.9, and beta2 0.99.

We split each dataset into train, validation, and test sets using stratified 80/10/10 splitting. The TFIDF-based models are fitted on the train sets and evaluated on the test sets. The deep learning models are trained on the train set, and every 50,000 documents we evaluate on the validation set until the model converges. We save the model parameters with the highest validation accuracy and use those parameters to evaluate on the test set.

### 4.4 Results

The results of our experiments are displayed in Table 2. For each deep learning model, we record the final test accuracy, average time to train on a single
Table 2: Test set accuracy, mean training time for a single document, and total training time on each task

<table>
<thead>
<tr>
<th>Classifier</th>
<th>Yelp 2016</th>
<th>Amazon Sentiment</th>
<th>Amazon Category</th>
<th>Pubmed</th>
</tr>
</thead>
<tbody>
<tr>
<td>Naive Bayes</td>
<td>63.12</td>
<td>61.66</td>
<td>88.14</td>
<td>75.81</td>
</tr>
<tr>
<td>−, 1.8s</td>
<td>−, 0.8s</td>
<td>−, 1.3s</td>
<td>−, 4.2s</td>
<td></td>
</tr>
<tr>
<td>Logistic Regression</td>
<td>71.31</td>
<td>67.57</td>
<td>88.69</td>
<td>78.57</td>
</tr>
<tr>
<td>−, 306s</td>
<td>−, 101s</td>
<td>−, 173s</td>
<td>−, 463s</td>
<td></td>
</tr>
<tr>
<td>Word shallow-and-wide CNN</td>
<td>74.44</td>
<td>70.75</td>
<td>88.20</td>
<td>78.15</td>
</tr>
<tr>
<td></td>
<td>17ms, 9hr</td>
<td>15ms, 5hr</td>
<td>15ms, 5hr</td>
<td>35ms, 22hr</td>
</tr>
<tr>
<td>Hierarchical Attention Network</td>
<td>76.30</td>
<td>72.56</td>
<td>89.68</td>
<td>79.89</td>
</tr>
<tr>
<td></td>
<td>96ms, 67hr</td>
<td>97ms, 35hr</td>
<td>113ms, 37hr</td>
<td>167ms, 110hr</td>
</tr>
<tr>
<td>Conv Attention Network (One Self-Attention, Maxpool)</td>
<td>75.01</td>
<td>71.24</td>
<td>89.27</td>
<td>79.21</td>
</tr>
<tr>
<td></td>
<td>19ms, 13hr</td>
<td>19ms, 8hr</td>
<td>19ms, 8hr</td>
<td>38ms, 25hr</td>
</tr>
<tr>
<td>Conv Attention Network (One Self-Attention, Target Attention)</td>
<td>75.17</td>
<td>71.45</td>
<td>89.35</td>
<td>79.70</td>
</tr>
<tr>
<td></td>
<td>21ms, 14hr</td>
<td>21ms, 9hr</td>
<td>21ms, 9hr</td>
<td>39ms, 26hr</td>
</tr>
<tr>
<td>Conv Attention Network (Two Self-Attention, Maxpool)</td>
<td>75.21</td>
<td>71.45</td>
<td>89.41</td>
<td>79.86</td>
</tr>
<tr>
<td></td>
<td>23ms, 15hr</td>
<td>22ms, 9hr</td>
<td>22ms, 9hr</td>
<td>41ms, 27hr</td>
</tr>
<tr>
<td>Conv Attention Network (Two Self-Attention, Target Attention)</td>
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<td>71.78</td>
<td>89.71</td>
<td>79.95</td>
</tr>
<tr>
<td></td>
<td>25ms, 17hr</td>
<td>24ms, 10hr</td>
<td>24ms, 10hr</td>
<td>43ms, 29hr</td>
</tr>
<tr>
<td>Hierarchical Conv Attention Network (One Self-Attention, Maxpool)</td>
<td>75.00</td>
<td>71.09</td>
<td>88.85</td>
<td>79.31</td>
</tr>
<tr>
<td></td>
<td>24ms, 16hr</td>
<td>23ms, 9hr</td>
<td>23ms, 9hr</td>
<td>42ms, 28hr</td>
</tr>
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<td>Hierarchical Conv Attention Network (One Self-Attention, Target Attention)</td>
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<td>72.33</td>
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<td>79.91</td>
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<tr>
<td></td>
<td>34ms, 23hr</td>
<td>29ms, 12hr</td>
<td>29ms, 12hr</td>
<td>47ms, 31hr</td>
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<tr>
<td>Hierarchical Conv Attention Network (Two Self-Attention, Maxpool)</td>
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<td>72.43</td>
<td>89.34</td>
<td>80.09</td>
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<tr>
<td></td>
<td>31ms, 21hr</td>
<td>31ms, 13hr</td>
<td>31ms, 13hr</td>
<td>50ms, 33hr</td>
</tr>
<tr>
<td>Hierarchical Conv Attention Network (Two Self-Attention, Target Attention)</td>
<td><strong>76.51</strong></td>
<td><strong>72.85</strong></td>
<td><strong>89.89</strong></td>
<td><strong>80.13</strong></td>
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<tr>
<td></td>
<td>49ms, 32hr</td>
<td>38ms, 16hr</td>
<td>38ms, 16hr</td>
<td>53ms, 35hr</td>
</tr>
</tbody>
</table>

Figures 3: Validation accuracy vs time on Amazon sentiment analysis task.

In all four tasks, the HCAN achieves the highest test accuracy. Furthermore, HCANs process documents and converge more than twice as fast as the HAN (Figure 3). Within the HCAN, using a hierarchical structure achieves better accuracy than not using a hierarchical structure, using two parallel self-attentions achieves better accuracy than using a single self-attention, and using target attention outperforms using maxpool, especially when using a hierarchical structure.

We note that the difference in accuracy between the deep learning approaches and traditional machine learning approaches is greater in the sentiment classification tasks than in the topic classification tasks. We expect that this is because sentiment classification requires more semantic nuance, which can be difficult to capture via TFIDF features. On the other hand, topic classification may require the presence of only a few specific words to indicate a specific topic.

5 Discussion

Based on our results, we see that the two best performing architectures are the HCAN and the HAN. Unlike the shallow-and-wide CNN,
HCANs and HANs utilize a hierarchical structure that first breaks a document down into its constituent sentences. Using this structure, the networks first locate the most critical information within each sentence and then establish the relationships between the critical information found from each sentence. Our results suggest that this approach works better for text classification than scanning the entire document in one single pass to look for key features.

On our tasks, we see that the HCAN achieves similar performance with the HAN but trains much faster. We attribute this to the fact that HCANs utilize a self-attention-based architecture instead of an RNN-based architecture to extract features. Self-attention utilizes a feed-forward structure, whereas an RNN must backpropagate onto itself over time. When computing gradients, this means that much more calculation is required for RNN-based architectures. For our HCAN, we utilized a self-attention mechanism with a width of 512 neurons and were able to perform faster than our HAN that used an RNN with only 50 GRUs.

Another important implication of self-attention is that it is easier to parallelize than RNNs. Self-attention utilizes a fixed number of feed-forward steps that remain the same regardless of the length of the input sequences. This makes it simple to split the model parameters associated with self-attention across multiple GPUs even when processing multiple documents of different lengths. On the other hand, the number of operations for an RNN is dependent on the length of the input sequence. This makes it challenging to efficiently split RNN parameters across multiple GPUs when dealing with a mini-batch of documents that vary in length (Huang et al., 2013).

Utilizing an attention-based approach also increases the interpretability of the model. By examining the attention weights assigned to each word/sentence by the target attention mechanisms in each hierarchy of the HCAN, we can locate the words/sentences in a document that contribute most to its final label (Figure 4). Furthermore, we can also examine the attention weights assigned to each word/sentence in the self-attention mechanisms to establish how the HCAN is finding relationships between individual words/sentences when extracting important features (Figure 4).

Our results show that using two parallel self-attention mechanisms results in higher accuracy than using a single self-attention mechanism. Upon analyzing the attention weights assigned by the self-attention mechanisms, we found that using two parallel self-attendations captures more relationships involving modifier words than one single self-attention mechanism alone (Figure 5). Furthermore, we analyzed the documents that two parallel self-attendations classified correctly but one single self-attention did not. In sentiment analysis tasks, we found that many of these documents (1) begin positively but conclude negatively or vice versa, (2) contain a mix of positive and negative words, or (3) contain words that scale the meaning of another word or phrase (Supplementary A). This supports our hypothesis that two parallel self-attendations better distinguishes complex word relationships like scaling and negation.

To better understand how the HCAN functions in comparison with the HAN, we compared the attention weights assigned to each word/sentence by the target attention mechanisms in the two net-
Figure 5: Self-attention weights assigned to a sample word ‘it’ by (top) HCAN with a single self-attention and (bottom) HCAN with two parallel self-attentions. With two self-attentions, the first self-attention captures the relationship between ‘it’ and ‘doesnt’ and the second self-attention captures the relationship between ‘it’ and ‘chop’. This is a more nuanced negation relationship that isn’t captured when using a single self-attention.

works. We found the HCAN weight assignments to be more spread out than those from the HAN (Supplementary B). Further analysis revealed that the self-attention mechanisms in the HCAN distribute the meaning of important keywords across many other words before the sequence is fed into the target attention mechanism, thus resulting in the wider distribution of attention weights.

6 Conclusion

In this work, we introduced a new self-attention-based text classification architecture, HCANs, and compared its performance with the current state-of-the-art, HANS, in four classification tasks: Yelp review sentiment, Amazon review sentiment, Amazon review product category, and Pubmed abstract topic. In all four tasks HCANs achieved slightly better performance than HANs while being more than twice as fast to train. Our results show that in time-critical NLP tasks, self-attention-based architectures may be able to replace RNN-based architectures to reduce training time without sacrificing accuracy. Moving forward, we plan to explore efficient implementations of data and model parallelism for self-attention-based architectures such as the HCAN. The code for our experiments is available online at https://code.ornl.gov/v33/HCAN/.

Acknowledgments

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A  Yelp Reviews Misclassified by Single Self-Attention

The following are examples of Yelp reviews that were misclassified when using a single self-attention mechanism but correctly classified when using two parallel self-attention mechanisms. Note in many of these reviews, one section of the review negates or scales the meaning in another section.

i got this at a grocery store thinking it would be great since i only drink a little bit of wine or sake at a time. i ended up giving it away to goodwill after a few months because it doesn’t really help the wine or sake at least not for weeks like im prone to need between glasses and it is annoying to use the plastic thingy trying to get it tight and worrying that you’re going to break the bottle. i think a nice reusable cork kind of gadget would do just as good a job take up less drawer space and look prettier in the bottle.

for those of you who criticized this book for lack of a plot i can only assume that you are much more suited to books in the mystery thriller genre. i loved it and found the characters very real and compelling. if you are a reader who likes books about relationships you are going to love it too.

i hesitated buying this grill because there were so many negative reviews. im glad i decided to buy the grill. we’ve used it 5 times so far. to address some of the negative reviews. you can cook with the grill on both high and low with the cover closed. in the instructions you are actually directed to clean the grill for the first time with the burners on high and the cover closed. the stand is excellent. we’ve been using this at the beach. the stand and fold out tables save packing additional cargo in the car. as far as cleaning i dont know what people are expecting. its a bbq it gets dirty. the grates clean up real nice with brillo. the chrome area under the grill plates cleans up with a fantastic type cleaner.

a feel good read. dean koontz does this type of book very very well. no horrid monsters except for the unscrupulous government people so dont expect nightmares from this one. it does have its suspense however.

its fun in the beginning. but the levels get harder and game play is not as fun. it got so hard it was not much fun to play. and has not much variety in it.

B  Comparing Attention Weights from HAN and HCAN

The attention weights assigned by the target-attention for the HAN (Figure 6) are more focused on keywords than for the HCAN (Figure 7). This is because the self-attention in the HCAN redistributes the content of important keywords across other words before the sequence is sent into the target-attention (Figure 8).

Figure 6: HAN target-attention weights assigned to a sample Yelp review. We see that the weights are primarily focused on sentiment keywords.
Figure 7: HCAN target-attention weights assigned to a sample Yelp review. We see that the weights are more spread out than in the HAN target-attention.

Figure 8: HCAN self-attention weights assigned to the words "the" and "only" in a sample Yelp review sentence. We see that meaning from sentiment keywords are redistributed among other words. In the two example shown above, we see that "best" is reassigned to "the" and "awesome" is reassigned to "only". Therefore, the HCAN target-attention weighs the words "the" and "only" higher because they contain content from sentiment keywords.
Extrofitting: Enriching Word Representation and its Vector Space with Semantic Lexicons

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Abstract

We propose post-processing method for enriching not only word representation but also its vector space using semantic lexicons, which we call extrofitting. The method consists of 3 steps as follows: (i) Expanding 1 or more dimension(s) on all the word vectors, filling with their representative value. (ii) Transferring semantic knowledge by averaging each representative values of synonyms and filling them in the expanded dimension(s). These two steps make representations of the synonyms close together. (iii) Projecting the vector space using Linear Discriminant Analysis, which eliminates the expanded dimension(s) with semantic knowledge.

1 Introduction

As a method to represent natural language on computer, researchers have utilized distributed word representation. The distributed word representation is to represent a word as n-dimensional float vector, hypothesizing that some or all of the dimensions may capture semantic meaning of the word. The representation has worked well in various NLP tasks, substituting one-hot representation (Turian et al., 2010). Two major algorithms learning the distributed word representation are CBOW (Continuous Bag-of-Words) and skip-gram (Mikolov et al., 2013b). Both CBOW and skip-gram learn the representation using one hidden neural networks. The difference is that CBOW learns the representation of a center word from neighbor words whereas skip-gram gets the representation of neighbor words from a center word. Therefore, the algorithms have to depend on word order, because their objective function is to maximize the probability of occurrence of neighbor words given the center word. Then a problem occurs because the word representations do not have any information to distinguish synonyms and antonyms. For example, worthy and desirable should be mapped closely on the vector space as well as agree and disagree should be mapped apart, although they occur on a very similar pattern. Researchers have focused on the problem, and their main approaches are to use semantic lexicons (Faruqui et al., 2014; Mrkšić et al., 2016; Speer et al., 2017; Vulić et al., 2017; Camacho-Collados et al., 2015). One of the successful works is Faruqui’s retrofitting1, which can be summarized as pulling word vectors of synonyms close together by weighted averaging the word vectors on a fixed vector space (it will be explained in Section 2.1). The retrofitting greatly improves word similarity between synonyms, and the result not only corresponds with human intuition on words but also performs better on document classification tasks with comparison to original word embeddings (Kiela et al., 2015). From the idea of retrofitting, our method hypothesize that we can enrich not only word representation but also its vector space using semantic lexicons2. We call our method as extrofitting, which retrofit word vectors by expanding its dimensions.

1The retrofitting codes are available at https://github.com/mfaruqui/retrofitting
2Our codes are available at https://github.com/HwiyeolJo/Extrofitting
2 Backgrounds

2.1 Retrofitting

Retrofitting (Faruqui et al., 2014) is a post-processing method to enrich word vectors using synonyms in semantic lexicons. The algorithm learns the word embedding matrix $Q = \{q_1, q_2, \ldots, q_n\}$ with the objective function $\Psi(Q)$:

$$\Psi(Q) = \sum_{i=1}^{n} [\alpha||q_i - \hat{q}_i||^2 + \sum_{(i,j) \in E} \beta_{ij}||q_i - q_j||^2]$$

(1)

where an original word vector is $q_i$, its synonym vector is $q_j$, and inferred word vector is $\hat{q}_i$. The hyperparameter $\alpha$ and $\beta$ control the relative strengths of associations. The $\hat{q}_i$ can be derived by the following online update: $\hat{q}_i = \frac{\sum_{(r,l) \in E} \beta_{rl} q_l + \alpha q_i}{\sum_{j, (i,j) \in E} \beta_{ij} + \alpha}$.

2.2 Linear Discriminant Analysis (LDA)

LDA (Welling, 2005) is one of the dimension reduction algorithms that project data into different vector space, while minimizing the loss of class information as much as possible. As a result, the algorithm finds linear vector spaces which minimize the distance of data in the same class as well as maximize the distance among the different class. The algorithm can be summarized as follows:

Calculating between-class scatter matrix $S_B$ and within-class scatter matrix $S_W$.

When we denote data as $x$, classes as $c$, $S_B$ and $S_W$ can be formulated as follows:

$$S_B = \sum_c (\mu_i - \mu)(\mu_i - \mu)^T, \quad (2)$$

$$S_W = \sum_c \sum_{i \in c} (x_i - \mu_c)(x_i - \mu_c)^T, \quad (3)$$

where the overall average of $x$ is $\mu$, and the partial average in class $i$ is denoted by $\mu_i$.

Maximizing the objective function $J(w)$.

The objective function $J(w)$ that we should maximize can be defined as

$$J(w) = \frac{|U^T S_B U|}{|U^T S_W U|}, \quad (4)$$

and its solution can be reduced to find $U$ that satisfies $S_W^{-1} S_B = U \Lambda U^T$. Therefore, $U$ is derived by eigen-decomposition of $S_W^{-1} S_B$; choosing $q$ eigen vectors which have the top-$q$ eigen values, and composing transform matrix of $U$.

Transforming data onto new vector space

Using transform matrix $U$, we can get transformed data by $y = U^T x$.

3 Enriching Representations of Word Vector and The Vector Space

3.1 Expanding Word Vector with Enrichment

We simply enrich the word vectors by expanding dimension(s) that add 1 or more dimension to original vectors, filling with its representative value $r_i$, which can be a mean value. We denote an original word vectors as $q_i = (e_1, e_2, \ldots, e_D)$ where $D$ denotes the number of word vector dimension. Then, the representative value $r_i$ can be formulated as $r_i = mean(e_1, e_2, \ldots, e_D)$. Intuitively, if we expand more additional dimensions, the word vectors will strengthen its own meaning. Likewise, the ratio of the number of expanded dimension to the number of original dimensions will affect the meaning of the word vectors.

3.2 Transferring Semantic Knowledge

To transfer semantic knowledge on the representative value $r_i$, we can also take a simple approach of averaging all the representative values of each synonym pair, substituting each of its previous value. We get the synonym pairs from lexicons we introduced in Section 3. The transferred representative value $\bar{r}_i$ can be formulated as $\bar{r}_i = \frac{\sum_{s \in L} r_s}{N}$ where $L$ refers to the lexicon consisting of synonym pairs $s$, and $N$ is the number of synonyms. This manipulation makes the representation of the synonym pairs close to one another.

3.3 Enriching Vector Space

With the enriched vectors and the semantic knowledge, we perform Linear Discriminant Analysis for dimension reduction as well as clustering the synonyms from semantic knowledge. LDA finds new vector spaces to cluster and differentiate the labeled data, which are synonym pairs in this experiment. We can get the extrofitted word embedding matrix $\tilde{w}$ as follows:

$$\tilde{Q} = \text{LDA}(\bar{Q} \oplus \bar{r}, l) \quad (5)$$

where $\bar{Q}$ is the word embedding matrix composed of word vectors $q$ and $l$ is the index of the synonym pair. We implement our method using Python2.7 with scikit-learn (Pedregosa et al., 2011).

4 Experiment Data

4.1 Pretrained Word Vectors

GloVe (Pennington et al., 2014) has lots of variations in respect to word dimension, number of to-
4.2 Semantic Lexicons

We borrow the semantic lexicons from retrofitting (Faruqui et al., 2014). Faruqui et al. extracted the synonyms from PPDB (Ganitkevitch et al., 2013) by finding a word that more than two words in another language are corresponding with. Retrofitting also used WordNet (Miller, 1995) database which grouped words into set of synonyms (synsets). We used two versions of WordNet lexicon, one which consists of synonym only (WordNet_syn) and the other with additional hypernyms, hyponyms included (WordNet_all). Lastly, synonyms were extracted from FrameNet (Baker et al., 1998), which contains more than 200,000 manually annotated sentences linked to semantic frames. Faruqui et al. regarded words as synonyms if the words can be grouped with any of the frames.

4.3 Evaluation Data

We evaluate our methods on word similarity tasks using 4 different kinds of dataset. MEN-3k (Bruni et al., 2014) consists of 3000-word pairs rated from 0 to 50. WordSim-353 (Finkelstein et al., 2001) consists of 353-word pairs rated from 0 to 10. SimLex-999 (Hill et al., 2015) includes 999-word pairs rated from 0 to 10. RG-65 (Rubenstein and Goodenough, 1965) has 65 words paired scored from 0 to 4. MEN-3k and WordSim-353 were split into train (or dev) set and test set, but we combined them together solely for evaluation purpose. The other datasets have lots of out-of-vocabulary, so we disregard them for future work.

5 Experiments on Word Similarity Task

The word similarity task is to calculate Spearman’s correlation (Daniel, 1990) between two words as word vector format. We first apply extrofitting to GloVe from different data sources and present the result in Table 1. The result shows that although the number of the extrofitted word with FrameNet is less than the other lexicons, its performance is on par with other lexicons. We can also ensure that our method improves the performance of original pretrained word vectors.

Next, we perform extrofitting on GloVe in different word dimension and compare the performance with retrofitting. We use WordNet_all lexicon on both retrofitting and extrofitting to compare the performances in the ideal environment for retrofitting. We present the results in Table 2. We can demonstrate that our method outperforms retrofitting on some of word similarity tasks, MEN-3k and WordSim-353. We believe that extrofitting on SimLex-999 and RG-65 is less powerful because all word pairs in the datasets are included on WordNet_all lexicon. Since retrofitting forces the word similarity to be

---

Table 1: Spearman’s correlation of extrofitted word vectors for word similarity tasks using semantic lexicon. Our method improves pretrained GloVe in different vocabulary size.

<table>
<thead>
<tr>
<th></th>
<th>MEN-3k</th>
<th>WS353</th>
<th>SL-999</th>
<th>RG-65</th>
<th>#Extrofitted</th>
<th>#Vocab.</th>
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<tbody>
<tr>
<td>glove.6B.300d</td>
<td>0.7486</td>
<td>0.5170</td>
<td>0.3705</td>
<td>0.7693</td>
<td>-</td>
<td>0.4M</td>
</tr>
<tr>
<td>+ PPDB</td>
<td>0.7949</td>
<td>0.5826</td>
<td>0.4387</td>
<td>0.8177</td>
<td>67,729</td>
<td>-</td>
</tr>
<tr>
<td>+ WordNet_syn</td>
<td>0.7884</td>
<td>0.5805</td>
<td>0.4409</td>
<td>0.7943</td>
<td>55,388</td>
<td>-</td>
</tr>
<tr>
<td>+ WordNet_all</td>
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<td>0.5714</td>
<td>0.4353</td>
<td>0.8010</td>
<td>55,388</td>
<td>-</td>
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<tr>
<td>+ FrameNet</td>
<td>0.7840</td>
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<td>7,592</td>
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<td>0.8362</td>
<td>76,631</td>
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<tr>
<td>+ WordNet_syn</td>
<td>0.8230</td>
<td>0.6605</td>
<td>0.4884</td>
<td>0.8634</td>
<td>70,411</td>
<td>-</td>
</tr>
<tr>
<td>+ WordNet_all</td>
<td>0.8223</td>
<td>0.6638</td>
<td>0.4858</td>
<td>0.8561</td>
<td>70,411</td>
<td>-</td>
</tr>
<tr>
<td>+ FrameNet</td>
<td>0.8123</td>
<td>0.6448</td>
<td>0.4601</td>
<td>0.8556</td>
<td>7,809</td>
<td>-</td>
</tr>
</tbody>
</table>
improved by weighted averaging their word vectors, it is prone to be overfitted on semantic lexicons. On the other hand, extrofitting also uses synonyms to improve word similarity but it works differently that extrofitting projects the synonyms both close together on a new vector space and far from the other words. Therefore, our method can make more generalized word representation than retrofitting. We plot top-100 nearest words using t-SNE (Maaten and Hinton, 2008), as shown in Figure 1. We can find that retrofitting strongly collects synonym words together whereas extrofitting weakly disperses the words, resulting loss in cosine similarity score. However, the result of extrofitting can be interpreted as generalization that the word vectors strengthen its own meaning by being far away from each other, still keeping synonyms relatively close together (see Table 3). When we list up top-10 nearest words, extrofitting shows more favorable results than retrofitting. We can also observe that extrofitting even can be applied to words which are not included in semantic lexicons.

Lastly, we apply extrofitting to other well-known pretrained word vectors trained by different algorithms (see Subsection 4.1). The result is presented in Table 4. Extrofitting can be also applied to Word2Vec and Fasttext, enriching their word
representations except on WordSim-353 and RG-65, respectively. We find that our method can distort the well-established word embeddings. However, our results are noteworthy in that extrofitting can be applied to other kinds of pretrained word vectors for further enrichment.

6 Conclusion

We propose post-processing method for enriching not only word representation but also its vector space using semantic lexicons, which we call extrofitting. Our method takes a simple approach that (i) expanding word dimension (ii) transferring semantic knowledge on the word vectors (iii) projecting the vector space with enrichment. We show that our method outperforms another post-processing method, retrofitting, on some of word similarity task. Our method is robust in respect to the dimension of word vector and the size of vocabulary, only including an explainable hyperparameter; the number of dimension to be expanded. Further, our method does not depend on the order of synonym pairs. As a future work, we will do further research about our method to generalize and improve its performance; First, we can experiment on other word similarity datasets for generalization. Second, we can also utilize Autoencoder (Bengio et al., 2009) for non-linear projection with a constraint of preserving spatial information of each dimension of word vector.

Acknowledgments

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References


Chat Discrimination for Intelligent Conversational Agents with a Hybrid CNN-LMTGRU Network

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Abstract

Recently, intelligent dialog systems and smart assistants have attracted the attention of many, and development of novel dialogue agents have become a research challenge. Intelligent agents that can handle both domain-specific task-oriented and open-domain chit-chat dialogs are one of the major requirements in the current systems. In order to address this issue and to realize such smart hybrid dialogue systems, we develop a model to discriminate user utterance between task-oriented and chit-chat conversations. We introduce a hybrid of convolutional neural network (CNN) and a lateral multiple timescale gated recurrent units (LMTGRU) that can represent multiple temporal scale dependencies for the discrimination task. With the help of the combined slow and fast units of the LMTGRU, our model effectively determines whether a user will have a chit-chat conversation or a task-specific conversation with the system. We also show that the LMTGRU structure helps the model to perform well on longer text inputs. We address the lack of dataset by constructing a dataset using Twitter and Maluuba Frames data. The results of the experiments demonstrate that the proposed hybrid network outperforms the conventional models on the chat discrimination task as well as performed comparable to the baselines on various benchmark datasets.

1 Introduction

Dialogue systems can be classified as domain-specific task-oriented and open-domain chit-chat dialog systems (Williams and Young, 2007; Wallace, 2009). The task-oriented dialog systems help users complete tasks in specific domains. The chit-chat dialog systems enable users to have an open-ended chat conversations with the system. While most of the functionalities offered by the two types of systems are complementary to each other, there have been very little efforts made to combine these two type of systems. Therefore, the potential of chat agents have been limited.

Recently, intelligent assistants have become popular with the integration of such systems in smartphones and home appliances. These intelligent assistants typically perform various tasks including weather forecast alerts, alarm settings, web search, and so on. Moreover, such assistants need to have the ability to perform chit-chat conversation with the users. This has led to the need for the development of novel and hybrid multi-domain task-oriented agents and open-domain chit-chat agents.

In order to develop such hybrid agents, we have to determine whether a user will have a chit-chat with the system or the user is looking for a task completion. For example, if a user says “Hi, how are you doing?”, then the user can be considered to have a chat with the system. Alternatively, if the user says “I want a flight to Los Angeles,” then the user is looking for a completion of a specific task. We address this task as a binary classification problem and call this task as chat discrimination.

Chat discrimination has not been sufficiently investigated in recent times. This is mainly because there are not enough studies to develop hybrids of task-oriented and chit-chat agents. Although task-oriented and chit-chat agents have long research histories, they do not require chat discrimination. We usually assume that the users of task-oriented agents will have task-oriented conversations with the systems and the users of chit-chat
agents will always have non-task-specific conversations with the systems. In a recent study, researchers in (Akasaki and Kaji, 2017) have tried chat detection using conventional classifiers with the help of a newly created dataset in Japanese language. But this dataset has not been released for further research or comparison.

In this work, we develop a hybrid network for chat discrimination by combining a convolutional neural network (CNN) and a gated recurrent unit (GRU). CNNs have been proven to be suitable for text classification problems (Kim, 2014; Johnson and Zhang, 2015a,b). Moreover, the temporal hierarchy concept with multiple timescale gated recurrent unit (MTGRU) (Kim et al., 2016) has also been proven to perform well in language modeling (Moirangthem and Lee, 2017; Moirangthem et al., 2017) and summarization (Kim et al., 2016) tasks. The MTGRU is known to handle long term dependency better with the help of the varying timescales to represent multiple composition- alities of language. The temporal hierarchy approach has also been shown to eliminate the need for complex structures and normalization techniques (Cooijmans et al., 2017; Krueger and Memisevic, 2016; Chung et al., 2017; Ha et al., 2017), and thereby increasing the computational efficiency of the model.

For our classification model, we develop a lateral multiple timescale structure. Our proposed lateral multiple timescale gated recurrent unit (LMTGRU) is significantly different from the conventional hierarchical MTGRU structure. The conventional MTGRU is most effective for handling long term dependencies in very long text inputs for applications such as summarization but performs comparable to vanilla GRU with shorter text inputs. Unlike the hierarchical architecture, the lateral connections in an LMTGRU will enable encoding of rich features that have different temporal dependencies from the input utterances in order to help classify the information correctly. LMTGRU follows a lateral (branch or root) architecture where the slow and fast units are directly connected to the inputs and the final output of the units are combined to form the final representation. This structure enables all the layers with different timescales to capture relevant features directly from the inputs unlike hierarchical multi-layer structures. Since the data consist of utterances as input, and the input to the RNN is represented as higher order features from the CNN, LMTGRU proves to be more suitable for this task. Our major contributions are as follows:

- We introduce a hybrid CNN-LMTGRU structure to build rich features from input texts to classify utterances correctly.
- The LMTGRU architecture enables our model to perform well on longer text sequences with the help of the slow layer as well as maintain comparable performance on shorter sequences.
- To address the lack of dataset, we create a dataset using Twitter data (Microsoft Research Social Media Conversation Corpus) (Sordoni et al., 2015) for chit-chat conversations and Maluuba Frames data (El Asri et al., 2017) for task-oriented conversations.
- In order to demonstrate that the proposed model performs well on other text classification tasks and to compare it to the existing baselines, we report the performance on various sentence classification benchmark datasets. The results of our experiments demonstrate that the proposed model performs well on the benchmark datasets as well.

2 Related Work

Although there have been enough studies for task-oriented and chit-chat agents independently, developing hybrid models of the two types of agents has not been explored enough. Therefore, few attempts have been made to develop a chat discrimination model.

Niculescu and Banchs (2015) tried to combine task-oriented agents and chit-chat agents, but the authors did not have a clear way to automatically determine when to switch back to the chit-chat agent. Lee et al. (2009) proposed to combine task-oriented and chit-chat agents with the help of an example-based dialogue manager, but it is difficult to integrate the current state-of-the-art deep learning model based classifiers as a component in such a framework.

Wang et al. (2014) and Sarikaya (2017) proposed to combine a multi-domain task-oriented agents and chit-chat agents using machine-learning-based frameworks. Robichaud et al. (2014); Sarikaya et al. (2016) approached domain
classification as ranking between alternate “dialog experts”. In a recent study, Akasaki and Kaji (2017) tried chat detection using conventional classifiers with the help of a newly created dataset in Japanese language. They used concatenated features from multiple feature extractors for the classification. An end-to-end model was not explored. Moreover, the dataset has not been released for further research or comparison.

Deep learning based models have achieved great success in many NLP tasks, including learning distributed word, sentence and document representation (Mikolov et al., 2013; Le and Mikolov, 2014), parsing (Socher et al., 2013), statistical machine translation (Cho et al., 2014), sentiment classification (Kim, 2014), etc. Learning distributed sentence representation through neural network models requires little external domain knowledge and can reach satisfactory results in related tasks like sentiment classification, text categorization etc.

In recent sentence representation learning works, neural network models are constructed upon either the input word sequences or the transformed syntactic parse tree. Among them, convolutional neural network (CNN) and recurrent neural network (RNN) are two popular ones. The capability of capturing local correlations along with extracting higher-level correlations through pooling empowers CNN to model sentences naturally from consecutive context windows. Kim (2014) proposed a CNN architecture with multiple filters and multiple channels for text classification.

RNNs are able to deal with variable-length input sequences and discover long-term dependencies. Various variants of RNNs have been proposed to better store and access memories. The most popular variants are long short-term memory (LSTM) (Hochreiter and Schmidhuber, 1997) and GRU (Cho et al., 2014). Recently proposed MTGRU (Kim et al., 2016), inspired by the concept of temporal hierarchy found in the human brain (Botvinick, 2007; Meunier et al., 2010), demonstrates the ability to capture multiple compositionalities similar to the findings of Ding et al. (2016). This better representation learning capability enhances the ability of the network to model longer sequences of text.

In this paper, we develop a hybrid of CNN and LMTGRU in a unified architecture for semantic sequence modeling. We apply CNN to text data and feed the features directly to the LMTGRU, and hence our architecture enables the network to learn multiple temporal scale dependencies from higher-order features. We hypothesize that the combination of slow and fast features will be beneficial for the chat discrimination task.

3 Proposed Model

We formulate chat discrimination as a binary classification problem. In this section, we explain the proposed hybrid classifier model shown in Figure 1.

3.1 The Convolutional Neural Network Layer

The CNN layer shown in Figure 1 is implemented using a single convolution and max-pooling layer and use a rectified linear unit (ReLU) as the non-linear activation function following Kim (2014). Let \( x_i \in \mathbb{R}^d \) be the word vector of dimension \( d \) corresponding to the \( i \)-th word in the input utterance. An utterance of length \( n \), which are padded if necessary, can be represented as

\[
x_{1:n} = x_1 \oplus x_2 \oplus \ldots \oplus x_n,
\]

where \( \oplus \) is the concatenation operator. Let \( x_{i:i+j} \) be to the concatenation of words \( x_i, x_{i+1}, \ldots, x_{i+j} \). A convolution operation involves a filter \( w \in \mathbb{R}^{hd} \), which is applied to a window of \( h \) words to produce a new feature. For example, a feature \( c_i \) is generated from a window of words \( x_{i:i+h-1} \) by

\[
c_i = f(w \cdot x_{i:i+h-1} + b).
\]

Here \( b \in \mathbb{R} \) is a bias term and \( f \) is a non-linear function. This filter is applied to each possible window of words in the sentence \( \{ x_{1:h}, x_{2:h+1}, \ldots, x_{n-h+1:n} \} \) to produce a feature map

\[
c = [c_1, c_2, \ldots, c_{n-h+1}],
\]

with \( c \in \mathbb{R}^{n-h+1} \). A max pooling operation (Collobert et al., 2011) over the feature map is applied, which takes the maximum value \( \hat{c} = \max\{c\} \) as the feature corresponding to this particular filter. The idea is to capture only the most important features.

The processes described above is for one feature being extracted from one filter. The proposed CNN model includes a number of filters with multiple window sizes to obtain various features. These features are then split into
Figure 1: The proposed CNN-LMTGRU classifier. The input to the model is “How are you doing today?”

The multiple timescales in an MTGRU network is implemented by applying a timescale variable at the end of a conventional GRU unit, essentially adding another gating unit that modulates the mixture of the past and current hidden states. In an MTGRU, each step takes as input $x_t, h_{t-1}$ and produces the hidden $h_t$. The timescale $\tau$ added to the activation $h_t$ of the MTGRU is shown in Eq. (4). $\tau$ is used to control the timescale of each GRU cell. Larger $\tau$ results in slower cell outputs but it makes the cell focus on the slow features and vice-versa. The timescale variable $\tau$ is scalar and one $\tau$ controls the slow cells and another $\tau$ controls the fast cells. We initialize the $\tau$ for each group of cells, e.g. larger $\tau$ for slow cells and smaller $\tau$ for fast cells. The $\tau$ is made as a trainable variable like any other weight of the network and is optimized during the training based on the final loss. An MTGRU cell is illustrated in Figure 2.

$$
\begin{align*}
    r_t &= \sigma(W_{xr}x_t + W_{hr}h_{t-1}) \\
    z_t &= \sigma(W_{xz}x_t + W_{hz}h_{t-1}) \\
    u_t &= \tanh(W_{zu}x_t + W_{hu}(r_t \odot h_{t-1})) \\
    \tilde{h}_t &= z_th_{t-1} + (1 - z_t)u_t \\
    h_t &= \frac{\tilde{h}_t}{\tau} + (1 - \frac{1}{\tau})h_{t-1}
\end{align*}
$$

where $\sigma(\cdot)$ and $\tanh(\cdot)$ are the sigmoid and tangent hyperbolic activation functions, $\odot$ denotes

$n/max$ pool size outputs and are passed on to the LMTGRU layer.

3.2 The Lateral MTGRU Layer

For this classification task, we implement a lateral multiple timescale architecture where half of the MTGRU units are fast and the remaining half are slow as shown in Figure 1. The fast and slow units can capture different temporal dependencies from the input sequence. The fast timescale layer can capture fast changing features (e.g. character or word) whereas slower timescales can represent phrase or sentence level features (Moirangthem et al., 2017). The proposed LMTGRU structure follows a lateral (branch or root) architecture where the slow and fast units are directly connected to the inputs. This lateral architecture is different from the conventional MTGRU with a hierarchical layer architecture, since the LMTGRU does not follow a multilayer structure. The LMTGRU structure is implemented using multiple single layer MTGRU networks whose timescales are different and the input to each layer comes directly from the input features. And the final output representation features of each layer are combined to form the penultimate representation of the input sequence that includes both fast and slow features.
the element-wise multiplication operator, and $r_t$, $z_t$ are referred to as reset, update gates respectively. $u_t$ and $\tilde{h}_t$ are the candidate activation and candidate hidden state of the MTGRU.

The proposed CNN-LMTGRU hybrid network consists of a CNN layer followed by a fast and a slow LMTGRU layer. The fast units as well as the slow units are directly connected to the CNN features. Finally the combined last hidden representation of the LMTGRU is passed to a fully connected softmax layer whose output is the probability distribution over the labels.

4 Chat Discrimination Dataset

Chat discrimination task requires a chat dataset like the one shown in Table 1. We address the lack of such a dataset by using the Microsoft Research Social Media Conversation Corpus\footnote{https://www.microsoft.com/en-us/download/details.aspx?id=52375} and Maluuba Frames\footnote{https://datasets.maluuba.com/} datasets. Microsoft Research Social Media Conversation Corpus is a collection of conversational snippets extracted from Twitter logs. The advantage of using this dataset is that it has been evaluated by crowd sourced annotators measuring quality of the response. These data are suitable for detecting open-domain non-task oriented chats. On the other hand, we use the Maluuba Frames dataset for the domain task-specific conversations. This corpus is for the travel agent domain where the users can inquire the agent and ask for booking of hotels and flights. The dialogs were recorded using 12 participants over a period of 20 days. We process the data to utilize only the user utterances in our chat discrimination dataset. Finally, we have 20,532 utterances with 10,266 in each class. We divide the data into 10% for validation, 10% for test, and the remaining for train.

5 Experiments and Results

We evaluate the performance of the proposed method and compare it to the conventional models using our chat discrimination dataset. In order to demonstrate that the proposed model performs well on other text classification datasets and to compare it to the existing baselines, we report the performance on various sentence classification benchmark datasets as well.

5.1 Experiment settings

We trained the proposed CNN-LMTGRU model in an end-to-end fashion, where we do not use any pre-trained word embedding. An embedding of size 300 was used for the model and was trained with the model. We used 128 filters of sizes $\{3, 4, 5\}$ for the CNN.

We used 300 units of MTGRU where half of the units are fast and the remaining are slow units to construct the LMTGRU structure. The $\tau$ for the fast units and the slow units were initialized to 1.0 and 1.25, respectively. We follow Moirangthem et al. (2017) to initialize the timescale parameter.

In order to control the $\tau$ during training, we set the lower bound to 1.0 using clip by value. This is done as the fastest layer should have a $\tau$ of 1.0, however there is no upper bound for the slow layers. After training, the final $\tau$ values are 1.16 and 1.37 for the fast and the slow layers, respectively. The learning rate to update the $\tau$, which is different from the global learning rate, is set to 0.00001 in order to avoid large changes in the timescale.

We used the RMSProp Optimizer (Tieleman and Hinton, 2012) to perform stochastic gradient descent with the decay set to 0.9 and the global learning rate to 0.001. For regularization we employ dropout of 0.5 on the final CNN output as well as in the LMTGRU layers to avoid overfitting. We utilized the validation performance for early stopping of the training for better generalization.

5.2 Baseline Models

The baseline models implemented for the comparison using our chat discrimination dataset are described as follows:

CNN We used the same parameters as before except the number of filters were increased to
Type | Example
-----|-----------------
Chit-Chat | Let’s meet at the coffee place and talk about you.
What is your hobby?
I will visit my parents for the vacation.
I like pop music.
Do you like soccer?
I don’t know you, but you seem to be a serious person.
Task-oriented | Hello, I am looking to book a trip for 2 adults and 6 children.
We are departing from Kochi for Denver.
When would I be leaving for each of them?
I would like to spend as much time in Denver as my budget will allow.
Do these packages have different departure dates?
Ok, I would like to purchase the trip with the 4-star hotel.

Table 1: Example utterances of the two kinds of conversations.

256. We followed (Kim, 2014) and used a fully connected softmax layer for the binary classification.

**LSTM/GRU** The same parameters were used as before except the number of hidden units is increased to 500. The LSTM/GRU takes every word vector in a sequence as input and the final representation is passed to a softmax layer for classification.

**LMTGRU** This LMTGRU model consists of a fast and a slow layer with 250 hidden units in each layer. The remaining settings are the same as the LSTM/GRU model.

**CNN-LSTM/GRU** This structure is almost identical to the proposed model, but instead of the LMTGRU, LSTM/GRU is used for comparison. The parameters remain the same.

5.3 Evaluation on Benchmark Datasets

Following Kim (2014), we test our model on various benchmarks. Summary statistics of the datasets are given below.

- **MR**: Movie reviews with one sentence per review. This binary classification task involves detecting positive/negative reviews (Pang and Lee, 2005). The average sequence length is 20 and the dataset size is 10,662.

- **SST-1**: This is the Stanford Sentiment Treebank is an extension of MR with multiple labels (very positive, positive, neutral, negative, very negative) (Socher et al., 2013). The average sequence length is 18 and the dataset size is 11,855.

- **SST-2**: This is similar to SST-1 but with binary labels. The average sequence length is 19 and the dataset size is 9,613.

- **Subj**: Subjectivity dataset consists of sentences with binary labels (subjective or objective). The average sequence length is 23 and the dataset size is 10,000 (Pang and Lee, 2004).

- **TREC**: The TREC task is a classification task to classify 6 types of question (questions about person, location, numeric information, etc.). The average sequence length is 10 and the dataset size is 5,952 (Li and Roth, 2002).

- **CR**: Customer reviews of various products with positive/negative labels. The average sequence length is 19 and the dataset size is 3,775 (Hu and Liu, 2004).

- **MPQA**: Opinion polarity detection is a sub-task of the MPQA dataset with 2 classes. The average sequence length is 3 and the dataset size is 10,606 (Wiebe et al., 2005).

For the evaluation on the benchmark datasets, we implemented a CNN-LMTGRU model that is identical to the one described in Section 5.1. The data for train, validation, and test for the benchmark datasets follow the previous works (Kim, 2014; Kalchbrenner et al., 2014).

5.4 Results

Table 2 illustrates the classification performance of the various models. The performance is given in accuracy and the results show that the proposed hybrid CNN-LMTGRU model outperforms
Table 2: Chat classification results on the test set.

<table>
<thead>
<tr>
<th>Model</th>
<th>Accuracy (%)</th>
</tr>
</thead>
<tbody>
<tr>
<td>CNN</td>
<td>91.12</td>
</tr>
<tr>
<td>LSTM</td>
<td>89.67</td>
</tr>
<tr>
<td>GRU</td>
<td>90.56</td>
</tr>
<tr>
<td>LMTGRU</td>
<td>90.64</td>
</tr>
<tr>
<td>CNN-LSTM</td>
<td>92.31</td>
</tr>
<tr>
<td>CNN-GRU</td>
<td>93.01</td>
</tr>
<tr>
<td>Proposed CNN-LMTGRU</td>
<td>94.69</td>
</tr>
</tbody>
</table>

Figure 3: Classification accuracy curve on the validation set of the proposed method and the hybrid baseline models.

the baseline models. The performance curve of the hybrid models is shown in Figure 3, respectively.

In order to differentiate the performance of the proposed CNN-LMTGRU model and the CNN-GRU model, we divide the test data of the dialog classification dataset according to the length of the texts. Figure 4 shows the comparison of the performance accuracy on different lengths of test data. It can be seen that the LMTGRU structure enables the model to outperform GRU on longer text inputs and there is no significant performance degradation with the increase in input length. Whereas, the performance of GRU drops significantly with longer text inputs.

Table 3 shows the result of the comparison of our model with various other models using publicly available sentence classification datasets. These results illustrate that our proposed model either performed comparable to or outperformed existing models.

6 Discussion

When we look at the results illustrated in Table 2, the performance of the proposed CNN-LMTGRU increased significantly compared to CNN-GRU. As shown in Eq. (4), we know that if \( \tau \) is close to 1, which is the case of a fast LMTGRU layer, the model becomes a vanilla GRU. Therefore, a vanilla GRU is considered as a fast layer and hence, a CNN-GRU network can be considered as a network with only fast units. The difference in performance when we have all the RNN units as fast, i.e. CNN-GRU, and when we have a combination of slow and fast units, i.e. CNN-LMTGRU, show the effectiveness of the multiple timescale approach. The results in Figure 4 also show the significance of the features from slow and fast layers, where the fast features helps maintain the performance with shorter text inputs and the slow features enable the model to perform significantly better with longer text inputs. This confirms our hypotheses that the proposed LMTGRU with the help of both slow and fast units can help encode different dynamic features in order to help classify the sentences and utterances correctly. The results indicate that the LMTGRU architecture increases the capability of the model to learn multiple temporal dependencies better for the discrimination task. The results also demonstrate that our hybrid CNN-LMTGRU network performs significantly better than the existing hybrid models.

The results in Table 3 shows that our model performed fairly comparable to the baseline models. The enhanced performance of the proposed model in both SST-2 (average length of 19 words) and MPQA (average length of 3 words) over the baseline models also confirms our hypothesis that the rich features of the slow and fast layers help in the discrimination task even with diverse sequence lengths. However, for some of the datasets such as TREC, our end-to-end learning model cannot outperform the conventional models like SVM due to the limited size of the dataset.

The increased ability of the proposed model to
Table 3: Results of our CNN-LMTGRU model against other methods on various sentence classification benchmark datasets.

<table>
<thead>
<tr>
<th>Model</th>
<th>MR</th>
<th>SST-1</th>
<th>SST-2</th>
<th>Subj</th>
<th>TREC</th>
<th>CR</th>
<th>MPQA</th>
</tr>
</thead>
<tbody>
<tr>
<td>CNN-static (Kim, 2014)</td>
<td>81.0</td>
<td>45.5</td>
<td>86.8</td>
<td>93.0</td>
<td>92.8</td>
<td>84.7</td>
<td>89.6</td>
</tr>
<tr>
<td>CNN-non-static (Kim, 2014)</td>
<td>81.5</td>
<td>48.0</td>
<td>87.2</td>
<td>93.4</td>
<td>93.6</td>
<td>84.3</td>
<td>89.5</td>
</tr>
<tr>
<td>CNN-multichannel (Kim, 2014)</td>
<td>81.1</td>
<td>47.4</td>
<td>88.1</td>
<td>93.2</td>
<td>92.2</td>
<td>85.0</td>
<td>89.4</td>
</tr>
<tr>
<td>RAE (Socher et al., 2011)</td>
<td>77.7</td>
<td>43.2</td>
<td>82.4</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>86.4</td>
</tr>
<tr>
<td>MV-RNN (Socher et al., 2012)</td>
<td>79.0</td>
<td>44.4</td>
<td>82.9</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>−</td>
</tr>
<tr>
<td>RNTN (Socher et al., 2013)</td>
<td>−</td>
<td>45.7</td>
<td>85.4</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>−</td>
</tr>
<tr>
<td>DCNN (Kalchbrenner et al., 2014)</td>
<td>−</td>
<td>48.5</td>
<td>86.8</td>
<td>−</td>
<td>93.0</td>
<td>−</td>
<td>−</td>
</tr>
<tr>
<td>Paragraph-Vec (Le and Mikolov, 2014)</td>
<td>−</td>
<td>48.7</td>
<td>87.8</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>−</td>
</tr>
<tr>
<td>CCAE (Hermann and Blunsom, 2013)</td>
<td>77.8</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>87.2</td>
</tr>
<tr>
<td>Sent-Parser (Dong et al., 2015)</td>
<td>79.5</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>86.3</td>
</tr>
<tr>
<td>NBSVM (Wang and Manning, 2012)</td>
<td>79.4</td>
<td>−</td>
<td>−</td>
<td>93.2</td>
<td>−</td>
<td>81.8</td>
<td>86.3</td>
</tr>
<tr>
<td>MNB (Wang and Manning, 2012)</td>
<td>79.0</td>
<td>−</td>
<td>−</td>
<td>93.6</td>
<td>−</td>
<td>80.0</td>
<td>86.3</td>
</tr>
<tr>
<td>G-Dropout (Wang and Manning, 2013)</td>
<td>79.0</td>
<td>−</td>
<td>−</td>
<td>93.4</td>
<td>−</td>
<td>82.1</td>
<td>86.1</td>
</tr>
<tr>
<td>F-Dropout (Wang and Manning, 2013)</td>
<td>79.1</td>
<td>−</td>
<td>−</td>
<td>93.6</td>
<td>−</td>
<td>81.9</td>
<td>86.3</td>
</tr>
<tr>
<td>Tree-CRF (Nakagawa et al., 2010)</td>
<td>77.3</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>81.4</td>
<td>86.1</td>
</tr>
<tr>
<td>CRF-PR (Yang and Cardie, 2014)</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>82.7</td>
<td>−</td>
</tr>
<tr>
<td>SVM_{S} (Silva et al., 2011)</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>−</td>
<td>95.0</td>
<td>−</td>
<td>−</td>
</tr>
<tr>
<td>Proposed CNN-LMTGRU</td>
<td>80.9</td>
<td>48.4</td>
<td>89.4</td>
<td>93.4</td>
<td>93.8</td>
<td>84.8</td>
<td>90.8</td>
</tr>
</tbody>
</table>

discriminate between open-domain chit-chat conversations and domain-specific task-oriented utterances will definitely help in the development of hybrid intelligent dialog systems that can handle both types of conversation. Moreover, with the help of this kind of classifier, the chat agents can dynamically switch between utterances in order to conduct a more natural and intelligent conversation with the users.

7 Conclusion and Future Work

This paper addressed the issue of discriminating conversations for combining domain-specific task-oriented agents and open-domain chit-chat agents. We developed a hybrid model consisting of a CNN and an LMTGRU network to classify the conversations. The proposed LMTGRU was able to effectively determine the type of conversation that a user will have with a dialog system. Moreover, we addressed the lack of dataset by constructing a dataset with chit-chat conversations and a task-oriented conversation corpus. We also evaluated the performance of the proposed hybrid model on various benchmark sentence classification datasets in order to compare to several existing models. The results of our experiments illustrated that the proposed end-to-end learning hybrid network with multiple timescales not only performed significantly better in case of longer texts inputs but also maintained good performance in case of shorter texts.

In the future, we plan to develop a more sophisticated dialog discrimination model to handle user utterances that are ambiguous in nature. It will be difficult for the standard classifiers to determine the actual type of conversation in such cases. One of the possible solution is to instruct the chat agent to follow up with clarification questions in case of ambiguity (Schlöder and Fernández, 2015). Another solution is to utilize contextual information by using previous dialogs from the system (Xu and Sarikaya, 2014). We plan to integrate features from the previous utterances for classification. This can be achieved by integrating the lateral architecture of an LMTGRU and the hierarchical organization of MTGRU along with the CNN features from the current and previous utterances to make the decision.

Although the studies on conversational agents have made significant progress in the recent years, it is still difficult for the systems to have a fluent conversation with the users (Higashinaka et al., 2015). We further plan to utilize the chat discrimination model to develop a hybrid system in order to improve such dialog agents. This will also allow us to evaluate the effectiveness of our model.
Acknowledgments

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Text Completion using a Context-Integrating Dependency Parser

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Abstract

Incomplete linguistic input, i.e. due to a noisy environment, is one of the challenges that a successful communication system has to deal with. In this paper, we study text completion with a data set composed of sentences with gaps where a successful completion cannot be achieved through a uni-modal (language-based) approach. We present a solution based on a context-integrating dependency parser incorporating an additional non-linguistic modality. An incompleteness in one channel is compensated by information from another one and the parser learns the association between the two modalities from a multiple level knowledge representation. We examined several model variations by adjusting the degree of influence of different modalities in the decision making on possible filler words and their exact reference to a non-linguistic context element. Our model is able to fill the gap with 95.4% word and 95.2% exact reference accuracy hence the successful prediction can be achieved not only on the word level (such as mug) but also with respect to the correct identification of its context reference (such as mug_2 among several mug instances).

1 Introduction

Text completion/prediction is a crucial element of communication systems, due to its role in increasing the fluency and the effectiveness of the communication in scenarios where the environment is noisy, or the communication partner suffers from a motor, or cognitive impairment (Garay-Vitoria and Abascal, 2004). In this study, we tackle the problem of compensating the incompleteness of the verbal channel by additional information from visual modality. This capability for multi-modal integration can be a very specific yet crucial feature in resolving references and/or performing commands for i.e. a helper robot that aids people in their daily activities. To the authors’ knowledge, there is no multi-modal data set for a text completion task that systematically addresses challenging linguistic structures (i.e. syntactic or referential ambiguities) for environments where helper robots, who have access to visual information, would be employed.

The completion is performed by predicting tenable fillers for the missing, unknown, or vague parts in the input sentences through varying techniques, using single or hybrid methods. The prediction process utilizes the available resources, usually linguistic information (morphological, syntactic, and semantic properties). It can also use additional information sources such as the linguistic, or visual context (Garay-Vitoria and Abascal, 2006). If only the linguistic level is available, a language model can be used to predict the probability of a syntactic category in a certain context (Asnani et al., 2015; Bickel et al., 2005). N-grams is a popular method for this task since they provide very robust predictions for local dependencies. Nevertheless, they loose their power for structures with long-range dependencies. Furthermore, if there are multiple instances of the same object class (c.f. Figure 1), a text completion based on N-gram could not differentiate between them to select the proper instance reference. As shown in several studies (Mirowski and Vlachos, 2015; Gubbins and Vlachos, 2013), a language model employing the syntactic dependencies of a sentence brings the relevant contexts...
closer. Using the Microsoft Research Sentence Completion Challenge (Zweig and Burges, 2012), Gubbins and Vlachos (2013) have showed that incorporating syntactic information leads to grammatically better options for a semantic text completion task.

On the other hand, semantic clustering or classification (like in ontologies) can be used to derive predictions on the semantic level. However, when it comes to the description of daily activities, contextual information coming from another modality would be more beneficial, since linguistic distributions alone could hardly contribute enough clues to distinguish the action of washing a pan from washing a mug, which is a crucial difference for helper robots.

A popular trick in natural language processing consists in training a model on one task, and then apply it to an entirely different one. We adopt this method by training a multi-modal dependency parser using noise-free sentences combined with a description of their visual context. In the second step, we make use of the trained parser to predict the best fillers of the gaps (guided by the context modality).

The paper starts by introducing our multi-modal approach for the text completion task. In section 3, we present the experimental setup including the compiled dataset. The implementation is described in section 4. Experimental results are presented and discussed in section 5. Conclusions are drawn and future directions of research are pointed out at the end of the paper.

2 A Multi-Modal Approach for a Text Completion Task

Although closing the gaps in a sentence based only on a language model is a simple way to tackle the issue, in extremely ambiguous situations, gap reconstruction is almost impossible on a purely unimodal base. In this paper, we work on multi-modal data that consists of linguistic and context information. The linguistic part is provided by natural language sentences that refer to a particular visual scene. The context information is a meta-data description of that scene. Per input sentence, the context channel contains a set of context relations: \((argument, relation\_type, predicate)\) where \(relation\_type\) is one of a predefined set of accepted relations, such as agent or theme while \(Predicate\) and \(Argument\) are tokens of the input sentence. The complexity of the text completion task is controlled by creating challenging scenes along the following dimensions:

- Each scene is composed of different components (i.e., persons and objects).
- A scene might contain multiple instances of the same class (i.e., a blue mug (id: \textit{mug}$_1$) and a green mug (id: \textit{mug}$_2$).
- The different instances are taking part in various relations (more details are given in Section 2.2).

In a series of experiments, we assess the potential of a context-integrating dependency parser for correctly solving the text completion task. We not only try to determine whether we can fill the gap in the sentence with the correct word but also whether it is possible to correctly determine the exact reference to an entity in the context description given the contextual information, in particular if the linguistic input is noisy and a token of the input sentence is missing. At this stage of research, we work only on one gap per sentence.

2.1 Context-integrating Dependency Parser

Dependency parsing is an essential NLP task that determines the syntactic structure of the input sentence in form of a dependency tree. Each token of the input is represented as a tree node. The tree consists of the dependency relations between each word of the sentence and its head word (Nivre, 2004).

The standard input of a parser is a natural language sentence. To supply such a parser with additional information required for text completion in a multi-modal environment we have to make it sensitive to cues from the context.

In our previous research (Salama and Menzel, 2018, 2017), we have introduced a multi-modal dependency parser adopting the graph-based approach of Eisner (1996) and Mcdonald and Pereira (2006). Our model, called RBG-2, extends the RBG parser (Zhang et al., 2014) by enabling multi-channel input providing the parsing process with context information in addition to the natural language sentence. Integration is achieved by combining features from both input channels during the normal training procedure of the RBG parser.
2.2 The Data Set

In order to test how the model behaves for different linguistic structures, we used the nine different grammatical templates\(^1\) given in Table 1 featuring active/passive voice, PP-attachments, relative clause (RC) attachments, and conjunctions. They are combined with several actions performed by different agents. The dependency structures are represented in the CONLL-X format. The data set consists of 429 individual sentences for 20 different visual scenes. We performed a 10-fold cross validation and introduced exactly one gap for either a noun, verb or adjective into each test sentence obtaining 1457 test sentences in total.

2.2.1 Linguistic Structures

In this section, we exemplify the nine grammatical templates used in our data-set. The following examples belong to the scene in Figure 1:

- **T1. RC\(^2\)** Attachment Ambiguity-1
  
  \(T1A.\) Active voice in RC. “It is a mug on a vitrine that the woman damages.”
  
  Either the relative clause is low-attached (the woman damages the vitrine) or high-attached (the woman damages the mug).
  
  \(T1B.\) Passive voice in RC.“It is a mug on a vitrine that is damaged by the woman.”

- **T2. RC Attachment Ambiguity-2**
  
  \(T2A.\) Active voice. “The woman damages the vitrine with a mug on it.”
  
  \(T2B.\) Passive voice. “The vitrine with a mug on it is damaged by the woman.”

- **T3. RC Attachment Ambiguity with a Genitive Object-3**
  
  \(T3A.\) Active voice in RC. “The woman removes the label of the medicine that lies on the shelf”
  
  \(T3B.\) Passive voice in RC. “The label of the medicine that lies on the shelf is removed by the woman.”

- **T4. Scope Ambiguity**
  
  “There are a mug, a candle and books [that lie/lying] on the vitrine.”

- **T5. Simple Imperative sentence**
  
  “bring me the mug [that lies/lying] on the vitrine [that the woman cleans].”

- **T6. Imperative sentence with modifiers**
  
  “bring me the blue mug [that lies/lying] on the vitrine [that the woman cleans].”

2.2.2 Context Representations

The visual information of a picture is represented in a knowledge base that contains the relationships between objects, characters and actions in the scene. This information has been manually annotated as triplets composed of argument, relation type and predicate. Currently, we consider six different context relations, namely agent, theme, location, next-to, part-of/own, as well as property assignments for color, material, shape etc. (e.g., a blue mug or a ceramic vase). Figure 1 exemplifies the context annotations of a visual scene with an additional concept map representation (bottom). In this scene, the woman is the agent, who performs the cleaning action, the vitrine is the theme, i.e. the entity undergoing a change of state, caused by the action. The entire data set and source code can be accessed from https://github.com/rekaby/MD-TC.V1.0

For the current study, the pictures, as the one given in Figure 1, serve illustrative purposes, because the computational model does only have access to the manually annotated representations. An automatic relation extraction is not within the scope of this study.

The different semantic roles are distributed in the data set as follows: Agent (%13.6), Theme (%13.6), Location (%33.1), Next to (%9.8), Property (%19.5), Own (%10.3). Table 2 presents a statistics for the amount of contextual information per scene.

3 Implementation

RBG-2 parser starts by creating a fully connected graph representing the input tokens as nodes. The parser decodes a minimum spanning tree out of the graph maximizing the aggregated scores of the arcs. The scores are calculated by combining the weights of linguistic features and context features between the pair of tokens as follows:

\[
g = \max_{y \in T(x,c)} \sum_{i=1}^{n} \omega_i f(x_i, x_j, y) + \omega_c f(c_i, c_j, y)
\]

(1)

Where \(g\) is the best dependency tree, \(T(x,c)\) is a set of all possible dependency trees for input sentence \(X\) and context \(c\). The linguistic feature vector between node \(x_i\) and its dependency head \(x_j\) is

\(^1\) inspired by the experimental setup of psycholinguistic research

\(^2\) Relative Clause
Table 1: POS templates, the number of sentences and gaps for each sentence types, and the number of gaps for each POS category

<table>
<thead>
<tr>
<th>Types</th>
<th>Templates</th>
<th>Sentences</th>
<th>Gaps</th>
<th>Gap Types</th>
</tr>
</thead>
<tbody>
<tr>
<td>T1A</td>
<td>PRO1nom  VP1 NP1acc NP2gen, WDT*acc PRO2nom VP2</td>
<td>41</td>
<td>130</td>
<td>NP(114), VP(16)</td>
</tr>
<tr>
<td>T1B</td>
<td>PRO1nom  VP1 NP1acc NP2gen, WDT*acc PRO2nom VP2</td>
<td>40</td>
<td>130</td>
<td>NP(115), VP(15)</td>
</tr>
<tr>
<td>T2A</td>
<td>PRO1nom  VP1 NP1nom.pl NP2nom.pl, WDT acc, VP2 PP1</td>
<td>50</td>
<td>177</td>
<td>NP(153), VP(24)</td>
</tr>
<tr>
<td>T2B</td>
<td>PRO1nom  VP1 NP1nom.pl NP2nom.pl, WDT acc, VP2 PP2</td>
<td>50</td>
<td>177</td>
<td>NP(153), VP(24)</td>
</tr>
<tr>
<td>T3A</td>
<td>NPn-clom VP1 NP1nom NP2dat, WDT dat PRO1dat, PRO2dat, ADV VP2</td>
<td>36</td>
<td>177</td>
<td>NP(139), VP(36), ADJ(2)</td>
</tr>
<tr>
<td>T3B</td>
<td>NPn-clom VP1 NP1nom NP2dat, WDT dat PRO1dat, PRO2dat, ADV VP2</td>
<td>29</td>
<td>145</td>
<td>NP(113), VP(30), ADJ(2)</td>
</tr>
<tr>
<td>T4</td>
<td>EX V natt NP1nom (CONJ NP2) WDT* nom VP1 Prep. NP3</td>
<td>38</td>
<td>156</td>
<td>NP(154), VP(60), ADJ(2)</td>
</tr>
<tr>
<td>T5</td>
<td>VP1 (PRO1nom (WDT) VP2 Prep. NP2)</td>
<td>63</td>
<td>141</td>
<td>NP(137), VP(1), ADJ(3)</td>
</tr>
<tr>
<td>T6</td>
<td>VP1 (PRO1nom (Adj1)) NP1 (WDT1) VP2 Prep. (ADJ2) NP2</td>
<td>62</td>
<td>224</td>
<td>NP(135), VP(9), ADJ(80)</td>
</tr>
<tr>
<td><strong>Total</strong></td>
<td></td>
<td><strong>429</strong></td>
<td><strong>1457</strong></td>
<td>NP(1213), VP(155), ADJ(89)</td>
</tr>
</tbody>
</table>

Table 2: Complexity of the contextual information for the visual scenes in the data set

<table>
<thead>
<tr>
<th>Items</th>
<th>Mean</th>
</tr>
</thead>
<tbody>
<tr>
<td>Relations (Min.=28, Max.=41)</td>
<td>34.8</td>
</tr>
<tr>
<td>Context entities (Min.=30, Max.=41)</td>
<td>35.6</td>
</tr>
<tr>
<td>Unique entities * (Min.=21, Max.=28)</td>
<td>25</td>
</tr>
</tbody>
</table>

Algorithm 1 Text Completion Workflow using RBG-2

\[
TR-L \leftarrow \text{Training data (complete sentences)}.
\]

\[
TR-C \leftarrow \text{Training data (context)}.
\]

\[
TE-L \leftarrow \text{Testing data (sentences with gaps)}.
\]

\[
TE-C \leftarrow \text{Testing data (context)}.
\]

\[
model \leftarrow \text{train RBG-2(TR-L, TR-C)}.
\]

for each pair TE-L_{i},TE-C_{i} do

\[
\text{bestFillerScore} \leftarrow -\infty.
\]

for each component TE-C_{ij} \in TE-C_{i} do

for each POS_{i} \in POS tags do

\[
\text{TE-L}_{i} \leftarrow \text{SetGap(TE-C_{ij},POS_{i})}.
\]

\[
\text{score} \leftarrow \text{parse(model,TE-L_{i})}.
\]

if score > bestFillerScore then

\[
\text{bestFillerScore} \leftarrow \text{score}.
\]

\[
\text{bestFiller} \leftarrow \text{TE-C}_{ij}.
\]

\[
\text{bestPOS} \leftarrow \text{POS}_{i}.
\]

\[
\text{TE-L}_{i} \leftarrow \text{fillGap(bestFiller,bestPOS)}.
\]

- the filler word (woman),
- the context filler reference (woman)_2,
- the context filler reference for all the other non-gap tokens in the input (if they exist). They are \{vitrine\_2, sofa\_1, clean\_1\},
- the POS tag of the filler (NN).

As shown in Algorithm 1, we train our data-driven RGB-2 parser on the multi-modal training set described above to learn the associations between the context knowledge representation and the dependency structures. In the testing phase, we fill the gap by all the possible context components and parse the sentence in a multi-modal setup. We also iterate over different POS tags for the filler to compare the resulting dependency tree scores. The best filler (word, context-reference, and POS) means that this word/context-reference
Figure 1: The corresponding image for the sentences above and the semantic representations of the actions and relations in the image.

is the best matching one that combines two perspectives: grammatical correctness and compatibility with the context information.

Although the ratio of contextual features to syntactic ones (first-order features) is 1:2.3, which is not high, trying all the possible context elements is rather expensive. For each sentence, we need to build $G*C*P*M$ dependency trees that have to be ranked to find the best one. Here, $G$ is the number of gaps (1 in our experiments), $C$ the number of context entities (35.6 in average), $P$ the number of PoS tags (3) and $M = \prod_{i=1}^{N} M_i$, where $M_i$ is the count of possible candidates references and $N$ the number of sentence tokens with probable context references.

The search space could be reduced by avoiding irregular combinations of POS and filler words. In this stage of research, however, we do not prune it at all.

3.1 Context Data Preprocessing

In a preprocessing phase, we enrich the context information by inferring new relations from the original ones (colored red in Figure 1 and Figure 2). We have used two kinds of inferred relations:

**Location to Agent/Theme:** If we have a context relation such as $(X, Location, Y)$, this might appear in the linguistic modality in two different forms having either direct or indirect syntactic dependency. For example, *mug* and *vitrine* as in Figure 2A and 2B have a direct syntactic dependency and context relation respectively. In other sentence forms as in Figure 2C and 2D, there is no direct correspondence between the linguistic dependency and the context relation. Contextually, the two tokens are related through the *Location* relation, but syntactically they are daughters of the same action *lie* (no direct dependency). In this form, the *Location* relation is presented in the lin-
To enrich the context representation with information corresponding directly to the linguistic one, we define a set of verbs (LV) that have a location meaning (i.e., lie, stand, hang). From any location relation \((X, Location, Y)\), we infer another two relations \((X, \text{Agent}, LV_i)\) and \((X, \text{Theme}, LV_i)\), where \(LV_i \in LV\) and \(LV_i\) is a token in the input sentence.

**Location to Next To relations:** Given each pair of location relations \((X, Location, Z)\) and \((Y, Location, Z)\) we infer new relation \((X, \text{Next} - \text{To}, Y)\), where \(X, Y, Z \in W\), \(W\) is the set of the input tokens. The inferred relations are added to the original list of the context input. In the rest of this paper, we use (IC) to refer to the Inferred Context relations and (OC) for the Original Context relations.

### 3.2 Model Variations

**Varying syntactic/context’s weight ratios (S2C):** In the testing phase, we experiment with different ratios giving more influence (weight) to the context relations than to the linguistic ones. We assess different ratios (1to1, 1to5, 1to10, and 1to25).

**Original/Inferred relations’ weight ratio (OC2IC):** Similar to S2C, in the testing phase, we give more weight to the original relations than to the inferred ones by assigning the OC2IC ratio to 5to1.

### 4 Results

In order to show the effect of contextual information and to optimize the performance of the current model, we carried out several experiments with different parameters of the model by keeping the data set constant. We used 18 scenes (386 sentences in average) for training and kept the remaining 2 scenes (146 sentences on average) for test using a 10-fold cross-validation. In case the gap can be filled with more than one reference (< 5% of our dataset), we consider any possible one of them as correct. We used five evaluation metrics as listed below.

- POS-tag Accuracy
- Filler Word Accuracy
- Exact Filler Identification (EFI) Accuracy (i.e, \(mug_1\) in contrast to \(mug_2\))
- Non-gap Identification Accuracy, for all the other tokens in the input sentence.
- Complete Sentence Identification Accuracy
- Dependency Tree Accuracy (unlabeled attachment score, UAS)

Table 3 presents the results obtained from different variations of the model described in the previous section. We test a uni-modal parser (linguistic-only) only to show that the data set indeed is consisting of sentences, where reference resolution/text completion cannot be achieved on a purely uni-modal sense. For that purpose, the contribution of contextual information is turned off. Because of the uniform structure of the training dataset, the POS and dependency tree accuracies are very high 97.6% and 95.6% respectively. However, the model’s prediction performance is drastically low for the gap words; 13.5% for the filler word and 7.8% for the exact filler identification.

As described in Section 3.2, the first model is based on having equal weights (S2C-1to1) for both syntactic (S) and contextual features (C) and the weights of original contextual (OC) features to the inferred features (IC) are kept equal as well (OC2IC-1to1). Giving equal weights leads to approx. 83% accuracy in both filler word and exact filler ID predictions, while increasing the influence of the context resulted in 95% accuracy.\(^6\)

\(^6\)The other models with weights > 5 produced almost similar results.
Furthermore, giving more weight to the original relations over the inferred ones resulted in lower accuracy, therefore OCTolC-1to1 variation is chosen as the standard for the analysis in this section. It is apparent from Table 3, a higher influence of the context is beneficial for a correct reference prediction. However, it should be noted that giving more weight to contextual features causes the model to be less sensitive about choosing a correct dependency tree. A closer look at the differences between the predictions of the S2C-1to5 and S2C-1to10 variations showed that 60 instances either in the dependency tree or in the filler ID were observed in the results. While S2C-1to5 builds 51 correct dependency trees and 43 correct references, S2C-1to10 chooses the correct dependency tree in only 12 instances, but even if the dependency tree is wrong, it fills the gap correctly in 48 out of 60 instances.

95 inaccurate EFI in 73 test sentences were observed. False predictions of the model variations can be categorized into several groups:

**Inferred Relations.** 60% of the inaccurate predictions occurred within this category. As explained in the Section 3.1, a phrase like “an entity-1 that lies on an entity-2” can be resolved due to an inferred relation. However, for sentences containing structures like “an entity-1 that lie/stand/hang(s) next to an entity-2” with a gap in a position of entity-2, the model prefers the most plausible filler that has a location relation (either original or inferred) with the entity-1 instead of having a next to relation with it.

**A Chain of Relations.** This problem arises when for example there is a chain of location relations among the entities (7.4%), i.e. (bird$_1$, Location, cage$_1$), and (cage$_1$, Location, chest$_1$) with a description “It is a cage on a chest that the man cleans” with a gap in a chest position. While the S2C-1to5 model correctly fills the gap, S2C-1to10 chooses bird for the gap position. Assigning more weight to the context information leads to similar scores for the various entities of the chain, which may cause some wrong filler predictions.

**Less represented PP associations.** Syntactically, all prepositions (with, of, on and next to) have the same PoS tag but semantically they differ. While preposition of is associated with the own relation, and preposition next to with next-to, there are two prepositions which are related to the location relation: on/in and with. The distribution of them is as follows; with: 21.3%, and on/in/under: 78.7%. As shown in Figure 3, the most likely association between syntactic and contextual features (w.r.t. location relations) is head to argument and dependent to predicate. This association is flipped for the prepositional phrase like “entity-2 with entity-1 on it”. Regardless of giving more influence to the context in that case, the model makes the prediction more strongly biased to the canonical direction of prepositional phrases resulting a wrong text prediction.

**A Verb in a Noun Position.** This error occurs irrespective of the linguistic structure if more weight (1to10 or 1to25) is given to the context (6.3%). As an example, a gap in the shelf position in the sentence “There are a cat, a flower and books on the shelf” is filled with chase, caused by the (cat$_1$, Theme, chase$_1$) relation. In that case, a stronger contextual influence overrides the syntactic form of the PP-attachment, and favors a reference with the theme relation, which has a consistent syntactic representation; its argument always points to the predicate. The goal to find syntactically correct PP-attachment is overruled by the more powerful features of the context relations, and so chase is selected considering that a cat is the only entity with a theme relation among others.

**Far-Attachments.** Far-attachments of the relative clauses or prepositional phrases are not that frequent as short-attachments, yet they are grammatically correct and occur in a data set. The results indicate that giving more influence to contextual information (S2C-1to10 and 1to25) helps to correctly fill the gap, while a model with lower weight for the contextual information (S2C-1to1 and -1to5) tends to choose the wrong reference for the gap position. To illustrate, the sentence “It is a blanket on a couch that is grasped by the woman” refers to one instance of a blanket class, and the context contains two instances: blanket$_1$ and blanket$_2$, where blanket$_2$ is the theme of the grasp action. When the gap is in the couch position, S2C-1to5 chooses a dependency tree with a short attachment of the RC. It attaches the gap to the action grasp and thus fills it with blanket$_2$. This is consistent with the theme relation in the context, resulting a sentence “It is a blanket$_1$ on

---

$^{2}$excluding the occurrences of “on/in/under” in the reflexive phrases as in “with a mug on it”
Table 3: The results of the different model variations

<table>
<thead>
<tr>
<th>Model Variations</th>
<th>PoS</th>
<th>Filler Word</th>
<th>EFI</th>
<th>Non-Gap</th>
<th>Complete Sentence</th>
<th>DP-UAS</th>
</tr>
</thead>
<tbody>
<tr>
<td>Uni-model (linguistic only)</td>
<td>97.58</td>
<td>13.50</td>
<td>7.75</td>
<td>62.05</td>
<td>2.21</td>
<td>95.60</td>
</tr>
<tr>
<td>S2C Weight (1to1) + OC2IC Weight (1to1)</td>
<td>98.34</td>
<td>83.53</td>
<td>83.11</td>
<td>97.35</td>
<td>83.60</td>
<td>95.60</td>
</tr>
<tr>
<td>S2C Weight (1to5) + OC2IC Weight (1to1)</td>
<td>98.89</td>
<td>95.36</td>
<td>95.22</td>
<td>99.24</td>
<td>94.67</td>
<td>95.11</td>
</tr>
<tr>
<td>S2C Weight (1to10) + OC2IC Weight (1to1)</td>
<td>98.48</td>
<td>95.57</td>
<td>95.50</td>
<td>99.36</td>
<td>94.81</td>
<td>94.80</td>
</tr>
<tr>
<td>S2C Weight (1to25) + OC2IC Weight (1to1)</td>
<td>98.13</td>
<td>95.57</td>
<td>95.50</td>
<td>99.36</td>
<td>94.81</td>
<td>94.51</td>
</tr>
<tr>
<td>S2C Weight (1to5) + OC2IC Weight (5to1)</td>
<td>98.82</td>
<td>92.39</td>
<td>92.32</td>
<td>98.85</td>
<td>91.76</td>
<td>95.05</td>
</tr>
</tbody>
</table>

Figure 3: Syntactic/Context feature association for the prepositions on and with

A blanket_2 that is grasped by the woman^8. If the context had only one blanket, that instance of the blanket had to be assigned to a non-gap blanket position in the sentence, and then the model is forced to switch to another dependency tree with a lower score but a better alignment. On the other hand, a 1to10 model gives more influence to the context, resulting in a correct completion even if the dependency structure is wrong. This may indicate that in order to deal with more challenging contexts or less represented linguistic structures (like far-attachments) increasing the influence of the contextual information would be beneficial.

Contextually Challenging Cases. This category covers 10.5% of the errors. To illustrate, lets consider a context, which contains two different roles for the same agent man_1 together with a sentence like “The handle of the mug on the counter is hold by the man”; namely an action wash with a theme relation to a mug and another action hold with a theme relation to a handle. Another relevant relation for this sentence is (mug_1, Own, handle_1). If in such a case the gap is in the verb position, the model can choose the alternative actions associated with a mug instead of forcing a far-attachment which is also favored contextually.

5 Conclusion and Future Directions

In this paper, we present a data set for sentence completion consisting of problematic instances, which can not be effectively handled using linguistic features alone. We apply a context-integrating dependency parser to solve this problem. There are number of assumptions and constraints of the current model. First, allowed gaps are only nouns, verbs and adjectives. Pronouns are not used as a possible gap filler. Furthermore, each individual instance is allowed to occur in the sentence once^9, thus a context reference (i.e. mug_1) can not be assigned to more than one token of the input sentence. Moreover, the set of context relations is restricted to the six relations. Further studies will need to cover more variety to relax these limitations.

The results indicate that incorporating contextual information and giving a strong enough influence to them helps to solve a majority of the problems concerning different sentence structures with conjunctions, relative clauses or PP-attachments. There are still some challenging situations originating from high degrees of linguistic or contextual complexity, which need to be addressed in future work. Furthermore, we plan to address noisier linguistic input with multiple gaps in a sentence as well as mismatches between the sentence and its contextual information. We also target reference resolution at the earliest time possible by employing incremental processing.

Acknowledgments

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^8Our assumption is not using same context reference twice

^9This constraint is not based on linguistic phenomena, it is just the design decision for the current solution
References


Quantum-inspired Complex Word Embedding

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Abstract

A challenging task for word embeddings is to capture the emergent meaning or polarity of a combination of individual words. For example, existing approaches in word embeddings will assign high probabilities to the words "Penguin" and "Fly" if they frequently co-occur, but it fails to capture the fact that they occur in an opposite sense - Penguins do not fly. We hypothesize that humans do not associate a single polarity or sentiment to each word. The word contributes to the overall polarity of a combination of words depending upon which other words it is combined with. This is analogous to the behavior of microscopic particles which exist in all possible states at the same time and interfere with each other to give rise to new states depending upon their relative phases. We make use of the Hilbert Space representation of such particles in Quantum Mechanics where we subscribe a relative phase to each word, which is a complex number, and investigate two such quantum inspired models to derive the meaning of a combination of words. The proposed models 1 achieve better performances than state-of-the-art non-quantum models on the binary sentence classification task.

1 Introduction

Word embeddings (Bengio et al., 2003; Mikolov et al., 2013; Pennington et al., 2014) are the current state of art techniques to form semantic representations of words based on their contexts. They have been successfully used in various downstream tasks such as text classification, text generation, etc. Building on word embeddings, various unsupervised (Kiros et al., 2015; Hill et al., 2016a) and supervised (Conneau et al., 2017) models for sentence embeddings have been proposed. The general idea behind word embeddings is to use word co-occurrence as the basis of semantic relationship between words. This naturally brings about the difficulty for word embedding approaches in capturing the emergent meaning of a combination of words, such as a phrase or a sentence. For example, the phrase "ivory tower" can hardly be modeled as a semantic combination of "ivory" and "tower". Or, the high frequency of occurrence of the words "Penguin" and "Fly" fails to suggest that they are negative correlated.

In the field of information retrieval (IR), various models based on the mathematical framework of Quantum Theory have been applied to capture and represent dependencies between words (Sordoni et al., 2013; Xie et al., 2015; Zhang et al., 2018), inspired by the pioneering work of Van Rijsbergen (2004). Sordoni et al. (2013) models a segment of text as a quantum mixed state, represented by a positive semi-definite matrix called density matrix in a Hilbert Space, whose non-diagonal entries entail word relations in a quantum manner (Quantum Interference). The resulting Quantum Language Model (QLM) outperforms various classical models on ad-hoc retrieval tasks. Xie et al. (2015) captures Unconditional Pure Dependence (UPD) (Hou et al., 2013) between words in a quantum way by demonstrating the equivalence relation between UPD and Quantum Entan-
glement (QE) and providing a way to incorporate UPD information into QLM, leading to improved performance over the original QLM. Zhang et al. (2018) develops a well-performing question answering (QA) system by extracting various features and learning to compare the density matrices between a question and an answer.

The successful application of quantum-inspired models onto IR tasks (Wang et al., 2016) to some extent demonstrates the non-classical nature of word dependency relations. However, all these models simplify the space of interest to be space of real vectors $\mathbb{R}^n$, with the representation of a word or a text segment being a real-valued vector or matrix, largely due to the lack of proper textual features corresponding to the imaginary part. Since quantum phenomena cannot be faithfully expressed without complex numbers, these models are theoretically limited. In a recent work, Aerts et al. (2017) presents a theoretical quantum framework for modeling a collection of documents called QWeb, in which a concept is represented as a state in a Hilbert Space, and concept combination is represented as a superposition of the concept states. Under this framework, the complex phases of each concept have a natural correspondence to the extent of interference between concepts. However, the framework has not given rise to any applicable models onto IR or NLP tasks to the authors’ knowledge.

Inspired by the potential of quantum-inspired models to represent word relations, we seek to build quantum models to represent words and word combinations, and explore the use of complex numbers in the modeling process. Our model is built on top of two hypothesis: I) A word is a linear combination of latent concepts with complex weights. II) A combination of words is viewed as a complex combination of word states, either a superposition state or a mixed state. The first hypothesis agrees with QWeb, but here we concretize a concept in QWeb to be a word. The second hypothesis is an extension of both QWeb and the work by Zhang et al. (2018), because QWeb restricts a combination of concepts to be a superposition state while the work by Zhang et al. (2018) assumes that a sentence is a complex mixture of word projectors.

This study sets foot in sentence-level analysis, and treats a sentence as a combination of words. We intend to model a word as a quantum state containing two parts: amplitudes and complex phases, and expect to capture the low-level word co-occurrence information by the amplitudes, while using the phases to represent the emergent meaning or polarity when a word is combined with other words. We investigate on two models to represent the combination of words, either as a superposition of word states or as a mixture of word projectors. The effectiveness of the two models are evaluated on 5 benchmarking binary sentence classification datasets, and the results show that the mixture model outperforms state-of-the-art word embedding approaches.

The motivation behind this paper stems from an analogy with Quantum Physics. Consider the phrase "Penguins fly". If we model it along the lines of the famous double slit experiment in Quantum Physics, the two slits corresponds to human interpretation of words "Penguins" and "Fly" (Verb sense of Fly). When only one slit is open at a time, the waves corresponding to the individual word will go through the slit and register onto the screen. The screen is made of a set of polarity detectors judging opinion or sentiment polarities.

In Figure 1.a the human mind sees the word 'Penguins' alone and detects it as a neutral word with a very high probability. This is analogous to the double slit experiment with one slit open. The same is the case for the word 'Fly' considered in isolation. By classical logic, when the two words are taken together as a phrase 'Penguins fly', the human mind should assign a high probability of it being neutral again. However, we know that it is a false statement (Figure 1.c).

Different from classical representation, this study hypothesizes that the combination of words can be viewed as a superposition or complex mixture of quantum entities which gives rise to a new state. In this way, the emerging meaning or polarity of a combination of words will manifest in the interference between words, and be captured inherently in the density matrix representation. For example, two or more words having a neutral sense individually may combine to give a negative sense, just like the case in the analogy given above.

2 Hilbert Space Representation of Words and Sentences

The section introduces the proposed quantum framework for representing words and sentences.
Our research scope is currently limited to sentence and word level analysis. However, our proposed model is potentially capable of representing higher-level concepts such as paragraphs and documents, which we will investigate in the future. To be consistent with the quantum framework, we use Dirac notations, in which a unit vector $\vec{\mu}$ and its transpose $\vec{\mu}^T$ are denoted as ket $|u\rangle$ and bra $\langle u|$ respectively.

Suppose there are $n$ independent latent concepts in the text collection, we then model words and sentences as quantum concepts defined on an $n$-dimensional Hilbert Space $\mathbb{H}^n$, where latent concepts form a set of pure orthonormal states of the space. Using Dirac notations, the concepts are denoted as $\{|C_i\rangle\}_{i=1}^n$. Intuitively, latent concepts correspond to the contexts in which words are used.

Each word $t$ is modeled as a superposition state (Nielsen and Chuang, 2011) in the $n$-dimensional Hilbert Space $\mathbb{H}^n$. Equivalently, it can be viewed as a linear combination of $\{|C_i\rangle\}_{i=1}^n$ with complex weights, i.e. $|t\rangle = \sum_{k=1}^n e^{i\theta_k} w_k |C_k\rangle$, in which $\{w_i\}_{i=1}^n$ are real-valued amplitudes with $w_i > 0$ and $\sum_{i=1}^n w_i^2 = 1$, and $\theta_i \in [-\pi, \pi]$, $i = 1, 2, ..., n$ are the corresponding complex phases. This representation can be seen as a generalization of previous word embedding approaches (Bengio et al., 2003; Mikolov et al., 2013; Pennington et al., 2014) in that it can be regarded as a complex embedding with unitary length of word vectors. A word has many different contexts associated with it. For example, 'Penguin' is associated with 'Bird', 'Antarctica', 'Snow', etc. When a quantum particle(e.g. electron) is said to be in a superposition state, it exists in a new state(e.g. position) of all of its possible outcomes(at all positions) at the same time. A particular outcome is observed upon measurement. Similarly a word exists in all of its contexts at the same time and depending upon its interaction with other words in a combination, a particular context is materialized. Note that because of reduced dimensionality, the contexts are latent concepts.

A sentence is a non-classical combination of words. Since each word is a superposition of latent concepts, a sentence $s$ is also a non-classical combination of latent concepts $\{|C_i\rangle\}_{i=1}^n$. It is represented by a $n \times n$ density matrix $\rho$ which is positive semi-definite with unitary trace: $\rho \geq 0$, $\text{Tr}(\rho) = 1$. The real diagonal values of $\rho$ reflects the strength of concepts in the sentence, whereas the non-diagonal values encodes correlations between concepts in a quantum manner. The density ma-
trix can be computed from the word states either directly or through a training strategy.

Our proposed approach is related to but largely differs from Sordoni et al. (2013) and Zhang et al. (2018). Sordoni et al. (2013) models queries and documents as density matrices and provides a training method for constructing density matrices from texts. Zhang et al. (2018) directly computes the density matrix of a sentence and put it into an end-to-end neural network for handling the Question Answering (QA) task. Both works view a segment of texts as a mixed state (Nielsen and Chuang, 2011) and use real-valued density matrix as a representation. Our study also directly computes the sentence representation from the word superposition states. However, different from both works, our study explores on treating a sentence as either a strictly mixed state or a superposition state. In either case, it can be represented as a complex density matrix with complex values for non-diagonal entries.

On top of the obtained sentence representation, different quantum operations can be applied to achieve a particular NLP target at hand. For sentence classification tasks, one can perform projective measurements onto the sentence representation to determine the sentiment polarity; for sentence text similarity task, the amplitude of the inner product between a sentence pair may provide evidence for judging to what extent they are similar to each other. Projective measurements and inner products are methods to compute probabilities in Quantum Theory (Nielsen and Chuang, 2011).

3 Complex Embedding Network for Text Classification

In this paper, we build a complex embedding network for text classification on the basis of Hilbert Space representation for words and sentences. The end-to-end network accepts a sentence sequence as input and computes its classification label in the procedure shown by Figure 2:

The input one-hot sequence is passed through an embedding layer with a complex valued lookup table, which maps each word into a complex vector representing its superposition state, resulting in a sequence of complex embedding vectors. Then the density matrix of the sentence is computed from the complex embedding vectors. Finally, a square projection matrix takes control of the measurement. For any sentence state \( \rho \), the measurement probability is computed through Born’s rule (Born, 1926):

\[
p = Tr(P\rho)
\]

Where \( P \) is a projection matrix satisfying \( P^2 = P, P = P^T \). The value of \( p \) determines the class of this sentence. The lookup table determining the complex embedding for each word is learned by feeding the network with a sufficient number of training data.

The crucial step of the process falls on how to compute the sentence density matrix from the sequence of complex word embeddings. As no previous research has attempted to build complex networks for text classification task, we investigate on two approaches for this step:

I) A sentence is viewed as a linear combination of all word vectors in the sentence, i.e. \( |S\rangle = \sum_{l=1}^{m} \lambda_l |t_l\rangle \), with \( \sum_{l=1}^{m} \lambda_l = 1 \). Here \( \lambda_l \)s are real-valued weights indicating the relative degree of importance for each word in the sentence, and the state is divided by its 2-norm in order to guarantee it is a legal quantum state (i.e.,with unit length). The sentence is then a pure superposition state and the density matrix can be computed simply as \( \rho = |S\rangle\langle S| \).

II) A sentence is viewed as a classical mixture of the word states in the sentence, i.e. \( \rho = \sum_{l=1}^{m} \lambda_l |t_l\rangle\langle t_l| \), with \( \sum_{l=1}^{m} \lambda_l = 1 \). Here \( |t_l\rangle\langle t_l| \) is the density matrix representing the superposition state of a word \( t_l \). This equation guarantees the obtained \( \rho \) is a legal density matrix without any further normalization.

The constructed density matrix representing a sentence has real values for diagonal entries and non-zero complex values for non-diagonal entries. Intuitively, the diagonal entries tell us something about the distribution of latent concepts in the sentence, whereas the non-diagonal values entail information regarding the emergent meanings. Consider a very simple example where the complex phases represent positive, neutral or negative senses. Independently, both the words "Penguin" and "Fly" have neutral sense, \( \theta_P = \theta_F = 0 \). When they are combined together in a sentence, then sentence density matrix has a negative-phased complex value in the entry corresponding to them, i.e. \( \theta_{PF} < 0 \). Therefore, the combination of these two words will have a negative complex phase, implying the negative sense "Penguins cannot fly".
In practice, the connections between words are much more complicated, but we believe that by feeding the above-mentioned models with enough data, the constructed density matrix will be able to effectively capture and represent the emergent meanings of sentences.

The above-mentioned approaches lead to two different models, resulting in different embeddings learned from the same training data. Hence we name them as complex embedding superposition (CE-Sup) network and complex embedding mixture (CE-Mix) network respectively. For sake of simplicity, we assign equal importance of each word in the sentence representation in both models, i.e. \( \lambda_l = \frac{1}{m}, l = 1, 2, ..., m \). In a relevant research, Zhang et al. (2018) learns the values of \( \lambda_l \)'s in the training framework, while enforcing the word embeddings \( |t_i| \)'s to be fixed. By fixing \( \lambda_l \)'s and learning \( |t_i| \)'s from the data, this paper is essentially aiming at obtaining better representation of each word from the training data, whereas Zhang et al.’s work directly takes existing word vectors trained from external corpus. It would be interesting to see what a co-training of \( |t_i| \)'s and \( \lambda_l \)'s will bring about in future works.

### 4 Experimental Setup

The experiments are conducted on five benchmarking datasets for binary text classification: Customer Review dataset (CR) (Hu and Liu, 2014), Opinion polarity dataset (MPQA) (Wiebe et al., 2005), Sentence Subjectivity dataset (SUBJ) (Pang and Lee, 2005), Movie Review dataset (MR) (Pang and Lee, 2005), and Stanford Sentiment Treebank (SST) dataset\(^2\). The statistics for the datasets are shown in Table 1.

<table>
<thead>
<tr>
<th>Dataset</th>
<th>#Count</th>
<th>Task</th>
<th>Classes</th>
</tr>
</thead>
<tbody>
<tr>
<td>CR</td>
<td>4k</td>
<td>product reviews</td>
<td>pos/neg</td>
</tr>
<tr>
<td>MPQA</td>
<td>11k</td>
<td>opinion polarity</td>
<td>pos/neg</td>
</tr>
<tr>
<td>SUBJ</td>
<td>10k</td>
<td>subjectivity</td>
<td>subj/obj</td>
</tr>
<tr>
<td>MR</td>
<td>11k</td>
<td>movie reviews</td>
<td>pos/neg</td>
</tr>
<tr>
<td>SST</td>
<td>70k</td>
<td>movie reviews</td>
<td>pos/neg</td>
</tr>
</tbody>
</table>

In this paper, we compare the classification accuracy of our proposed Complex Embedding Superposition (CE-Sup) network and Complex Embedding Mixture (CE-Mix) network with three existing unsupervised representation training models, Unigram-TFIDF and fastText Bag-of-Words (BOW), as well as two existing supervised representation training models, namely CaptionRep BOW (Hill et al., 2016b) and DictRep BOW (Hill et al., 2016c). We directly take the performances of these systems on the 5 datasets from existing works. Since the performances for CaptionRep and DictRep are not available on SST, we use the performance of another model called Paragraph-Phrase (Bansal and Livescu, 2016). For a fair comparison, we also implement an end-to-end supervised real embedding network (Real-Embed),

\(^2\)https://nlp.stanford.edu/sentiment/index.html
where each word is mapped to a real-valued vector in the embedding layer, based on which the sentence representation is obtained by averaging the embedding vectors for all words in the sentence, and a fully connected layer maps the sentence vector to the classification label. CE-mixture, CE-Superposition and Real-Embed are trained and tested in a completely identical process.

For the construction of training, validation and test data, they are readily available for SST dataset, and for the other four datasets we randomly split the whole data into 8:1:1 for training, validation and test data respectively. The embedding dimension is set to be 100. We use batch training with batch size being 32 for SST and 16 for the other datasets. We adopt Adam as the optimizer and use the default parameters for Adam in Keras 3.

The experiments are implemented in Keras and Tensorflow 4 under Python 3.6.4. The experiment is run on a desktop with NVidia Quadro M4000 and 16GB RAM.

5 Results and Discussion

In this study, we seek to answer the following two research questions:

RQ1. Do the proposed quantum-inspired complex embedding models outperform state-of-the-art non-quantum approaches?

RQ2. Out of the two proposed model in this study, which one performs better?

Table 2 presents the classification accuracy values of all models experimented in this paper, where the bold values indicate the best-performing models for each dataset. It can be clearly seen from the table that CE-Mix is the best-performing model, because it occupies the highest accuracy value on 4 out of 5 benchmarking datasets, and on the remaining dataset it performs only slightly worse than the best-performed model.

In order to make the results more convincing, we also conduct two-tailed p-tests on the performances. The hypotheses are:

H0. There is no difference between two groups of performances on a particular dataset.

H1. There is a difference between two groups of performances on a particular dataset.

We use the threshold 0.05 to accept or reject the null hypothesis: when the obtained p-value < 0.05, the null hypothesis is rejected; when p-value >0.05, the null hypothesis is accepted.

Regarding RQ1, it can be observed that CE-Sup and CE-Mix achieves consistently higher or comparable accuracy than non-quantum models under experiment. It illustrates the superiority of complex embedding network over traditional language model (Unigram-TFIDF) (p-value < 0.05 on all datasets, rejecting the null hypothesis, and so forth), unsupervised embeddings trained from external corpus (word2vec, fastText) (p-value < 0.05 on all datasets except MPQA), as well as supervised embedding methods (CaptionRep, DictRep and Paragram-Phrase) (p-value< on all datasets except MPQA). The fair comparison with real embedding network (p-value < 0.05 on all datasets) confirms the superiority of complex embedding over real embedding techniques.

Regarding RQ2, out of the two complex embedding models proposed in this study, CE-Mix performs consistently but insignificantly (p-value > 0.05) better than CE-Sup in all datasets. Even though it is yet a fully convincing evidence, this result provides us with some intuition that it seems better to model a sentence as a classical mixture of word projectors rather than as a superposition state of latent concepts. For future work we will evaluate the performances of these two models on other datasets as well as other tasks to reach a more solid conclusion.

6 Conclusion and Future Work

This paper attempts to address the challenge of representing the combinatorial meaning of words for word embedding. The successful applications of quantum-based models in IR tasks inspires us to construct Hilbert Space representation of words and sentences, and explore to build two quantum models for solving sentence classification task. The experimental result on five benchmarking datasets demonstrates their effectiveness.

This work contributes to the fields of both word embeddings and quantum-inspired IR. On the one hand, our work can be interpreted as an improved embedding approach, which tackles the challenge...
Table 2: Experimental Results in percentage(%) The best performed value for each dataset is in bold.

<table>
<thead>
<tr>
<th>Model</th>
<th>CR</th>
<th>MPQA</th>
<th>MR</th>
<th>SST</th>
<th>SUBJ</th>
</tr>
</thead>
<tbody>
<tr>
<td>Unigram-TFIDF</td>
<td>79.2</td>
<td>82.4</td>
<td>73.7</td>
<td>-</td>
<td>90.3</td>
</tr>
<tr>
<td>word2vec BOW</td>
<td>79.8</td>
<td><strong>88.3</strong></td>
<td>77.7</td>
<td>79.7</td>
<td>90.9</td>
</tr>
<tr>
<td>fastText BOW</td>
<td>78.9</td>
<td>87.4</td>
<td>76.5</td>
<td>78.8</td>
<td>91.6</td>
</tr>
<tr>
<td>CaptionRep BOW</td>
<td>69.3</td>
<td>70.8</td>
<td>61.9</td>
<td>-</td>
<td>77.4</td>
</tr>
<tr>
<td>DictRep BOW</td>
<td>78.7</td>
<td>87.2</td>
<td>76.7</td>
<td>-</td>
<td>90.7</td>
</tr>
<tr>
<td>Paragram-Phrase</td>
<td>-</td>
<td>-</td>
<td>-</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>Real-Embed</td>
<td>77.5</td>
<td>84.7</td>
<td>77.0</td>
<td>80.0</td>
<td>92.0</td>
</tr>
<tr>
<td>CE-Sup</td>
<td>80.0</td>
<td>85.7</td>
<td>78.4</td>
<td>82.6</td>
<td>92.6</td>
</tr>
<tr>
<td>CE-Mix</td>
<td><strong>81.1</strong></td>
<td>86.6</td>
<td><strong>79.8</strong></td>
<td><strong>83.3</strong></td>
<td>92.8</td>
</tr>
</tbody>
</table>

of capturing the emergent meaning of a combination of words. On the other hand, this can be viewed as a pioneering study on quantum-inspired language models with complex numbers, and also an trial effort to adopt the theoretical QWeb framework onto an application context.

For future work, it is necessary to conduct a more comprehensive evaluation of the proposed models, either by evaluating on more datasets or by evaluating the qualities of the trained complex embeddings. We are also looking forward to seek additional ways to model a sentence based on the word states, and the application of the models onto other NLP tasks.

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Natural Language Inference with Definition Embedding Considering Context On the Fly

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Abstract

Natural language inference (NLI) is one of the most important tasks in NLP. In this study, we propose a novel method using word dictionaries, which are pairs of a word and its definition, as external knowledge. Our neural definition embedding mechanism encodes input sentences with the definitions of each word of the sentences on the fly. It can encode definitions of words considering the context of the input sentences by using an attention mechanism. We evaluated our method using WordNet as a dictionary and confirmed that it performed better than baseline models when using the full or a subset of 100d GloVe as word embeddings.

1 Introduction

Recognition of the entailment relationship between two sentences is one of the most important tasks in the field of natural language processing. An understanding of entailment relationships among sentences is useful for performing tasks such as question answering, information retrieval, and summarization.

The task of recognizing the entailment relationship between two sentences is called recognizing textual entailment (RTE) or natural language inference (NLI). NLI has recently been getting more attention from researchers, owing to the release of large-scale corpora such as SNLI (Bowman et al., 2015) and MNLI (Williams et al., 2018).

These corpora consist of pairs of sentences, such as ‘A soccer game with multiple males playing.’ and ‘Some men are playing a sport.’, and ground-truth labels. Each label is a judgment of whether the latter sentence, which is the premise, is inferred from the former one, which is the hypothesis. In this example, the label is ‘entailment’.

In this study, we propose a novel method that uses word dictionaries as external knowledge. Word dictionaries are useful for domain adaptation, where we need to understand rare or novel words in which we do not have good embedding representations. For NLI, there is related work that does use dictionaries (Bahdanau et al., 2017). In it, a definition embedding method is proposed that obtains representations of out-of-vocabulary (OOV) words from dictionaries on the fly. In this method, however, the description of a word is converted into the same embedding anytime without considering the context of the input sentences.

On the other hand, we consider that word representation from dictionaries should reflect the context of the input sentences. In the dictionary, we can explain the meaning of a word from many aspects. However, the required information varies depending on the context of the input sentences. This problem also occurs for pre-trained word embeddings, which are usually fixed for all contexts in the previous studies.

The proposed method can obtain different representations of words according to the contexts of the input sentences. It introduces an attention mechanism that improves the encoded representations of each word in input sentences, by using the encoded word definitions of each word in the input sentences. Moreover, unlike Bahdanau’s method, it obtains the representation of all words from dictionaries on the fly in order to improve the representations of non-OOV words.

2 Task Definition

We follow the task definition of SNLI (Bowman et al., 2015) and MNLI (Williams et al., 2018). We define a dictionary as follows.
**Def. 1** (Dictionary). A dictionary \( D \) has the following components.

**Headword** \( y \) is an arbitrary token. **Definition** \( D^y \) is represented as a token sequence that defines the headword \( y \). This study assumes that each headword has only one definition. For a polysemic headword with multiple definitions, we use the concatenation of the definitions. **Vocabulary** \( V_D \) is the set of all headwords in the dictionary.

3 Related Work

Bahdanau et al. (2017) proposed a method that enables dictionary information to be used in NLP tasks, such as NLI, reading comprehension, and language modeling. Their method can obtain the embeddings of OOV words efficiently, because they obtain the definition embeddings for only OOV words instead of the random embeddings of the words. Our method is similar to theirs, but our purpose is different; we refine word embeddings considering the contexts of input sentences for all words.

The definition embedding is also useful for other tasks. Hill et al. (2016) used the definition embedding to understand the phrases. They presented two applications: reverse dictionaries and crossword question answering. They tackled these applications with phrase embeddings obtained from their definitions. Long et al. (2016) used the encoding of the word definition for the initialization of TransE (Bordes et al., 2013), which obtains the embedding of the relationship between two entities.

There is related work that uses other external resources for refining word representations. For NLI, Chen et al. (2017a) proposed a model that uses a knowledge graph to reflect word relationships (e.g., synonymy, hypernymy). Their method achieved state-of-the-art performance on SNLI; however, it cannot handle the definition description of each word.

Moreover, there are general frameworks to refine word embeddings by using external knowledge. Weissenborn et al. (2017) proposed a method that refines the word embedding by encoding the text transformation of ConceptNet (Speer and Havasi, 2012). McCann et al. (2017) proposed context vectors (CoVe), which uses an RNN encoder trained on machine translation datasets to introduce context information to the word embedding. Peters et al. (2018) proposed the Embeddings from Language Models (ELMo), which obtains contextualized word representations. They used the states of the middle layers in the deep language model. These methods are also effective at the NLI task.

4 Existing Methods

This section outlines the existing NLI models and describes the conventional model that uses a dictionary as external knowledge.

4.1 NLI model

In the architecture of a general NLI model (Bowman et al., 2015; Rocktäschel et al., 2016; Chen et al., 2017b), the input of the model is a pair of token sequences \( \{ X^s = \{ x_1^s, \ldots, x_{ls}^s \} : s \in \{ P, H \} \} \), where \( ls \) is the length of \( X^s \). \( s \in \{ P, H \} \) means a premise or hypothesis.

We call the following two layers together the Encoder.

**Encoder Word Embedding Layer (WEL)**

This layer takes \( X^s \) as input. Let \( e(y) \in \mathbb{R}^{ne} \) be the embedding of token \( y \). It outputs a vector sequence \( E^s = (e(x_1^s), \ldots, e(x_{ls}^s)) \in \mathbb{R}^{ne \times ls} \).

**Encoder Context Embedding Layer (CEL)**

This layer converts the vector sequence \( E^s \) into a contextualized vector sequence, \( C^s = f(E^s) \in \mathbb{R}^{ne \times ls} \). The most common approach is to use an RNN as \( f \).

The encoder outputs \( C^P \) and \( C^H \) for the premise and hypothesis sentences, respectively.

**Decoder**

The input of the decoder is a pair of vector sequences \( \{ C^P, C^H \} \). The decoder outputs the score vector of the classification labels.

4.2 Definition Embedding Mechanism

We summarize the definition embedding mechanism (DEM) (Bahdanau et al., 2017) as it relates to NLI. They proposed dictionary embedding mechanisms with many variations, such as mean pooling or an RNN. We select one of their models with an RNN, because we also use an RNN for the definition embedding.

The DEM acts on each premise and hypothesis. Its input is a token sequence \( X^s \) and the encoder word embedding sequence \( E^s \). The output is \( E^s \), and \( E^s \) is passed to the encoder CEL instead of \( E^s \). \( E^s \) is obtained by adding \( E^s \) to the final state of the RNN encoding of the definition. The sizes of \( E^s \) and \( E^s \) are each \( ne \times ls \).
5 Proposed Method

We propose a novel DEM considering the contexts of the input sentences. Our contributions are threefold. First, we introduce an attention mechanism. Second, we implement the mechanism after the encoder. Third, we consider definition embeddings of words including non-OOV ones.

The input is a token sequence \( X^s \) together with the encoder word and context embedding sequence \( E^s \) and \( C^s \), and the output is \( C^{ts} \). \( C^{ts} \) is passed to the decoder instead of \( C^s \), where the sizes of \( C^s \) and \( C^{ts} \) are each \( n_c \times l_s \). The proposed mechanism has the following layers.

**Definition Extracting Layer** Let \( V^s \) be the set of target tokens of the definition embedding which are in both the token sequence \( X^s \) and the vocabulary of the dictionary \( V_D \). The definition \( D^y \) of token \( y \in V^s \) is obtained from the dictionary \( D \). Let \( m_y \) be the length of \( D^y \). This layer outputs a set of target tokens \( V^s \) and a set of definitions \( \{D^y : y \in V^s\} \).

**Definition WEL** This layer has the same parameters as the encoder WEL. For each element of \( D^y \), it outputs a vector sequence \( E^y = \{e(d^y), \cdots, e(d^y_{m_y})\} \in \mathbb{R}^{n_c \times m_y} \).

**Definition CEL** This layer has the same model as the encoder CEL. Parameters are not shared with the encoder CEL. It converts the vector sequence \( E^y \) into the output of this layer \( C^y = f(E^y) \in \mathbb{R}^{n_c \times m_y} \).

**Definition Attention Layer** This layer obtains a fixed-length vector representation of definition \( D^y \) with an attention mechanism. It takes the outputs of the previous layers \( E^y, C^y, C^s \), and \( C^{ts} \) as input, where \( s \in \{P, H\} \) indicates that either the premise or hypothesis is different from \( s \).

For \( C^y \in \mathbb{R}^{n_c \times m_y}, C^s \in \mathbb{R}^{n_c \times l_s} \), we define an attention matrix \( A^{y,s} = \frac{1}{\sqrt{n_c}} C^s \top C^y \), and an attention vector \( a^{y,s} = \left( \frac{1}{l_s} \sum_j A_{ij}^{y,s} \right)_{j=1:\cdots,m_y} \in \mathbb{R}^{m_y} \). The attention vector \( a^{y,s} \) represents the extent that each token in definition \( D^y \) is related with the input sentence \( X^s \). The attended definition vector to the input sentence \( X^s \) is

\[
h^{y,s} = \sum_i \text{softmax}_i(a^{y,s})c^y_i \in \mathbb{R}^{nc},
\]

where \( c^y_i \) is the \( i \)-th state of the definition context embedding \( C^y \).

The last state of the definition context embedding is \( c_{m_y}^y \in \mathbb{R}^{nc} \). The output of this layer is a linear combination of the enhancements (Chen et al., 2017b) of the attended definition vectors,

\[
z^y = \left[c_{m_y}^y, h^{s,y,a} - c_{m_y}^y, h^{s,y,a} \odot c_{m_y}^y, h^{s,y,a} \{s}, h^{s,y,a} \odot c_{m_y}^y\right]w,
\]

where \( w \in \mathbb{R}^7 \) is a trainable parameter and \( \odot \) is the element-wise product. \( n_c \) is the size of \( z^y \).

**Output Layer** The output of the proposed mechanism is expressed as

\[
c^s_i = \left\{ \begin{array}{ll} c^s_i + z^y_i, & (x^s_i \in V^s) \\ c^s_i, & \text{otherwise} \end{array} \right. \]

The decoder receives \( C^{ts} \) instead of \( C^s \).

Algorithm 1 is the pseudo code of the definition embedding mechanism.

The above explanation only covers the case of NLI. However, the proposed method can be applied to any number of input sentences, because Equation (1) can take an arbitrary number of arguments. Therefore, it is applicable to other tasks that have text inputs, such as question answering and machine translation.

**Algorithm 1 Definition Embedding**

| Input: | \( X^s, E^s, C^s, C^s \) |
| Output: | \( C^{ts} \) |
| 1: | \( V^s, \{D^y : y \in V^s\} \leftarrow \text{Def. Ext.}(X^s) \) |
| 2: | for all \( y \) in \( V^s \) do |
| 3: | \( E^y \leftarrow \text{Def. Word Emb.}(D^y) \) |
| 4: | \( C^y \leftarrow \text{Def. Context Emb.}(E^y) \) |
| 5: | \( z^y \leftarrow \text{Def. Att.}(E^y, C^y, C^s, C^s) \) |
| 6: | end for |
| 7: | \( C^{ts} \leftarrow \text{Out}(E^s, C^s, V^s, \{z^y : y \in V^s\}) \) |

6 Experiments

This section describes the results of the evaluation of the proposed method.

6.1 Experimental Setup

We chose ESIM (Chen et al., 2017b) and one of the methods in Bahdanau et al. (2017) (BDN) as the baseline models. ESIM is based on the model in Section 4.1. BDN and our method each add a DEM to ESIM. In BDN, the target tokens of the definition embedding are not contained in the pre-trained word embedding vocabulary, because
BDN intends to supplement the embeddings of OOV words. However, in our method, the target tokens do not depend on a pre-trained word embedding vocabulary, because we intend to improve the representation of all the words by considering the context.

Our experiments were on the SNLI and MNLI benchmarks. For MNLI, we used a matched domain development dataset as our development data and a mismatched domain development dataset as our test data. The tokenizer was spaCy (Honnibal and Montani, 2018). The word embeddings were pre-trained 100d GloVe 6B vectors and 300d GloVe 840B vectors (Jeffrey Pennington and Manning, 2014). The embeddings were fixed during training, because we were interested in the difference in representation between pre-trained embeddings with and without dictionary information.

We used the vocabulary and definitions in WordNet (Miller, 1995) as dictionaries. For polysemic words with multiple definitions, we used the top-5 definitions connected in descending order of frequency of synsets, which are provided by WordNet. The number of headwords that appear in SNLI is 24103, and 45225 in MNLI.

The other settings are described in Appendix A.

6.2 Results

Does the proposed method refine the OOV word embedding? In order to investigate the effectiveness of our method against OOV words, we restricted the vocabulary of the 100d GloVe embedding to the most common 3000 words in each dataset and considered the other words as OOV (many-OOV setting). The word embeddings of the OOV words were randomly initialized according to a Gaussian distribution and fixed during training.

Table 1 shows the results. When there were many OOV words, our method improved test accuracy by 1.4% in SNLI and 1.5% in MNLI. In contrast, BDN did not improve accuracy in MNLI.

**Does the larger dictionary bring higher accuracy?** We also evaluated our method with the whole 100d GloVe embedding (not-many-OOV setting). In this experiment, we used the whole vocabulary of WordNet or restricted the WordNet vocabulary to the 1000 and 10000 most common words in the each dataset.

Figures 1a and 1b show the results when using 100d GloVe. We confirmed that the larger dictionary raises accuracy. We think that the pre-trained GloVe embeddings for the frequent words were more appropriate than those for the rare words. This means that our method was effective for words that had relatively poor embeddings and occur sufficiently often in the training data.

We confirmed that the threefold originality of our method contributed to the improvement in the whole WordNet setting. The proposed method using the whole WordNet achieved the higher test accuracy on each dataset. The improvement from ESIM was 1.0% in SNLI and 0.8% in MNLI. Moreover, our method without the definition attention mechanism performed worse by 0.4% in SNLI and 0.5% in MNLI in comparison with the method with it. This implies that our definition embedding layer plays an important role in the definition embedding. In particular, the implementation of the attention mechanism after the encoder, which is essential to reflecting the context of input sentences, contributes to a refined representation.

BDN did not perform well. The number of
OOV words in SNLI (MNLI) is 415 (913); therefore, BDN could not sufficiently train the representations of the words with the sentences in the datasets.

**Does the improvement depend on the quality of the word embedding?** Figures 1c and 1d show the results when using 300d GloVe. In this setting, our method provided no significant improvement. It performed slightly better (worse) than ESIM in SNLI (MNLI). BDN, as well, did not perform better than ESIM. We think the 300d GloVe has sufficiently correct embeddings for most of the words in SNLI and MNLI, because it was created from a much larger corpora (340 billion tokens) than that of the 100d one (eight billion tokens).

To summarize the experimental results for the first and third research questions, the effectiveness of our method is dependent on the quality and coverage of word embeddings. That is, our method is effective for rare or novel words.

**7 Conclusion**

We proposed a novel definition embedding method. The method considers the contexts of the input sentences with an attention mechanism for the definition embeddings. It considers the definition embeddings of words including non-OOV words. Experimental results showed that it is effective for rare or novel words that do not have good pre-trained word embeddings.

**References**


Adina Williams, Nikita Nangia, and Samuel R. Bowman. 2018. A broad-coverage challenge corpus for sentence understanding through inference. In NAACL.


A Details of the Implementation

The section describes our implementation so that our experiments can be reproduced.

We implemented our method in PyTorch (Paszke et al., 2017) and trained it on one Nvidia GeForce GTX 1080 GPU. The RNNs in the encoder, decoder, and definition embedding mechanism were two-layer bi-directional simple recurrent units (SRUs) (Lei and Zhang, 2017). The size of the output of the RNN was $n_e = 2n_e$. The activation function in the RNN was the tanh function. Dropout with a keep ratio of 0.8 was applied to the same layer as ESIM and the definition embedding layer.

The parameters of the weights were initialized using the Xavier normal initializer (Glorot and Bengio, 2010), and the parameters of the biases were initialized as zero vectors. Word embeddings not contained in pre-trained GloVe were randomized according to a Gaussian distribution.

The mini-batch size was set to 16. The optimizer was Adadelta (Zeiler, 2012) with an initial learning rate of 0.075 and $\rho$ of 0.9. Early stopping with a patience of 7 was used to avoid overfitting.

We removed words whose definition length was one and stop words in the Natural Language Toolkit (Bird et al., 2009) from the vocabulary of the dictionary.
Comparison of Representations of Named Entities for Multi-label Document Classification with Convolutional Neural Networks

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Abstract

We explore representations for multi-token names in the context of the Reuters topic and sector classification tasks (RCV1). We find that: the best way to treat names is to split them into tokens and use each token as a separate feature; NEs have more impact on sector classification than on topic classification; replacing all NEs with special entity-type tokens is not an effective strategy; representing tokens by different embeddings for proper names vs. common nouns does not improve results. We highlight the improvements over state-of-the-art results that our CNN models yield.

1 Introduction

This paper addresses large-scale multi-class text classification tasks: categorizing articles in the Reuters news corpus (RCV1) according to topic and to industry sectors. A topic is a broad news category, e.g., “Economics,” “Sport,” “Health.” A sector defines a narrower business area, e.g., “Banking,” “Telecommunications,” “Insurance.”

We use convolutional neural networks (CNNs), which take word embeddings as input. Typically word embeddings are built by treating a corpus as a sequence of tokens, where named entities (NEs) receive no special treatment. Yet NEs may be important features in some classification tasks: companies, e.g., are often linked to particular industry sectors, and certain industries are linked to locations. Thus company and location names may be important features for sector classification.

RCV1 is much smaller than corpora typically used to build word embeddings. Thus we utilize external resources—a corpus of approximately 10 million business news articles, collected using the PULS news monitoring system (Pivovarova et al., 2013). While nominally RCV1 contains general news, it is skewed toward business; many of the topic labels are business-related (“Markets”, “Commodities”, “Share Capital,” etc.). Thus, we expect our business corpus to help in learning features for the Reuters classification tasks.

We compare several NE representation to find the most suitable name features for each task. We use the PULS NER system (Grishman et al., 2003; Huttunen et al., 2002a,b) to find NEs and their types—company, location, person, etc. We compare various representations of NEs, by building embeddings, and training CNNs to find the best representation. We also compare building embeddings on the RCV1 corpus vs. using much larger external corpora.

2 Data and Prior Work

RCV1 (Lewis et al., 2004) is a corpus of about 800K Reuters articles from 1996–1997 with manually assigned sector and topic labels. Both classifications are multi-label—each document may have zero or more labels. While all documents have topic labels, only 350K have sector labels.

While RCV1 appears frequently in published research, few authors tackle the full-scale classification problem. Typically they use subsets of the data: (Daniely et al., 2017; Duchi et al., 2011) use only the four most general topic labels; (Dredze et al., 2008) use 6 sector categories to explore binary classification, (Daniels and Metaxas, 2017) use a subset of 6K articles. Even when the entire dataset is used, the training-text split varies across papers, because the “original” split (Lewis et al., 2004) is impractical for most purposes: 23K instances for training, and 780K for testing.

Another problem that complicates comparison is the lack of consistency in evaluation metrics.
A US appeals Court revived a civil suit accusing Apple of creating a monopoly.

used to evaluate classifier performance. The most common measures for multi-class classification are macro- and micro-averaged F-measure, which we use in this paper. However, others use other metrics. For example, (Liu et al., 2017) use precision and cumulative gain at top K—measures adopted from information retrieval. This is not comparable with other work, because these metrics are used not only to report results, but also to optimize the algorithms during training. The notion of the best classifier differs depending on which evaluation measure is used. Thus, although RCV1 is frequently used, we find few papers directly comparable to our research, in the sense that they use the entire RCV1 dataset and report micro- and macro-averaged F-measure.

To the best of our knowledge, our previous work (Du et al., 2015) was the only study of the utility of NEs for RCV1 classification. We demonstrated that using a combination of keyword-based and NE-based classifiers works better than either classifier alone. In that paper we applied a rule-based approach for NEs, and did not use NEs as features for machine learning.

3 Model

The architecture of our CNN is shown in Figure 1. The inputs are fed into the network as zero-padded text fragments of fixed size, with each word represented by a fixed-dimensional embedding vector. The inputs are fed into a layer of convolutional filters with multiple widths, optionally followed by deeper convolutional layers. The results of the last convolutional layer are max-pooled, producing a vector with one scalar per filter. This is fed into a fully-connected layer with dropout regularization, with one sigmoid node in the output layer for each of the class labels. For each class label, a cross-entropy loss is computed. Losses are averaged across labels, and the gradient of the loss is back-propagated to update the weights. This is similar to the model (Kim, 2014) used for sentiment analysis. The key differences are that our model uses an arbitrary number of convolutions, and that we use sigmoid rather than softmax on output, since the labels are not mutually exclusive.

To train the model we used a random split: 80% of the data used for training, 10% development set, and 10% test set. The development set is used to determine when to stop training, and to tune a set of optimal thresholds \( \{ \theta_i \} \) for each label \( i \)—if the output probability \( p_i \) is higher than \( \theta_i \), the label is assigned to the instance, otherwise it is not. To find the optimal threshold, we optimize the F-measure for each label. The test set is used to obtain the final, reported performance scores.

Our focus is this paper is data representation, thus we defer the tuning of hyper-parameters for future work. All experiments use the same network structure: 3 convolution layers with filter sizes \{3,7,11\}, \{3,7,11\}, and \{3,11\}, with 512, 256 and 256 filters of each size, respectively. The runs differ only in the input embeddings they use.

4 Data Representation

We train the embeddings using GloVe (Pennington et al., 2014). As features we use lower-cased lemmas of all words. The rationale for this is that our
corpora are relatively small, so the data are sparse and not sufficient to build embeddings from surface forms. We tune the embeddings while training the CNN, updating them at each iteration.

We explore several name representations, using our NER system:

- **type**: each entity is represented by a special token denoting its type—C-company, C-person, C-location, etc, and C-name if the type is not determined. The model learns one embedding for each of these tokens.

- **name**: each name gets its own embedding; multi-word names treated as a single token.

- **split-name**: multi-word names are split into tokens, and each token has its own embedding; the motivation is that some company names may contain informative parts—e.g., Air Baltic, Delta Airlines—which may indicate that these companies operate in the same field; these name parts may be more useful than the name as a whole.

- **split-name+common**: similar to the above, but tokens inside names and in common context are distinguished; the motivation is that some words may be used in names without any relation to the company’s line of business—e.g., Apple, BlackBerry—and their usage inside names should not be mixed with their usage as common nouns.

In the experiments, we build GloVe embeddings from two corpora: RCV1 only, and RCV1 plus our external corpus. For comparison, we also use 200-dimensional embeddings trained on a 6 billion general corpus (glove-6B), provided by the GloVe project. This corresponds to our split-name representation mode.

To illustrate the effect of the different token representations, Table 1 shows ten words nearest to the sample lemmas: apple and airline. When name representation is used, the token apple is ambiguous, its nearest neighbors are both fruit words (pear) and computer words (apple_computer). In type representation, the “computer” meaning disappears, since all mentions of Apple as company are represented by the special token C-company. When using glove-6B, the fruit meaning is absent, and all neighbors are computer-related words. The token airline does not exhibit such ambiguity, and all representations produce similar nearest neighbors.

Table 1: Nearest neighbors for sample words using various word representations.

<table>
<thead>
<tr>
<th>name</th>
<th>type</th>
<th>glove-6B</th>
</tr>
</thead>
<tbody>
<tr>
<td>apple</td>
<td></td>
<td></td>
</tr>
<tr>
<td>pear</td>
<td>pear</td>
<td>phone</td>
</tr>
<tr>
<td>unpasteurized</td>
<td>unpasteurized</td>
<td>microsoft</td>
</tr>
<tr>
<td>juice</td>
<td>juice</td>
<td>intel</td>
</tr>
<tr>
<td>apple_computer</td>
<td>fruit</td>
<td>macintosh</td>
</tr>
<tr>
<td>odwalla</td>
<td>salmonella</td>
<td>ipod</td>
</tr>
<tr>
<td>strawberry</td>
<td>peach</td>
<td>ibm</td>
</tr>
<tr>
<td>macintosh</td>
<td>orange</td>
<td>ipad</td>
</tr>
<tr>
<td>meat</td>
<td>crate</td>
<td>software</td>
</tr>
<tr>
<td>pear_board</td>
<td>strawberry</td>
<td>google</td>
</tr>
<tr>
<td>airline</td>
<td></td>
<td>times</td>
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<table>
<thead>
<tr>
<th>airline</th>
<th></th>
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<tr>
<td>carrier</td>
<td>flight</td>
<td>airlines</td>
</tr>
<tr>
<td>british_airways</td>
<td>passenger</td>
<td>lufthansa</td>
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<td>carrier</td>
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<table>
<thead>
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<th>split-name+common</th>
<th></th>
<th></th>
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</thead>
<tbody>
<tr>
<td>apple</td>
<td>apple_NW</td>
<td>airline_NW</td>
</tr>
<tr>
<td>pear</td>
<td>computer_NE</td>
<td>airlines_NE</td>
</tr>
<tr>
<td>unpasteurized</td>
<td>macintosh_NE</td>
<td>airways_NE</td>
</tr>
<tr>
<td>odwalla_NE</td>
<td>amelius_NE</td>
<td>carrier</td>
</tr>
<tr>
<td>fruit</td>
<td>operating-system</td>
<td>airline</td>
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<tr>
<td>anthrax</td>
<td>compaq_NE</td>
<td>air_NE</td>
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</tr>
<tr>
<td>rotten</td>
<td>oracle_NE</td>
<td>lufthansa_NE</td>
</tr>
<tr>
<td>unpasteurized</td>
<td>ibm_NE</td>
<td>pilot</td>
</tr>
<tr>
<td>strawberry</td>
<td>software</td>
<td>aircraft</td>
</tr>
<tr>
<td></td>
<td></td>
<td>route</td>
</tr>
</tbody>
</table>

In the split-name+common representation mode, each lemma may produce two vectors, one for a common noun and one for a proper noun (inside a name). As the table shows, apple as a common noun has a clear “fruit” meaning; the one company appearing among the neighbors is a juice producer, Odwalla. The nearest neighbors for apple_NE, in name context, include IT companies. The tokens airline and airline_NE have no clear semantic distinction, with similar nearest neighbors. In such cases there is no clear advantage in using two embeddings rather than one.

We test all of the above name representations experimentally, to determine which is more useful in the document classification tasks.

5 Results and Discussion

Experimental results are presented in Tables 2 and 3. We compare our results with those found in related work, described in Section 2, focusing on micro- and macro-averaged F-measure—µ-F1 and M-F1, respectively. The experimental settings differ in the various papers, which makes precise comparison difficult. For example, several previous papers use the “standard split,” (proposed in (Lewis et al., 2004)), which contains only 23K
Algorithm (prior) & M-F1 & µ-F1 \\
SVM (Lewis et al., 2004) & 29.7 & 51.3 \\
SVM (Zhuang et al., 2005) & 30.1 & 52.0 \\
Naive Bayes (Puhrula, 2012) & — & 70.5 \\
Bloom Filters (Cisse et al., 2013) & 47.8 & 72.4 \\
SVM + NEs (Du et al., 2015) & 57.7 & 63.8 \\

<table>
<thead>
<tr>
<th>Algorithm (prior)</th>
<th>M-F1</th>
<th>µ-F1</th>
</tr>
</thead>
<tbody>
<tr>
<td>SVM (Lewis et al., 2004)</td>
<td>61.9</td>
<td>81.6</td>
</tr>
<tr>
<td>ANN (Nam et al., 2014)</td>
<td>69.2</td>
<td>85.3</td>
</tr>
<tr>
<td>CNN (Johnson and Zhang, 2015)</td>
<td>67.1</td>
<td>85.7</td>
</tr>
</tbody>
</table>

Table 2: Sector classification results on RCV1.

Table 3: Topic classification results on RCV1.

taining training instances, which is not sufficient for learning word embeddings.

Compared to the reported state-of-the-art results on Sector Classification (Table 2), our best model yields a 10% gain in µ-F1, (Cisse et al., 2013), and a 6% gain in M-F1 (Du et al., 2015). The best µ-F1 and M-F1 results are obtained by the same model.\(^1\)

On Topic Classification (Table 3), our µ-F1 results show a modest improvement of 0.5% in F-measure—or a 3.5% (averaged) error reduction—over state of the art (Johnson and Zhang, 2015).\(^2\)

As seen in Table 2, the best data representation for Sector Classification, is **split-name**, where each token has the same embedding regardless whether it is used in a proper-name or a common-noun context. The worst performing name representation is **type**, where names are mapped to special “concepts” (C-company, C-person etc.), and each concept has its own embedding. This indicates the importance of the tokens inside the named entities for Sector Classification, and supports the notion that company names mentioned in text correlate with sector labels.

Results for Topic Classification are in Table 3. The best data representation is again **split-name**, though the difference between representations is less pronounced than in the case of Sector Classification, and using **type** does not lead to a significant drop in model performance. This suggests that proper names are less important for Topic (event) classification, and supports the intuition that entity names (e.g., companies) are less correlated with the types of events in which the entities participate in business news. However, there may be correlations between industry sectors and topics/events: e.g., mining or petroleum companies rarely launch new products. This may explain why the **split-name** representation appears to be better for Topic Classification. One possible next step is to build CNNs that jointly model Topics and Sectors; we plan to explore this in future work.

Surprisingly, using external corpora did not improve the models’ performance, as indicated by both Sector and Topic results (Tables 2 and 3, respectively). This may mean that the genre and the time period of the news corpus are more relevant for building embeddings than the size of the corpora. However, other factors may contribute as well, e.g., our hyper-parameter combination may not be optimal for these embeddings. Nevertheless, the results follow the same pattern: the best name representation is **split-name** and the difference between representations is more pronounced for Sector than for Topic classification.

In conclusion, our contribution is two-fold. On one classic large-scale classification task, sectors, our proposed CNNs yield substantial improvements over state-of-the-art; on topics—a modest improvement in µ-F-measure. Further, to the best of our knowledge, this is the first attempt at a systematic comparison of NE representation for text classification. More effective ways of representing NEs should be explored in future work, given their importance for the classification tasks, as demonstrated by the experiments we present in this paper.
References


Dong Zhuang, Benyu Zhang, Qiang Yang, Jun Yan, Zheng Chen, and Ying Chen. 2005. Efficient text classification by weighted proximal SVM. In Fifth IEEE International Conference on Data Mining.
Abstract

Context plays an important role in human language understanding, thus it may also be useful for machines learning vector representations of language. In this paper, we explore an asymmetric encoder-decoder structure for unsupervised context-based sentence representation learning. We carefully designed experiments to show that neither an autoregressive decoder nor an RNN decoder is required. After that, we designed a model which still keeps an RNN as the encoder, while using a non-autoregressive convolutional decoder. We further combine a suite of effective designs to significantly improve model efficiency while also achieving better performance. Our model is trained on two different large unlabelled corpora, and in both cases the transferability is evaluated on a set of downstream NLP tasks. We empirically show that our model is simple and fast while producing rich sentence representations that excel in downstream tasks.

1 Introduction

Learning distributed representations of sentences is an important and hard topic in both the deep learning and natural language processing communities, since it requires machines to encode a sentence with rich language content into a fixed-dimension vector filled with real numbers. Our goal is to build a distributed sentence encoder learnt in an unsupervised fashion by exploiting the structure and relationships in a large unlabelled corpus.

Numerous studies in human language processing have supported that rich semantics of a word or sentence can be inferred from its context (Harris, 1954; Firth, 1957). The idea of learning from the co-occurrence (Turney and Pantel, 2010) was recently successfully applied to vector representation learning for words in Mikolov et al. (2013) and Pennington et al. (2014).

A very recent successful application of the distributional hypothesis (Harris, 1954) at the sentence-level is the skip-thoughts model (Kiros et al., 2015). The skip-thoughts model learns to encode the current sentence and decode the surrounding two sentences instead of the input sentence itself, which achieves overall good performance on all tested downstream NLP tasks that cover various topics. The major issue is that the training takes too long since there are two RNN decoders to reconstruct the previous sentence and the next one independently. Intuitively, given the current sentence, inferring the previous sentence and inferring the next one should be different, which supports the usage of two independent decoders in the skip-thoughts model. However, Tang et al. (2017) proposed the skip-thoughts neighbour model, which only decodes the next sentence based on the current one, and has similar performance on downstream tasks compared to that of their implementation of the skip-thoughts model.

In the encoder-decoder models for learning sentence representations, only the encoder will be used to map sentences to vectors after training, which implies that the quality of the generated language is not our main concern. This leads to our two-step experiment to check the necessity of applying an autoregressive model as the decoder. In other words, since the decoder’s performance on language modelling is not our main concern, it is preferred to reduce the complexity of the decoder to speed up the training process. In our experiments, the first step is to check whether “teacher-forcing” is required during training if we stick to using an autoregressive model as the decoder, and the second step is to check whether an autoregres-
sive decoder is necessary to learn a good sentence encoder. Briefly, the experimental results show that an autoregressive decoder is indeed not essential in learning a good sentence encoder; thus the two findings of our experiments lead to our final model design.

Our proposed model has an asymmetric encoder-decoder structure, which keeps an RNN as the encoder and has a CNN as the decoder, and the model explores using only the subsequent context information as the supervision. The asymmetry in both model architecture and training pair reduces a large amount of the training time.

The contribution of our work is summarised as:

1. We design experiments to show that an autoregressive decoder or an RNN decoder is not necessary in the encoder-decoder type of models for learning sentence representations, and based on our results, we present two findings. **Finding I:** It is not necessary to input the correct words into an autoregressive decoder for learning sentence representations. **Finding II:** The model with an autoregressive decoder performs similarly to the model with a predict-all-words decoder.

2. The two findings above lead to our final model design, which keeps an RNN encoder while using a CNN decoder and learns to encode the current sentence and decode the subsequent contiguous words all at once.

3. With a suite of techniques, our model performs decently on the downstream tasks, and can be trained efficiently on a large unlabelled corpus.

The following sections will introduce the components in our “RNN-CNN” model, and discuss our experimental design.

## 2 RNN-CNN Model

Our model is highly asymmetric in terms of both the training pairs and the model structure. Specifically, our model has an RNN as the encoder, and a CNN as the decoder. During training, the encoder takes the $i$-th sentence $S_i$ as the input, and then produces a fixed-dimension vector $z_i$, as the sentence representation; the decoder is applied to reconstruct the paired target sequence $T_i$ that contains the subsequent contiguous words. The distance between the generated sequence and the target one is measured by the cross-entropy loss at each position in $T_i$. An illustration is in Figure 1. (For simplicity, we omit the subscript $i$ in this section.)

### 1. Encoder: The encoder is a bi-directional Gated Recurrent Unit (GRU, Chung et al. (2014))

Suppose that an input sentence $S$ contains $M$ words, which are $\{w_1^1, w_2^1, \ldots, w_M^1\}$, and they are transformed by an embedding matrix $E \in \mathbb{R}^{d_e \times |V|}$ to word vectors $\{e_1^1, e_2^1, \ldots, e_M^1\}$. The bi-directional GRU takes one word vector at a time, and processes the input sentence in both the forward and backward directions; both sets of hidden states are concatenated to form the hidden state matrix $H = [h^1_1, h^1_2, \ldots, h^1_M] \in \mathbb{R}^{d_h \times M}$, where $d_h$ is the dimension of the hidden states $h^m_n = [\tilde{h}^m_n; \tilde{h}^m_{-n}]$ ($\forall m \in \{1, 2, \ldots, M\}$).

### 2. Representation: We aim to provide a model with faster training speed and better transferability than existing algorithms; thus we choose to apply a parameter-free composition function, which is a concatenation of the outputs from a global mean pooling over time and a global max pooling over time, on the computed sequence of hidden states $H$. The composition function is represented as

$$z = \left[\frac{1}{M} \sum_{m=1}^{M} h^m; \text{MaxPool}(H)\right]$$

where MaxPool is the max operation on each row of the matrix $H$, which outputs a vector with dimension $d_h$. Thus the representation $z \in \mathbb{R}^{2d_h}$.

### 3. Decoder: The decoder is a 3-layer CNN to reconstruct the paired target sequence $T$, which needs to expand $z$, which can be considered as a sequence with only one element, to a sequence with $T$ elements. Intuitively, the decoder could be a stack of deconvolution layers. For fast training speed, we optimised the architecture to make it possible to use fully-connected layers and convolution layers in the decoder, since generally, convolution layers run faster than deconvolution layers in modern deep learning frameworks.

Suppose that the target sequence $T$ has $N$ words, which are $\{w_1^1, w_2^1, \ldots, w_N^1\}$, the first layer of de-convolution will expand $z \in \mathbb{R}^{2d_h \times 1}$, into a feature

---

1 We experimented with both Long-short Term Memory (LSTM, Hochreiter and Schmidhuber (1997)) and GRU. Since LSTM didn’t give us significant performance boost, and generally GRU runs faster than LSTM, in our experiments, we stick to using GRU in the encoder.

2 $V$ is the vocabulary, and $|V|$ is the number of unique words in the vocabulary. $d_e$ is the dimension of each word vector.
Figure 1: Our proposed model is composed of an RNN encoder, and a CNN decoder. During training, a batch of sentences are sent to the model, and the RNN encoder computes a vector representation for each of sentences; then the CNN decoder needs to reconstruct the paired target sequence, which contains 30 contiguous words right after the input sentence, given the vector representation. 300 is the dimension of word vectors. $2d_h$ is the dimension of the sentence representation which varies with the RNN encoder size. (Better view in colour.)

map with $N$ elements. It can be easily implemented as a concatenation of outputs from $N$ linear transformations in parallel. Then the second and third layer are 1D-convolution layers. The output feature map is $U = [u^1, u^2, ..., u^N] \in \mathbb{R}^{d_e \times N}$, where $d_e$ is the dimension of the word vectors.

Note that our decoder is not an autoregressive model and has high training efficiency. We will discuss the reason for choosing this decoder which we call a predict-all-words CNN decoder.

4. Objective: The training objective is to maximise the likelihood of the target sequence being generated from the decoder. Since in our model, each word is predicted independently, a softmax layer is applied after the decoder to produce a probability distribution over words in $V$ at each position, thus the probability of generating a word $w^m_n$ in the target sequence is defined as:

$$P(w^m_n) = \frac{e^{E(w^m_n)^\top u^n}}{\sum_{w \in V} e^{E(w)^\top u^n}},$$

where $E(w)$ is the vector representation of $w$ in the embedding matrix $E$, and $E(w)^\top u^n$ is the dot-product between the word vector and the feature vector produced by the decoder at position $n$. The training objective is to minimise the sum of the negative log-likelihood over all positions in the target sequence $T$:

$$\mathcal{L}(\phi, \theta) = -\log P(T|S; \phi, \theta)$$

$$= -\sum_{n=1}^{N} \log P(w^n_1|w^n_1, w^n_2, ..., w^n_M; \phi, \theta),$$

where $\phi$ and $\theta$ contain the parameters in the encoder and the decoder, respectively. The training objective $\mathcal{L}(\phi, \theta)$ is summed over all sentences in the training corpus.

3 Architecture Design

We use an encoder-decoder model and use context for learning sentence representations in an unsupervised fashion. Since the decoder won’t be used after training, and the quality of the generated sequences is not our main focus, it is important to study the design of the decoder. Generally, a fast training algorithm is preferred; thus proposing a new decoder with high training efficiency and also strong transferability is crucial for an encoder-decoder model.

3.1 CNN as the decoder

Our design of the decoder is basically a 3-layer ConvNet that predicts all words in the target sequence at once. In contrast, existing work, such as skip-thoughts (Kiros et al., 2015), and CNN-LSTM (Gan et al., 2017), use autoregressive RNNs as the decoders. As known, an autoregressive model is good at generating sequences with high quality, such as language and speech. However, an autoregressive decoder seems to be unnecessary in an encoder-decoder model for learning sentence representations, since it won’t be used after training, and it takes up a large portion of the training time to compute the output and the gradient. Therefore, we conducted experiments to test the necessity of using an autoregressive decoder in learning sentence representations, and we had two findings.
Table 1: The models here all have a bi-directional GRU as the encoder (dimensionality 300 in each direction). The default way of producing the representation is a concatenation of outputs from a global mean-pooling and a global max-pooling, while “-MaxOnly” refers to the model with only global max-pooling. Bold numbers are the best results among all presented models. We found that 1) inputting correct words to an autoregressive decoder is not necessary; 2) predict-all-words decoders work roughly the same as autoregressive decoders; 3) mean+max pooling provides stronger transferability than the max-pooling alone does. The table supports our choice of the predict-all-words CNN decoder and the way of producing vector representations from the bi-directional RNN encoder.

Finding I: It is not necessary to input the correct words into an autoregressive decoder for learning sentence representations.

The experimental design was inspired by Bengio et al. (2015). The model we designed for the experiment has a bi-directional GRU as the encoder, and an autoregressive decoder, including both RNN and CNN. We started by analysing the effect of different sampling strategies of the input words on learning an autoregressive decoder.

We compared three sampling strategies of input words in decoding the target sequence with an autoregressive decoder: (1) **Teacher-Forcing**: the decoder always gets the ground-truth words; (2) **Always Sampling**: at time step $t$, a word is sampled from the multinomial distribution predicted at time step $t-1$; (3) **Uniform Sampling**: a word is uniformly sampled from the dictionary $V$, then fed to the decoder at every time step.

The results are presented in Table 1 (top two subparts). As we can see, the three decoding settings do not differ significantly in terms of the performance on selected downstream tasks, with RNN or CNN as the decoder. The results show that, in terms of learning good sentence representations, the autoregressive decoder doesn’t require the correct ground-truth words as the inputs.

Finding II: The model with an autoregressive decoder performs similarly to the model with a predict-all-words decoder.

With Finding I, we conducted an experiment to test whether the model needs an autoregressive decoder at all. In this experiment, the goal is to compare the performance of the predict-all-words decoders and that of the autoregressive decoders separate from the RNN/CNN distinction, thus we designed a predict-all-words CNN decoder and RNN decoder. The predict-all-words CNN decoder is described in Section 2, which is a stack of three convolutional layers, and all words are predicted once at the output layer of the decoder. The predict-all-words RNN decoder is built based on our CNN decoder. To keep the number of parameters of the two predict-all-words decoder roughly the same, we replaced the last two convolutional layers with a bidirectional GRU.

The results are also presented in Table 1 (3rd and 4th subparts). The performance of the predict-all-words RNN decoder does not significantly differ from that of any one of the autoregressive RNN de-
coders, and the same situation can be also observed in CNN decoders.

These two findings indeed support our choice of using a predict-all-words CNN as the decoder, as it brings the model high training efficiency while maintaining strong transferability.

3.2 Mean-Max Pooling

Since the encoder is a bi-directional RNN in our model, we have multiple ways to select/compute on the generated hidden states to produce a sentence representation. Instead of using the last hidden state as the sentence representation as done in skip-thoughts (Kiros et al., 2015) and SDAE (Hill et al., 2016), we followed the idea proposed in Chen et al. (2016). They built a model for supervised training on the SNLI dataset (Bowman et al., 2015) that concatenates the outputs from a global mean pooling over time and a global max pooling over time to serve as the sentence representation, and showed a performance boost on the SNLI task. Conneau et al. (2017) found that the model with global max pooling function provides stronger transferability than the model with a global mean pooling function does.

In our proposed RNN-CNN model, we empirically show that the mean+max pooling provides stronger transferability than the max pooling alone does, and the results are presented in the last two sections of Table 1. The concatenation of a mean-pooling and a max pooling function is actually a parameter-free composition function, and the computation load is negligible compared to all the heavy matrix multiplications in the model. Also, the non-linearity of the max pooling function augments the mean pooling function for constructing a representation that captures a more complex composition of the syntactic information.

3.3 Tying Word Embeddings and Word Prediction Layer

We choose to share the parameters in the word embedding layer of the RNN encoder and the word prediction layer of the CNN decoder. Tying was shown in both Inan et al. (2016) and Press and Wolf (2017), and it generally helped to learn a better language model. In our model, tying also drastically reduces the number of parameters, which could potentially prevent overfitting.

Furthermore, we initialise the word embeddings with pretrained word vectors, such as word2vec (Mikolov et al., 2013) and GloVe (Pennington et al., 2014), since it has been shown that these pretrained word vectors can serve as a good initialisation for deep learning models, and more likely lead to better results than a random initialisation.

3.4 Study of the Hyperparameters in Our Model Design

We studied hyperparameters in our model design based on three out of 10 downstream tasks, which are SICK-R, SICK-E (Marelli et al., 2014), and STS14 (Agirre et al., 2014). The first model we created, which is reported in Section 2, is a decent design, and the following variations didn’t give us much performance change except improvements brought by increasing the dimensionality of the encoder. However, we think it is worth mentioning the effect of hyperparameters in our model design. We present the Table 1 in the supplementary material and we summarise it as follows:

1. Decoding the next sentence performed similarly to decoding the subsequent contiguous words.
2. Decoding the subsequent 30 words, which was adopted from the skip-thought training code⁹, gave reasonably good performance. More words for decoding didn’t give us a significant performance gain, and took longer to train.
3. Adding more layers into the decoder and enlarging the dimension of the convolutional layers indeed slightly improved the performance on the three downstream tasks, but as training efficiency is one of our main concerns, it wasn’t worth sacrificing training efficiency for the minor performance gain.
4. Increasing the dimensionality of the RNN encoder improved the model performance, and the additional training time required was less than needed for increasing the complexity in the CNN decoder. We report results from both smallest and largest models in Table 2.

4 Experiment Settings

The vocabulary for unsupervised training contains the 20k most frequent words in BookCorpus. In order to generalise the model trained with a relatively small, fixed vocabulary to the much larger set of all possible English words, we followed the vocabulary expansion method proposed in Kiros et al. (2015), which learns a linear mapping from the pretrained word vectors to the learnt RNN word...
<table>
<thead>
<tr>
<th>Model</th>
<th>Hrs</th>
<th>SICK-R Acc.</th>
<th>SICK-E Acc.</th>
<th>STS14 Acc.</th>
<th>MSRP Acc.</th>
<th>TREC MR Acc.</th>
<th>CR Acc.</th>
<th>SUBJ Acc.</th>
<th>MPQA Acc.</th>
<th>SST Acc.</th>
</tr>
</thead>
<tbody>
<tr>
<td>ParagraphVec</td>
<td>4</td>
<td>-</td>
<td>-</td>
<td>0.42/0.43</td>
<td>72.9/81.1</td>
<td>59.4</td>
<td>60.2</td>
<td>66.9</td>
<td>73.6</td>
<td>70.7</td>
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<tr>
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<td>0.8300</td>
<td>78.7</td>
<td>0.65/0.64</td>
<td>72.5/84.1</td>
<td>83.6</td>
<td>77.7</td>
<td>79.8</td>
<td>90.9</td>
<td>88.3</td>
</tr>
<tr>
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<td>-</td>
<td>0.8000</td>
<td>77.9</td>
<td>0.63/0.62</td>
<td>72.4/81.2</td>
<td>81.8</td>
<td>76.5</td>
<td>78.9</td>
<td>91.6</td>
<td>87.4</td>
</tr>
<tr>
<td>SIF (GloVe+WR)</td>
<td>-</td>
<td>0.8603</td>
<td>84.6</td>
<td>0.69/-</td>
<td>-/-</td>
<td>-/-</td>
<td>-/-</td>
<td>-/-</td>
<td>-/-</td>
<td>82.2</td>
</tr>
<tr>
<td>GloVe BOW</td>
<td>-</td>
<td>0.8000</td>
<td>78.6</td>
<td>0.54/0.56</td>
<td>72.1/80.9</td>
<td>83.6</td>
<td>78.7</td>
<td>78.5</td>
<td>91.6</td>
<td>87.6</td>
</tr>
<tr>
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<td>-</td>
<td>0.37/0.38</td>
<td>73.7/80.7</td>
<td>-</td>
<td>78.4</td>
<td>74.6</td>
<td>78.0</td>
<td>90.8</td>
<td>86.9</td>
</tr>
</tbody>
</table>

**Unsupervised training with unordered sentences**

<table>
<thead>
<tr>
<th>Model</th>
<th>Measurement</th>
<th>r Acc.</th>
<th>r/ρ Acc./F1</th>
<th>Accuracy</th>
</tr>
</thead>
<tbody>
<tr>
<td>FastSent</td>
<td></td>
<td>0.63/0.64</td>
<td>72.8/80.3</td>
<td>76.8</td>
</tr>
<tr>
<td>FastSent+AE</td>
<td></td>
<td>0.62/0.62</td>
<td>71.2/79.1</td>
<td>80.4</td>
</tr>
<tr>
<td>Skip-thoughts</td>
<td>336</td>
<td>0.8580</td>
<td>82.3</td>
<td>73.0/82.0</td>
</tr>
<tr>
<td>Skip-thought+LN</td>
<td>720</td>
<td>0.8580</td>
<td>79.5</td>
<td>0.44/0.45</td>
</tr>
<tr>
<td>combine CNN-LSTM</td>
<td>-</td>
<td>0.8618</td>
<td>-</td>
<td>-</td>
</tr>
<tr>
<td>small RNN-CNN†</td>
<td></td>
<td>0.8530</td>
<td>82.6</td>
<td>0.58/0.56</td>
</tr>
<tr>
<td>large RNN-CNN†</td>
<td></td>
<td>0.8698</td>
<td>85.2</td>
<td>0.59/0.57</td>
</tr>
</tbody>
</table>

**Unsupervised training with ordered sentences - BookCorpus**

<table>
<thead>
<tr>
<th>Model</th>
<th>Measurement</th>
<th>r Acc.</th>
<th>r/ρ Acc./F1</th>
<th>Accuracy</th>
</tr>
</thead>
<tbody>
<tr>
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<td>0.8476</td>
<td>82.7</td>
<td>0.53/0.53</td>
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<tr>
<td>large RNN-CNN†</td>
<td></td>
<td>0.8616</td>
<td>84.3</td>
<td>0.51/0.51</td>
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</table>

**Unsupervised training with ordered sentences - Amazon Book Review**

<table>
<thead>
<tr>
<th>Model</th>
<th>Measurement</th>
<th>r Acc.</th>
<th>r/ρ Acc./F1</th>
<th>Accuracy</th>
</tr>
</thead>
<tbody>
<tr>
<td>small RNN-CNN†</td>
<td></td>
<td>0.8850</td>
<td>84.6</td>
<td>0.68/0.67</td>
</tr>
<tr>
<td>large RNN-CNN†</td>
<td></td>
<td>0.8840</td>
<td>86.3</td>
<td>0.70/0.67</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>Model</th>
<th>Measurement</th>
<th>r Acc.</th>
<th>r/ρ Acc./F1</th>
<th>Accuracy</th>
</tr>
</thead>
<tbody>
<tr>
<td>BYTE m-LSTM</td>
<td></td>
<td>0.7920</td>
<td>-</td>
<td>75.0/82.8</td>
</tr>
</tbody>
</table>

**Supervised training - Transfer learning**

<table>
<thead>
<tr>
<th>Model</th>
<th>Measurement</th>
<th>r Acc.</th>
<th>r/ρ Acc./F1</th>
<th>Accuracy</th>
</tr>
</thead>
<tbody>
<tr>
<td>DiscSent</td>
<td>8</td>
<td>-</td>
<td>-</td>
<td>75.0/-</td>
</tr>
<tr>
<td>DisSent Books 8</td>
<td>0.8170</td>
<td>81.5</td>
<td>-/-</td>
<td>87.2</td>
</tr>
<tr>
<td>CaptionRep BOW</td>
<td>24</td>
<td>-</td>
<td>0.46/0.42</td>
<td>72.2</td>
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<tr>
<td>DictRep BOW</td>
<td>24</td>
<td>-</td>
<td>0.67/0.70</td>
<td>68.4/67.8</td>
</tr>
<tr>
<td>InferSent(SNLI)</td>
<td>&lt;24</td>
<td>0.8850</td>
<td>84.6</td>
<td>0.68/0.67</td>
</tr>
<tr>
<td>InferSent(AINLI)</td>
<td>&lt;24</td>
<td>0.8840</td>
<td>86.3</td>
<td>0.70/0.67</td>
</tr>
</tbody>
</table>

**Unsupervised training with unsorted sentences**

Table 2: **Related Work and Comparison.** As presented, our designed asymmetric RNN-CNN model has strong transferability, and is overall better than existing unsupervised models in terms of fast training speed and good performance on evaluation tasks. “†”s refer to our models, and “small/large” refers to the dimension of representation as 1200/4800. Bold numbers are the best ones among the models with same training and transferring setting, and underlined numbers are best results among all transfer learning models. The training time of each model was collected from the paper that proposed it.

vectors. Thus, the model benefits from the generalisation ability of the pretrained word embeddings. The downstream tasks for evaluation include semantic relatedness (SICK, Marelli et al. (2014)), paraphrase detection (MSRP, Dolan et al. (2004)), question-type classification (TREC, Li and Roth (2002)), and five benchmark sentiment and subjective datasets, which include movie review sentiment (MR, Pang and Lee (2005), SST, Socher et al. (2013)), customer product reviews (CR, Hu and Liu (2004)), subjectivity/objectivity classification (SUBJ, Pang and Lee (2004)), opinion polarity (MPQA, Wiebe et al. (2005)), semantic textual similarity (STS14, Agirre et al. (2014)), and SNLI (Bowman et al., 2015). After unsupervised training, the encoder is fixed, and applied as a representation extractor on the 10 tasks.

To compare the effect of different corpora, we also trained two models on Amazon Book Review dataset (without ratings) which is the largest subset of the Amazon Review dataset (Mcauley et al., 2015) with 142 million sentences, about twice as large as BookCorpus. Both training and evaluation of our models were conducted in PyTorch4, and we used SentEval5 provided by Conneau et al. (2017) to evaluate the transferability of our models. All the models were trained for the same number of iterations with the same batch size, and the performance was measured at the end of training for each of the models.

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4http://pytorch.org/
5https://github.com/facebookresearch/SentEval
5 Related work and Comparison

Table 2 presents the results on 9 evaluation tasks of our proposed RNN-CNN models, and related work. The “small RNN-CNN” refers to the model with the dimension of representation as 1200, and the “large RNN-CNN” refers to that as 4800. The results of our “large RNN-CNN” model on SNLI is presented in Table 3.

<table>
<thead>
<tr>
<th>Model</th>
<th>SNLI (Acc %)</th>
</tr>
</thead>
<tbody>
<tr>
<td><strong>Unsupervised Transfer Learning</strong></td>
<td></td>
</tr>
<tr>
<td>skip-thoughts (Vendrov et al.)</td>
<td>81.5</td>
</tr>
<tr>
<td>large RNN-CNN BookCorpus</td>
<td>81.7</td>
</tr>
<tr>
<td>large RNN-CNN Amazon</td>
<td>81.5</td>
</tr>
<tr>
<td><strong>Supervised Training</strong></td>
<td></td>
</tr>
<tr>
<td>ESIM (Chen et al.)</td>
<td>86.7</td>
</tr>
<tr>
<td>DIIN (Gong et al.)</td>
<td>88.9</td>
</tr>
</tbody>
</table>

Table 3: We implemented the same classifier as mentioned in Vendrov et al. (2015) on top of the features computed by our model. Our proposed RNN-CNN model gets similar result on SNLI as skip-thoughts, but with much less training time.

Our work was inspired by analysing the skip-thoughts model (Kiros et al., 2015). The skip-thoughts model successfully applied this form of learning from the context information into unsupervised representation learning for sentences, and then, Ba et al. (2016) augmented the LSTM with proposed layer-normalisation (Skip-thought+LN), which improved the skip-thoughts model generally on downstream tasks. In contrast, Hill et al. (2016) proposed the FastSent model which only learns source and target word embeddings and is an adaptation of Skip-gram (Mikolov et al., 2013) to sentence-level learning without word order information. Gan et al. (2017) applied a CNN as the encoder, but still applied LSTMs for decoding the adjacent sentences, which is called CNN-LSTM.

Our RNN-CNN model falls in the same category as it is an encoder-decoder model. Instead of decoding the surrounding two sentences as in skip-thoughts, FastSent and the compositional CNN-LSTM, our model only decodes the subsequent sequence with a fixed length. Compared with the hierarchical CNN-LSTM, our model showed that, with a proper model design, the context information from the subsequent words is sufficient for learning sentence representations. Particularly, our proposed small RNN-CNN model runs roughly three times faster than our implemented skip-thoughts model on the same GPU machine during training.

Proposed by Radford et al. (2017), BYTE m-LSTM model uses a multiplicative LSTM unit (Krause et al., 2016) to learn a language model. This model is simple, providing next-byte prediction, but achieves good results likely due to the extremely large training corpus (Amazon Review data, McAuley et al. (2015)) that is also highly related to many of the sentiment analysis downstream tasks (domain matching).

We experimented with the Amazon Book review dataset, the largest subset of the Amazon Review. This subset is significantly smaller than the full Amazon Review dataset but twice as large as BookCorpus. Our RNN-CNN model trained on the Amazon Book review dataset resulted in performance improvement on all single-sentence classification tasks relative to that achieved with training under BookCorpus.

Unordered sentences are also useful for learning representations of sentences. ParagraphVec (Le and Mikolov, 2014) learns a fixed-dimension vector for each sentence by predicting the words within the given sentence. However, after training, the representation for a new sentence is hard to derive, since it requires optimising the sentence representation towards an objective. SDAE (Hill et al., 2016) learns the sentence representations with a denoising auto-encoder model. Our proposed RNN-CNN model trains faster than SDAE does, and also because we utilised the sentence-level continuity as a supervision which SDAE doesn’t, our model largely performs better than SDAE.

Another transfer approach is to learn a supervised discriminative classifier by distinguishing whether the sentence pair or triple comes from the same context. Li and Hovy (2014) proposed a model that learns to classify whether the input sentence triplet contains three contiguous sentences. DiscSent (Jernite et al., 2017) and DisSent (Nie et al., 2017) both utilise the annotated explicit discourse relations, which is also good for learning sentence representations. It is a very promising research direction since the proposed models are generally computational efficient and have clear intuition, yet more investigations need to be done to augment the performance.

Supervised training for transfer learning is
also promising when a large amount of human-annotated data is accessible. Conneau et al. (2017) proposed the InferSent model, which applies a bi-directional LSTM as the sentence encoder with multiple fully-connected layers to classify whether the hypothesis sentence entails the premise sentence in SNLI (Bowman et al., 2015), and MultiNLI (Williams et al., 2017). The trained model demonstrates a very impressive transferability on downstream tasks, including both supervised and unsupervised. Our RNN-CNN model trained on Amazon Book Review data in an unsupervised way has better results on supervised tasks than InferSent but slightly inferior results on semantic relatedness tasks. We argue that labelling a large amount of training data is time-consuming and costly, while unsupervised learning provides great performance at a fraction of the cost. It could potentially be leveraged to initialise or more generally augment the costly human labelling, and make the overall system less costly and more efficient.

6 Discussion

In Hill et al. (2016), internal consistency is measured on five single sentence classification tasks (MR, CR, SUBJ, MPQA, TREC), MSRP and STS-14, and was found to be only above the “acceptable” threshold. They empirically showed that models that worked well on supervised evaluation tasks generally didn’t perform well on unsupervised ones. This implies that we should consider supervised and unsupervised evaluations separately, since each group has higher internal consistency.

As presented in Table 2, the encoders that only sum over pretrained word vectors perform better overall than those with RNNs on unsupervised evaluation tasks, including STS14. In recent proposed log-bilinear models, such as FastSent (Hill et al., 2016) and SiameseBOW (Kenter et al., 2016), the sentence representation is composed by summing over all word representations, and the only tunable parameters in the models are word vectors. These resulting models perform very well on unsupervised tasks. By augmenting the pretrained word vectors with a weighted averaging process, and removing the top few principal components, which mainly encode frequently-used words, as proposed in Arora et al. (2017) and Mu et al. (2018), the performance on the unsupervised evaluation tasks gets even better. Prior work suggests that incorporating word-level information helps the model to perform better on cosine distance based semantic textual similarity tasks.

Our model predicts all words in the target sequence at once, without an autoregressive process, and ties the word embedding layer in the encoder with the prediction layer in the decoder, which explicitly uses the word vectors in the target sequence as the supervision in training. Thus, our model incorporates the word-level information by using word vectors as the targets, and it improves the model performance on STS14 compared to other RNN-based encoders.

Arora et al. (2017) conducted an experiment to show that the word order information is crucial in getting better results on supervised tasks. In our model, the encoder is still an RNN, which explicitly utilises the word order information. We believe that the combination of encoding a sentence with its word order information and decoding all words in a sentence independently inherently leverages the benefits from both log-linear models and RNN-based models.

7 Conclusion

Inspired by learning to exploit the contextual information present in adjacent sentences, we proposed an asymmetric encoder-decoder model with a suite of techniques for improving context-based unsupervised sentence representation learning. Since we believe that a simple model will be faster in training and easier to analyse, we opt to use simple techniques in our proposed model, including 1) an RNN as the encoder, and a predict-all-words CNN as the decoder, 2) learning by inferring subsequent contiguous words, 3) mean+max pooling, and 4) tying word vectors with word prediction. With thorough discussion and extensive evaluation, we justify our decision making for each component in our RNN-CNN model. In terms of the performance and the efficiency of training, we justify that our model is a fast and simple algorithm for learning generic sentence representations from unlabelled corpora. Further research will focus on how to maximise the utility of the context information, and how to design simple architectures to best make use of it.
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Connecting Supervised and Unsupervised Sentence Embeddings

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Abstract

Representing sentences as numerical vectors while capturing their semantic context is an important and useful intermediate step in natural language processing. Representations that are both general and discriminative can serve as a tool for tackling various NLP tasks.

While common sentence representation methods are unsupervised in nature, recently, an approach for learning universal sentence representation in a supervised setting was presented in (Conneau et al., 2017). We argue that although promising results were obtained, an improvement can be reached by adding various unsupervised constraints that are motivated by auto-encoders and by language models. We show that by adding such constraints, superior sentence embeddings can be achieved. We compare our method with the original implementation and show improvements in several tasks.

1 Introduction

Word embeddings are considered one of the key building blocks in natural language processing and are widely used for various applications (Mikolov et al., 2013; Pennington et al., 2014). While word representations have been successfully used, representing the more complicated and nuanced nature of the next element in the hierarchy - a full sentence - is still considered a challenge. Once trained, universal sentence representations can be used as an out-of-the-box tool for solving various NLP and computer vision problems. Even though their importance is unquestionable, it seems that current results are still far from satisfactory.

More concretely, given a set of sentences \( \{s_i\}^n_{i=1} \), sentence embedding methods are designed to map them to some feature space \( \mathcal{F} \) along with a distance metric \( \mathcal{M} \) such that given two sentences \( s_i \) and \( s_j \) that have similar semantic meaning, their distance \( \mathcal{M}(s_i, s_j) \) would be small. The challenge is learning a mapping \( \mathbf{T} : \{s_i\}^n_{i=1} \rightarrow \mathcal{F} \) that manages to capture the semantics of each \( s_i \). While sentence embedding are not always used in similarity probing, we find this formulation useful as the similarity assumption is implicitly made when training classifiers on top of the embeddings in downstream tasks.

Sentences embedding methods were mostly trained in an unsupervised setting. In (Le and Mikolov, 2014) the ParagraphVector model was proposed which is trained to predict words in the document. SkipThought (Kiros et al., 2015) vectors rely on the continuity of text to train an encoder-decoder model that tries to reconstruct the surrounding sentences of a given passage. In Sequential Denoising Autoencoders (SDAE) (Hill et al., 2016) high-dimensional input data is corrupted according to some noise function, and the model is trained to recover the original data from the corrupted version. FastSent (Hill et al., 2016) learns to predicts a Bag-Of-Word (BOW) representation of adjacent sentences given a BOW representation of some sentence. In (Klein et al., 2015) a Hybrid Gaussian Laplacian density function is fitted to the sentence to derive Fisher Vectors.

While previous methods train sentence embeddings in an unsupervised manner, a recent work (Conneau et al., 2017) argued that better representations can be achieved via supervised training on a general sentence inference dataset (Bowman et al., 2015). To this end, the authors use the Stanford Natural Language Inference (SNLI) dataset (Bowman et al., 2015) to train different models.
sentence embedding methods and compare them on various benchmarks. The SNLI dataset is composed of 570K pairs of sentences with a label depicting the relationship between them, which can be either 'neutral', 'contradiction' or 'entailment'. The authors show that by leveraging the dataset, state-of-the-art representations can be obtained which are universal and general enough for solving various NLP tasks.

A different, unsupervised, task in NLP is estimating the probability of word sequences. A family of algorithms for this task titled word language models seek to model the problem as estimating the probability of a word, given the previous words in the text. In (Bengio et al., 2003) neural networks were employed and (Mikolov et al., 2010) was among the first methods to use recurrent neural networks (RNN) for modeling the problem, where the probability of the a word is estimated based on the previous words fed to the RNN. A variant of RNN - Long Short Term Memory (LSTM) networks (Hochreiter and Schmidhuber, 1997) - were used in (Sundermeyer et al., 2012). Following that, (Zaremba et al., 2014) proposed a dropout augmented LSTM.

We note that there exists a connection between those two problems and try to model it more explicitly. Recently, the incorporation of the hidden states of neural language models in downstream supervised-learning models have been shown to improve the results of the latter (e.g. ElMo - Peters et al. (2018), CoVe - McCann et al. (2017) Peters et al. (2017), Salant and Berant (2017) – in this work we jointly train the unsupervised and supervised tasks. To this end, we incorporate unsupervised regularization terms motivated by language modeling and auto-encoders in the training framework proposed by (Conneau et al., 2017). We test our proposed model on a set of NLP tasks and show improved results over the baseline framework of (Conneau et al., 2017).

## 2 Method

Our approach builds upon the previous work of (Conneau et al., 2017). Specifically, we use their BiLSTM model with max pooling. More concretely, given a sequence of T words, \( \{w_t\}_{t=1,...,T} \) with given word embedding (Mikolov et al., 2013; Pennington et al., 2014) \( \{v_t\}_{t=1,...,T} \), a bidirectional LSTM computes a set of T vectors \( \{h_t\}_{t=1,...,T} \) where each \( h_t \) is the concatenation of a forward LSTM and a backward LSTM that read the sentences in two opposite directions. We denote \( \{\vec{h}_t\} \) and \( \{\hat{h}_t\} \) as the hidden states of the left and right LSTM’s respectively, where \( t = 1, \ldots, T \). The final sentence representation is obtained by taking the maximal value of each dimension of the \( \{h_t\} \) hidden units (i.e.: max pooling). The original model of (Conneau et al., 2017) was trained on the SNLI dataset in a supervised fashion - given pairs of sentences \( s_1 \) and \( s_2 \), denote their representation by \( \vec{s}_1 \) and \( \hat{s}_2 \). During training, the concatenation of \( \vec{s}_1, \hat{s}_2, |\vec{s}_1 - \hat{s}_2| \) and \( \vec{s}_1 \times \hat{s}_2 \) is fed to a three layer fully connected network followed by a softmax classifier.
2.1 Regularization terms

We note that by training on SNLI, the model might overfit and would not be general enough to provide universal sentence embedding. We devise several regularization criteria that incentivize the hidden states to maintain more information about the input sequence.

Specifically, denote the dimension of the word embedding by \( d \) and the dimension of the hidden state by \( l \). We add a linear transformation layer \( L_{l \times d} : H \to W \) on top of the BiLSTM to transform the hidden states back to the dimension of word embeddings and denote its output by \( \{w'_t\}_{t=1,...,T} \). Recall that in the training process, we minimize the log-likelihood loss of the fully connected network predictions which we denote by \( y_t \), where \( y_{gt} \) is the prediction score given to the correct ground truth class. Now, the total loss criteria with our regularization term can be written as

\[
\mathcal{L} = -\log \left( \frac{e^{y_{gt}}}{\sum_j e^{y_j}} \right) + \lambda \sum_{t=1}^{T} \|w'_t - w_t\|^2 \tag{1}
\]

or as

\[
\mathcal{L} = -\log \left( \frac{e^{y_{gt}}}{\sum_j e^{y_j}} \right) + \lambda \sum_{t=1}^{T-1} \|w'_t - w_{t+1}\|^2 \tag{2}
\]

where the first term in both (1) and (2) is the original classification loss. We call the second regularization term in (1) an auto-encoder regularization term and in (2) a language model regularization term. Intuitively, since each \( w'_t \) is obtained by a linear transformation of \( h_t \), it enforces the hidden state \( h_t \) to maintain enough information on each \( w_t \) such that it can be reconstructed back from \( h_t \) or such that the following word \( w_{t+1} \) can be predicted from \( h_t \). This aids in obtaining a more general sentence representation and mitigates the risk of overfitting to the SNLI training set. The constant \( \lambda \) in (1) and (2) is a hyper-parameter that controls the amount of regularization and was set to 1 in our experiments.

We have also experimented with combining the two terms, giving equal weight to each of them in optimizing the model.

2.2 Bi-directional Regularization terms

Similarly to regularization terms described in 2.1, we devise variants of (1) and (2) which take into account the bi-directional architecture of the model. Here, we add two linear transformation layers: \( \tilde{L}_{\frac{d}{2} \times d} : \tilde{H} \to \tilde{W} \) and \( \tilde{L}_{\frac{d}{2} \times d} : \tilde{H} \to W \) on top of the forward LSTM and backward LSTM, respectively, and denote their output as \( \{\tilde{w}'_t\} \) and \( \{\tilde{w}'_t\} \), respectively, where \( t = 1, \ldots, T \).

Now, equations (1) and (2) are re-written as:

\[
\mathcal{L} = -\log \left( \frac{e^{y_{gt}}}{\sum_j e^{y_j}} \right) + \lambda_1 \sum_{t=1}^{T} \|\tilde{w}'_t - w_t\|^2 \tag{3}
\]

\[
+ \lambda_2 \sum_{t=1}^{T} \|\tilde{w}'_t - w_t\|^2 
\]

and

\[
\mathcal{L} = -\log \left( \frac{e^{y_{gt}}}{\sum_j e^{y_j}} \right) + \lambda_1 \sum_{t=1}^{T-1} \|\tilde{w}'_t - w_{t+1}\|^2 \tag{4}
\]

\[
+ \lambda_2 \sum_{t=2}^{T} \|\tilde{w}'_t - w_{t-1}\|^2 
\]

We call the second regularization term in (3) a bi-directional auto-encoder regularizer and in (4) a bi-directional language model regularization term. Again, \( \lambda_1 \) and \( \lambda_2 \) are hyper-parameters controlling the amount of regularization and were set to 0.5 in our experiments.

3 Experiments

Following (Conneau et al., 2017) we have tested our approach on a wide array of classification tasks, including sentiment analysis (MR – Pang and Lee (2005), SST – Socher et al. (2013)), question-type (TREC – Li and Roth (2002)), product reviews (CR – Hu and Liu (2004)), subjectivity/objectivity (SUBJ – Pang and Lee (2005)) and opinion polarity (MPQA – Wiebe et al. (2005)). We also tested our approach on semantic textual similarity (STS 14 – Agirre et al. (2014)), paraphrase detection (MRPC – Dolan et al. (2004)), entailment and semantic relatedness tasks (SICK-R and SICK-E – Marelli et al. (2014)), though those tasks are more close in nature to the task of the SNLI dataset which the model was trained on. In our experiments we have set \( \lambda \) from eq. (1) and eq. (2) to be 1 and \( \lambda_1, \lambda_2 \) from eq. (3) and eq. (4) to be 0.5. All other hyper-parameters and implementation details were left unchanged to provide a fair comparison to the baseline method of (Conneau et al., 2017).
Our results are summarized in table 1. We compared our method against the baseline BiLSTM implementation of (Conneau et al., 2017) and included FastSent (Hill et al., 2016) and SkipThought vectors (Kiros et al., 2015) as a reference.

As evident from table 1 in almost all the tasks evaluated, adding the proposed regularization terms improves performance. This serve to show that in a supervised learning setting, additional information on the input sequence can be leveraged and injected to the model by adding simple unsupervised loss criteria.

4 Conclusions

In our work, we have sought to connect unsupervised and supervised learning in the context of sentence embeddings. Leveraging supervision given by some general task aided in obtaining state-of-the-art sentence representations (Conneau et al., 2017). However, every supervised learning tasks is prone to overfit. In this context, overfitting to the learning task will result in a model which generalizes less well to new tasks.

We alleviate this problem by incorporating unsupervised regularization criteria in the model’s loss function which are motivated by auto-encoders and language models. We note that the added regularization terms do come at the price of increasing the model size by ld parameters (where d and l are the dimensions of the word embedding and the LSTM hidden state, respectively) due to the added linear transformation (see 2.1). However, as evident from our results, this does not hinder the model performance, even though we did not increase the amount of training data. Moreover, since those term are unsupervised in nature, it is possible to pre-train the model on unlabeled data and then finetune it on the SNLI dataset.

In conclusion, our experiments show that adding the proposed regularization terms results in a more general model and superior sentence embeddings. This validates our assumption that while the a supervised signal is general enough for learning sentence embeddings, it can be further improved by incorporated a second unsupervised signal.

5 Acknowledgments

We would like to thank Shimi Salant and Ofir Press for their helpful comments.

References


A Hybrid Learning Scheme for Chinese Word Embedding

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Abstract

To improve word embedding, subword information has been widely employed in state-of-the-art methods. These methods can be classified to either compositional or predictive models. In this paper, we propose a hybrid learning scheme, which integrates compositional and predictive model for word embedding. Such a scheme can take advantage of both models, thus effectively learning word embedding. The proposed scheme has been applied to learn word representation on Chinese. Our results show that the proposed scheme can significantly improve the performance of word embedding in terms of analogical reasoning and is robust to the size of training data.

1 Introduction

Word embedding, also known as distributed word representation, represents a word as a real-valued low-dimensional vector and encodes its semantic meaning into the vector. It is a fundamental task of natural language processing (NLP), such as language modeling (Bengio et al., 2003; Mnih and Hinton, 2009), machine translation (Bahdanau et al., 2014; Sutskever et al., 2014), caption generation (Xu et al., 2015; Devlin et al., 2015) and question answering (Hermann et al., 2015).

Most previous word embedding methods suffer from high computational complexity and have difficulty to be applied to large-scale corpora. Recently, Continuous Bag-Of-Words (CBOW) and Skip-Gram (SG) models (Mikolov et al., 2013a), which can alleviate the above issue, have received much attention. However, these models take a word as a basic unit but ignore rich subword information, which could significantly limit their performance. To improve the performance of word embedding, subword information, such as morphemes and character n-grams, has been employed (Luong et al., 2013; Qiu et al., 2014; Cao and Rei, 2016; Sun et al., 2016a; Wieting et al., 2016; Bojanowski et al., 2017). While these methods are effective, they are originally developed for alphabetic writing systems and can’t be applied directly to other writing systems, like Chinese.

In Chinese, each word typically consists of less characters than in English, while each character can have a complicated structure of its meaning. Typically, a Chinese character can be decomposed into components (部), where each component has its own meaning. The internal semantic meaning of a Chinese word emerges from such a structure. For example, the Chinese word “海水 (seawater)” is composed by “海 (sea)” and “水 (water)”. The semantic component of “海 (sea)” is “氵”, which is the transformation of “水 (water)” and indicates it is related to “水 (water)”. Therefore, the word “海水 (seawater)” has the meaning of “water from the sea”.

Based on the linguistic feature of Chinese, recent methods have used subword information to improve Chinese word embedding. For example, Chen et al. (2015) proposed a character-enhanced word embedding (CWE) model, which departed from CBOW of representing context words with both character embeddings and word embeddings. Shi et al. (2015) proposed a radical embedding method, which used the CBOW framework but replacing word embeddings with radical embeddings. Yin et al. (2016) and Xu et al. (2016) extended the CWE model in different ways: the former presented a multi-granularity embedding (MGE) model, additionally using the embeddings associated with radicals detected in the target word; the latter proposed a similarity-based character-enhanced word embedding (SCWE) model, considering the similarity between a word and its

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¹https://en.wikipedia.org/wiki/Written_Chinese
component characters. Yu et al. (2017) introduced a joint learning word embedding (JWE) model, which jointly learned embeddings for words, characters and components, and predicted the target word, respectively. Cao et al. (2018), on the other hand, represented Chinese words as sequences of strokes and learned word embedding with stroke n-grams information.

The above methods can be divided into two types: compositional and predictive model. The compositional model composes rich information into one vector to predict the target word. In this type of model, information works in a cooperative manner for word embedding. By contrast, the predictive model decouples various information to predict the target word. The information in this type of model works competitively for word embedding. Both models can effectively learn word embedding and give good estimation for rare and unseen words. By combining richer information, the compositional model can more accurately represent the target word. However, information is usually composed in a sophisticated way. The predictive model, on the other hand, is simple and can directly capture the interaction between words and their internal information. This type of model, however, typically ignores the interrelationship between various information.

To take advantage of both models, in this paper, we propose a hybrid learning scheme for word embedding. The proposed scheme learns word embedding in a competitive and cooperative manner. Specifically, in our scheme, the decoupled representations are used to capture the semantic meaning of target word respectively while making their composition semantically consistent with the target word. The performance of proposed scheme has been evaluated on Chinese in terms of word similarity and analogy tasks. The results show that our proposed scheme can effectively learn word representation and is robust to the size of training data.

2 Proposed Scheme

In this section, we present the details of our proposed hybrid learning scheme for word embedding. We denote the proposed scheme as **Co-Opetition Word Embedding (COWE)**. It consists of predictive and compositional parts, which will be described in subsection 2.1 and subsection 2.2, respectively. This is followed by describing the objective function.

The meaning of notation used in this section is as follows. We denote the training corpus as \( \mathcal{D} \), word vocabulary as \( \mathcal{W} \), character vocabulary as \( \mathcal{C} \), components vocabulary as \( \mathcal{P} \). Each word \( w \in \mathcal{W} \), character \( c \in \mathcal{C} \) and component \( p \in \mathcal{P} \) are associated with vectors \( w \in \mathbb{R}^d \), \( c \in \mathbb{R}^d \), \( p \in \mathbb{R}^d \), respectively, where \( d \) is the vector dimension. The characters and components in word \( w_i \) are denoted as \( c_{i|j} \) and \( p_{i|j} \), where \( |c_{i|j}| \) and \( |p_{i|j}| \) denote the number of characters and components in \( w_i \), respectively.

2.1 Predictive Part

In the predictive part, the compositions of context words, characters and components as well as compositions of characters and components in target word are used to predict the target word, as illustrated in Figure 1. These separate predictions by various compositions can be considered as competitions for the semantic meaning of target word. In order to maintain similar length between different compositions, COWE uses an average operation as the composition operation.

---

The goal of this part is to maximize the sum of log likelihoods of all predictive conditional probabilities:

$$L_p(w_i) = \sum_{k=1}^{5} \log p(w_i | h_{ik}),$$

where $h_{i1}, h_{i2}, h_{i3}, h_{i4}$ and $h_{i5}$ correspond to the above mentioned five compositions, respectively. Here, $h_{i1}$ is defined as:

$$h_{i1} = \frac{1}{2N} \sum_{-N \leq j \leq N, j \neq 0} w_{i+j},$$

where $N$ is the context window size. $h_{i2}, h_{i3}, h_{i4}$ and $h_{i5}$ are defined in a similar way. The conditional probability is defined using a softmax function as:

$$p(w_i | h_{ik}) = \frac{\exp(w_i \cdot h_{ik})}{\sum_{j \in \mathcal{W}} \exp(w_j \cdot h_{ik})}, \quad k = 1, 2, 3, 4, 5.$$  (3)

This objective function is similar to the one used in JWE (Yu et al., 2017). The main difference is that we further decouple components in the context words and target word, and leverage characters in the target word in addition.

### 2.2 Compositional Part

In the compositional part, all compositions mentioned above work in a cooperative manner, where their composition is used to predict the target word. We consider the composition as semantic consistency point of various representations, and the prediction loss as consistency loss, as shown in Figure 2.

The goal of this part is to maximize the following objective function:

$$L_c(w_i) = \log p(w_i | a_i),$$

where $a_i$ is the semantic consistency point, and is defined as:

$$a_i = \frac{1}{5} \sum_{k=1}^{5} h_{ik}.$$  (5)

Similar to the predictive part, the conditional probability is defined using the softmax function (see Equation (3)).

### 2.3 Objective Function

As COWE consists of predictive and compositional parts, its objective function is therefore consisted of the sum of all prediction losses and the consistency loss:

$$L(\mathcal{D}) = \sum_{w_i \in \mathcal{D}} L_p(w_i) + L_c(w_i).$$  (6)

To solve the above optimization problem, we employ the negative sampling technique (Mikolov et al., 2013b). Note that only the consistency loss between semantic consistency point and target word is considered. In preliminary experiments, we also tried the consistency losses between semantic consistency point and sampled negative words, but observed reduced performance.

As a result, the final objective function can be written as:

$$L(\mathcal{D}) = \sum_{w_i \in \mathcal{D}} \sum_{k=1}^{5} \log \sigma(w_i \cdot h_{ik}) + \lambda \mathbb{E}_{\tilde{w} \sim P_{\tilde{w}}}[\sum_{k=1}^{5} \log \sigma(\tilde{w} \cdot h_{ik})] + \log \sigma(w_i \cdot a_i),$$  (7)

where $\sigma$ is a sigmoid function: $\sigma(x) = 1/(1 + \exp(-x))$, $\lambda$ is the number of negative words, $\tilde{w}$ is the sampled negative word and $P_{\tilde{w}}$ is the distribution of negative words.

### 3 Experiments

In this section, we evaluate COWE on Chinese in terms of word similarity computation and analogical reasoning.

#### 3.1 Experimental Settings

We use Chinese Wikipedia dump dated on March 1, 2018 for embedding learning, which contains 310K Chinese Wikipedia articles. The data is pre-processed as follows. Firstly, construct training corpus from the Wikipedia dump with WikiCorpus in the gensim toolkit. Secondly, convert traditional Chinese characters to simplified Chinese characters with the opencc toolkit. Thirdly, remove all non-Chinese characters and Chinese words whose frequencies are less than 10

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3 https://dumps.wikimedia.org/zhwiki/20180301/
4 https://radimrehurek.com/gensim/corpora/wikicorpus.html
5 https://github.com/BYVoid/OpenCC
in the corpus. Finally, perform Chinese word segmentation with THULAC\(^6\) (Sun et al., 2016b). In addition, we perform POS tagging on the training corpus using THULAC and identify all entity names for CWE (Chen et al., 2015), as it does not use the character information for non-compositional words. We use the subword files provided by Yu et al. (2017). As a result, we obtain a 1 GB training corpus with 165,507,601 words, 368,408 unique words, 20,885 unique characters and 13,232 unique components.

We compare COWE with CBOW (Mikolov et al., 2013a)\(^7\), CWE (Chen et al., 2015)\(^8\) and JWE (Yu et al., 2017)\(^9\). To further evaluate the effect of consistency loss and components, we create two variants of COWE, denoted as COWE-c2 and COWE-p. The former is indeed the JWE model with an additional consistency loss, while the latter is COWE without using component information. The same parameter settings are used for all models. Specifically, the vector dimension is set to 200, the training iteration is set to 100, both the size of context window and number of negative samples are set to 5, the initial learning rate is set to 0.025, and the subsampling threshold is set to 10\(^{-4}\).

### 3.2 Word Similarity

This task is to evaluate the effectiveness of word embedding in capturing semantic similarity of word pairs. Following Yu et al. (2017), we adopt wordsim-240 and wordsim-296 datasets (Jin and Wu, 2012). Both datasets contain manually-annotated similarity scores for word pairs. In wordsim-240, words in 234 pairs appear in the training corpus, and in wordsim-296, words in 286 pairs appear in the training corpus. Unseen words are removed. The performance of word embedding is evaluated by ranking the pairs according to their cosine similarity and measuring the Spearman correlation \(\rho\) with human ratings. The results are shown in Table 1.

The results, on the wordsim-240 dataset, show that CWE performs better than CBOW, but outperformed by all other models. This could indicate the benefits of using rich information. COWE-c2 is not so good as JWE, COWE-p and COWE perform even worse. This suggests that the introduction of consistency loss, to some extent, may limit the performance of word representation. This may be due to the fact that our average semantic consistency point considers the contributions of various representations equally. With the evolution of history, however, meanings of some Chinese characters or components have degraded, making them less expressive. We plan to investigate the composition operation further in future work.

### 3.3 Word Analogy

This task is to evaluate the effectiveness of word embedding in capturing semantic relations between pairs of words. The goal is to answer the analogy questions of the form “a is to a* as b is to b*”, where b* is hidden, and must be reasoned out from the vocabulary. We use the Chinese word analogy dataset provided by Chen et al. (2015). It consists of 1,124 analogy questions, categorized into 3 types: 1) capitals of countries (677 groups), 2) capitals of provinces/states (175 groups), and 3) family relationships (272 groups). The analogy questions are answered using 3CosAdd (Mikolov et al., 2013a) as well as 3CosMul (Levy and Goldberg, 2014)\(^10\). We abbreviate the two methods as “Add” and “Mul”, respectively. The evaluation metric for this task is the percentage of questions for which the argmax result is the correct answer b*. The results are shown in Table 2\(^11\).

It can be found that CBOW performs better than CWE and JWE on the Capital and Family tasks. This is due to that using internal information improperly could be harmful in cases where words are non-compositional or irrelevant words sharing similar internal structures. For example, the words “儿子 (son)” and “妻子 (wife)” share the same character “子”, which means “son” in the former but makes no sense in the latter. We observe that COWE-c2 achieves the best results.

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\(^6\) http://thulac.thunlp.org/

\(^7\) https://code.google.com/archive/p/word2vec/

\(^8\) https://github.com/Leonard-Xu/CWE

\(^9\) https://github.com/HKUST-KnowComp/JWE

\(^10\) https://bitbucket.org/omerlevy/hyperwords

\(^11\) The results do not agree with that reported in (Yu et al., 2017). We suggest that these discrepancies stem from differences in training corpus and parameter settings.

---

### Table 1: Results on word similarity evaluation.

<table>
<thead>
<tr>
<th>Model</th>
<th>wordsim-240</th>
<th>wordsim-296</th>
</tr>
</thead>
<tbody>
<tr>
<td>CBOW</td>
<td>0.4861</td>
<td>0.5658</td>
</tr>
<tr>
<td>CWE</td>
<td>0.5515</td>
<td>0.5684</td>
</tr>
<tr>
<td>JWE</td>
<td><strong>0.5496</strong></td>
<td><strong>0.6355</strong></td>
</tr>
<tr>
<td>COWE-c2</td>
<td>0.5473</td>
<td>0.5899</td>
</tr>
<tr>
<td>COWE-p</td>
<td>0.5180</td>
<td>0.5844</td>
</tr>
<tr>
<td>COWE</td>
<td>0.5412</td>
<td>0.5674</td>
</tr>
</tbody>
</table>
on the Family task and outperforms JWE by large margins. This shows the effectiveness of consistency loss in helping with learning from various information. COWE-p and COWE perform best on the other tasks, respectively. The fact suggests that different information could help in different ways.

### 3.4 Performance on Low-Resource Corpora

To evaluate the performance of different models on low-resource corpora, we conduct the same experiments on 5%, 10% and 20% randomly selected Wikipedia articles, respectively. As less training data introducing more noises, this makes it more difficult for models to learn good word representations. The results are shown in Table 3.

The results indicate the superiority of our models on low-resource corpora. We observe that as the size of dataset decreases, the performance of baselines drops rapidly, while the performance decrement of COWE and its variants is much smaller. This shows the robustness of our proposed models. COWE-p is generally more robust than COWE-c2, however, COWE-c2 performs more robustly on the Family task. Taking both characters and components into account, COWE achieves the most robust results.

We also observe that on the Capital task, the performance of CWE and JWE drops more quick than CBOW, which agrees with the previous findings. However, with the consistency loss, COWE-c2 always performs better than JWE, and usually outperforms CBOW. We believe that the consistency loss, in cases where some embeddings are useless, would encourage weak embeddings to close to strong embeddings, letting weak embeddings acquire some helpful features, and prevent strong embeddings from overfitting. On the State and Family tasks, where the character and component embeddings could be useful, all of our models still outperform the baselines by large margins. This should be due to the fact that the consistency loss prevents various learned embeddings from contradicting each other, thus making all of them close to the true target word embedding.

### 3.5 Case Study

To gain a better understanding of the quality of learned word embedding, we take the word “癌症 (cancer)” as an example and show its nearest neighbors in Table 4, where cosine similarity is used as the distance metric.

All words yielded by different models are disease-related. Specifically, words yielded by CWE contain the character “癌 (cancer),” including some weird words, like “国家癌症 (national cancer)” and “抗癌 (anti-cancer)”. This implies that CWE has overused the internal information. For

---

Table 2: Results on word analogy evaluation.

<table>
<thead>
<tr>
<th>Model</th>
<th>Capital Add/Mul</th>
<th>Capital Add/Mul</th>
<th>Family Add/Mul</th>
</tr>
</thead>
<tbody>
<tr>
<td>CBOW</td>
<td>70.80/85.82</td>
<td>93.14/92.00</td>
<td>76.84/73.90</td>
</tr>
<tr>
<td>CWE</td>
<td>86.71/85.08</td>
<td>91.43/90.29</td>
<td>75.74/70.96</td>
</tr>
<tr>
<td>JWE</td>
<td>86.12/83.90</td>
<td><strong>94.29/94.29</strong></td>
<td>70.96/69.49</td>
</tr>
<tr>
<td>COWE-c2</td>
<td>83.16/83.31</td>
<td>90.29/86.29</td>
<td><strong>77.94/74.63</strong></td>
</tr>
<tr>
<td>COWE-p</td>
<td>87.74/85.82</td>
<td>92.57/94.29</td>
<td>73.16/69.85</td>
</tr>
<tr>
<td>COWE</td>
<td>85.82/86.12</td>
<td><strong>94.29/93.71</strong></td>
<td>76.10/74.26</td>
</tr>
</tbody>
</table>

Table 3: Results on word analogy evaluation.

<table>
<thead>
<tr>
<th>Model</th>
<th>Capital Add/Mul</th>
<th>Capital Add/Mul</th>
<th>Family Add/Mul</th>
</tr>
</thead>
<tbody>
<tr>
<td>CBOW</td>
<td>57.46/52.14</td>
<td>28.00/23.43</td>
<td>34.19/29.04</td>
</tr>
<tr>
<td>CWE</td>
<td>51.99/47.12</td>
<td>36.00/31.43</td>
<td>16.18/13.60</td>
</tr>
<tr>
<td>JWE</td>
<td>44.76/40.77</td>
<td>49.14/44.57</td>
<td>31.99/27.57</td>
</tr>
<tr>
<td>COWE-c2</td>
<td>61.74/58.64</td>
<td>67.43/65.14</td>
<td><strong>44.49/35.29</strong></td>
</tr>
<tr>
<td>COWE-p</td>
<td>79.17/77.40</td>
<td>80.00/81.71</td>
<td>37.87/37.50</td>
</tr>
<tr>
<td>COWE</td>
<td>78.14/79.03</td>
<td><strong>81.71/82.86</strong></td>
<td>41.91/41.18</td>
</tr>
</tbody>
</table>

(a) 5% Wikipedia articles

<table>
<thead>
<tr>
<th>Model</th>
<th>Capital Add/Mul</th>
<th>Capital Add/Mul</th>
<th>Family Add/Mul</th>
</tr>
</thead>
<tbody>
<tr>
<td>CBOW</td>
<td>73.12/69.42</td>
<td>54.29/50.29</td>
<td>48.90/43.38</td>
</tr>
<tr>
<td>CWE</td>
<td>66.03/64.40</td>
<td>54.29/53.14</td>
<td>39.71/37.13</td>
</tr>
<tr>
<td>JWE</td>
<td>63.81/63.22</td>
<td>62.86/58.29</td>
<td>40.07/36.40</td>
</tr>
<tr>
<td>COWE-c2</td>
<td>70.16/67.50</td>
<td>77.71/73.14</td>
<td>59.19/56.62</td>
</tr>
<tr>
<td>COWE-p</td>
<td>77.70/78.58</td>
<td>79.43/80.00</td>
<td>54.78/52.21</td>
</tr>
<tr>
<td>COWE</td>
<td>78.43/78.58</td>
<td><strong>80.00/77.71</strong></td>
<td>60.29/56.99</td>
</tr>
</tbody>
</table>

(b) 10% Wikipedia articles

<table>
<thead>
<tr>
<th>Model</th>
<th>Capital Add/Mul</th>
<th>Capital Add/Mul</th>
<th>Family Add/Mul</th>
</tr>
</thead>
<tbody>
<tr>
<td>CBOW</td>
<td>70.75/68.39</td>
<td>69.71/64.57</td>
<td>59.19/54.78</td>
</tr>
<tr>
<td>CWE</td>
<td>67.80/65.58</td>
<td>66.29/63.43</td>
<td>50.00/44.49</td>
</tr>
<tr>
<td>JWE</td>
<td>70.46/69.28</td>
<td>81.71/78.86</td>
<td>48.90/48.16</td>
</tr>
<tr>
<td>COWE-c2</td>
<td>74.89/72.97</td>
<td><strong>90.29/87.43</strong></td>
<td>59.93/56.25</td>
</tr>
<tr>
<td>COWE-p</td>
<td>81.83/81.83</td>
<td>89.71/86.86</td>
<td>58.46/54.78</td>
</tr>
<tr>
<td>COWE</td>
<td><strong>84.79/83.60</strong></td>
<td>87.43/86.86</td>
<td>58.46/55.51</td>
</tr>
</tbody>
</table>

(c) 20% Wikipedia articles

---

12 Translation by Google Translate.
JWE and COWE, which directly capture the interaction between the words and their internal information, they yield disease-related words that do not contain the component “疒”, such as “肺炎” (pneumonia). This indicates that they make full use of external and internal information, and avoid the above issue. Compared to JWE, COWE yields more words that are semantically relevant to the target word.

4 Conclusion

This paper proposes a scheme, which combines predictive and compositional models to jointly learn various word representations in a competitive and cooperative manner. The predictive part of the proposed scheme is based on various external and internal information, which is used to capture corresponding representation. In the compositional part, the semantic consistency point and the consistency loss are introduced. They connect separate learned representations and prevent them from contradicting each other. The experimental results show that the proposed scheme outperforms baseline models on word analogy tasks and achieves competitive results on word similarity tasks. The results also show that our model is robust to the size of training data. Therefore, our proposed scheme is suitable to be applied on low-resource corpora, for example task-specific corpora, where data is often very scarce.

Acknowledgements

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References


Unsupervised Random Walk Sentence Embeddings: 
A Strong but Simple Baseline

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Abstract
Using a random walk model of text generation, Arora et al. (2017) proposed a strong baseline for computing sentence embeddings: take a weighted average of word embeddings and modify with SVD. This simple method even outperforms far more complex approaches such as LSTMs on textual similarity tasks. In this paper, we first show that word vector length has a confounding effect on the probability of a sentence being generated in Arora et al.’s model. We propose a random walk model that is robust to this confound, where the probability of word generation is inversely related to the angular distance between the word and sentence embeddings. Our approach beats Arora et al.’s by up to 44.4% on textual similarity tasks and is competitive with state-of-the-art methods. Unlike Arora et al.’s method, ours requires no hyperparameter tuning, which means it can be used when there is no labelled data.

1 Introduction
Distributed representations of words, better known as word embeddings, have become fixtures of current methods in natural language processing. Word embeddings can be generated in a number of ways (Bengio et al., 2003; Collobert and Weston, 2008; Pennington et al., 2014; Mikolov et al., 2013) by capturing the semantics of a word using the contexts it appears in. Recent work has tried to extend that intuition to sequences of words, using methods ranging from a weighted average of word embeddings to convolutional, recursive, and recurrent neural networks (Le and Mikolov, 2014; Kiros et al., 2015; Luong et al., 2013; Tai et al., 2015). Still, Wieting et al. (2016b) found that these sophisticated architectures are often outperformed, particularly in transfer learning settings, by sentence embeddings generated as a simple average of tuned word embeddings.
Arora et al. (2017) provided a more powerful approach: compute the sentence embeddings as weighted averages of word embeddings, then subtract from each one the vector projection on their first principal component. The weighting scheme, smoothed inverse frequency (SIF), is derived from a random walk model where words in a sentence $s$ are produced by the random walk of a latent discourse vector $c_s$. A word unrelated to $c_s$ can be produced by chance or if it is part of frequent discourse such as stopwords. This approach even outperforms more complex models such as LSTMs on textual similarity tasks. Arora et al. argued that the simplicity and effectiveness of their method make it a tough-to-beat baseline for sentence embeddings. Though they call their approach unsupervised, others have noted that it is actually ‘weakly supervised’, since it requires hyperparameter tuning (Cer et al., 2017).

In this paper, we first propose a class of worst-case scenarios for Arora et al.’s (2017) random walk model. Specifically, given some sentence $g$ that is dominated by words with zero similarity, and some sentence $h$ that is dominated by identical words, we show that their approach can return two discourse vectors $c_g$ and $c_h$ such that $p(g|c_g) \approx p(h|c_h)$, provided that the word vectors for $g$ have a sufficiently greater length than those for $h$. In other words, word vector length has a confounding effect on the probability of a sentence being generated, and this effect can be strong enough to yield completely unintuitive results. This problem is not endemic to these scenarios, though they are the most illustrative of it; because of the underlying log-linear word production model, Arora et al.’s model is fundamentally
sensitive to word vector length.

Our contributions in this paper are three-fold. First, we propose a random walk model that is robust to distortion by vector length, where the probability of a word vector being generated by a discourse vector is inversely related to the angular distance between them. Second, we derive a weighting scheme from this model and compute a MAP estimate for the sentence embedding as follows: normalize the word vectors, take a weighted average of them, and then subtract from each weighted average vector the projection on their first $m$ principal components. We call the weighting scheme derived from our random walk model unsupervised smoothed inverse frequency (uSIF). It is similar to SIF (Arora et al., 2017) in practice, but requires no hyperparameter tuning at all – it is completely unsupervised, allowing it to be used when there is no labelled data. Lastly, we show that our approach outperforms Arora et al.’s by up to 44.4% on textual similarity tasks, and is even competitive with state-of-the-art methods. Given the simplicity, effectiveness, and unsupervised nature of our method, we suggest it be used as a baseline for computing sentence embeddings.

2 Related Work

Word Embeddings Word embeddings are distributed representations of words, typically in a low-dimensional continuous space. These word vectors can capture semantic and lexical properties of words, even allowing some relationships to be captured algebraically (e.g., $v_{\text{Berlin}} - v_{\text{Germany}} + v_{\text{France}} \approx v_{\text{Paris}}$) (Mikolov et al., 2013). Word embeddings are generally obtained in two ways: (a) from internal representations of words in shallow neural networks (Bengio et al., 2003; Mikolov et al., 2013; Collobert and Weston, 2008); (b) from low rank approximations of co-occurrence matrices (Pennington et al., 2014).

Word Sequence Embeddings Embeddings for sequences of words (e.g., sentences) are created by composing word embeddings. This can be done simply, by doing coordinate-wise multiplication (Mitchell and Lapata, 2008) or taking an unweighted average (Mikolov et al., 2013) of the word vectors. More sophisticated architectures can also be used: for instance, recursive neural networks (Socher et al., 2011, 2013), LSTMs (Tai et al., 2015), and convolutional neural networks (Kalchbrenner et al., 2014) can be defined and trained on parse and dependency trees.

Other approaches are based on the presence of a latent vector for the entire sequence. Paragraph vectors (Le and Mikolov, 2014) are latent representations that influence the distribution of words. Skip-thought vectors (Kiros et al., 2015) are hidden representations of a neural network that encodes a sentence by trying to reconstruct its surrounding sentences. Conneau et al. (2017) leverage transfer learning by using the hidden representation of a sentence in an LSTM trained for another task, such as textual entailment. The inspiration for Arora et al. (2017) is Wieting et al. (2016b), who use word averaging after updating word embeddings by tuning them on paraphrase pairs. A later work by Wieting et al. (2017a) tried trigram-averaging and LSTM-averaging in addition to word-averaging. In that approach, vectors were tuned on the ParaNMT-50M dataset, created by using neural machine translation to translate 51M Czech-English sentence pairs into English-English pairs. This yielded state-of-the-art results on textual similarity tasks, beating the previous baseline by a wide margin.

3 Approach

3.1 The Log-Linear Random Walk Model

In Arora et al.’s original model (2016), words are generated dynamically by the random walk of a time-variant discourse vector $c_t \in \mathbb{R}^d$, representing “what is being talked about”. Words are represented as $v_w \in \mathbb{R}^d$. The probability of a word $w$ being generated at time $t$ is given by a log-linear production model (Mnih and Hinton, 2007):

$$p(w|c_t) \propto \exp((c_t, v_w))$$

(1)

Assuming that the discourse vector $c_t$ does not change much over the course of the sentence, Arora et al. replace the sequence of discourse vectors $\{c_t\}$ across all time steps with a single discourse vector $c_s$. The MAP estimate of $c_s$ is then the unweighted average of word vectors (ignoring any scalar multiplication).

Arora et al.’s improved random walk model (2017) allows words to also be generated: (a) by chance, with probability $\alpha \cdot p(w)$, where $\alpha$ is some scalar and $p(w)$ is the frequency; (b) if the word is correlated with the common discourse vector, which represents frequent discourse such as stop-words. We use $c_0$ to denote the common discourse vector, to be consistent with the literature. Among
other things, these changes help explain words that appear frequently despite being poorly correlated with the discourse vectors — words like the, for example. The probability of a word \( w \) being generated by a discourse vector \( c_s \) is then given as:

\[
p(w|c_s) = \alpha \cdot p(w) + (1 - \alpha) \cdot \frac{\exp(\langle \tilde{c}_s, v_w \rangle)}{Z_{c_s}},
\]

where \( \tilde{c}_s \equiv \beta \cdot c_0 + (1 - \beta) \cdot c_s, c_0 \perp c_s \)

\[
Z_{c_s} \equiv \sum_{w \in V} \exp(\langle \tilde{c}_s, v_w \rangle)
\]

where \( \alpha, \beta \) are scalar hyperparameters, \( V \) is the vocabulary, \( \tilde{c}_s \) is a linear combination of the discourse and common discourse vectors parameterized by \( \beta \), and \( Z_{c_s} \) is the partition function.

The sentence embedding for a sentence is defined as the MAP estimate of the discourse vector \( c_s \) that generated the sentence. To compute this tractably, Arora et al. (2017) assume that word vectors \( v_w \) are roughly uniformly dispersed in the latent space. This implies that the partition function \( Z_{c_s} \) is roughly the same for all \( \tilde{c}_s \), allowing it to be replaced with a constant \( Z \). Assuming a uniform prior over \( \tilde{c}_s \), the maximum likelihood estimator for \( \tilde{c}_s \) on the unit sphere (ignoring normalization) is then approximately proportional to:

\[
\frac{1}{|s|} \sum_{w \in s} \frac{a}{a + p(w)} \cdot v_w, \text{ where } a \equiv \frac{1 - \alpha}{\alpha \cdot Z}
\]

Since \( Z \) cannot be evaluated, and \( \alpha \) is not known, \( a \) is a hyperparameter that needs tuning. This weighting scheme is called smoothed inverse frequency (SIF) and places a lower weight on more frequent words. The first principal component of all \( \{ \tilde{c}_s \} \) in the corpus is used as the estimate for the common discourse vector \( c_0 \). The final discourse vector \( c_s \) is then produced by subtracting the projection of the weighted average on the common component (common component removal):

\[
c_s \equiv \tilde{c}_s - \text{proj}_{c_0} \tilde{c}_s
\]

Arora et al. call their approach unsupervised, but others (Cer et al., 2017) have correctly noted that it is weakly supervised, since the hyperparameter \( a \) needs to be tuned on a validation set.

### 3.2 The Confounding Effect of Vector Length

We now propose worst-case scenarios where word vector length clearly distorts \( p(s|c_s) \) due to the underlying log-linear word production model. Note that we discuss these scenarios because they are illustrative, not because they circumscribe the universe of all scenarios in which word vector length has a confounding effect.

Consider a sentence \( g \) comprising two rare words \( x \) and \( y \), where \( x \) and \( y \) have zero similarity. Also consider some sentence \( h \), where the only word \( z \) appearing twice. \( g \) might not occur naturally, but its weighted average \( \tilde{c}_g \) would be similar to that of some longer sentence where \( x,y \) are the only non-stopwords (i.e., those with non-negligible weight). For simplicity, further assume that common component removal has negligible effect:

\[
\left\langle v_x, v_y \right\rangle = 0
\]

\[
c_g = \tilde{c}_g = \frac{a}{a + p(x)} \cdot v_x + \frac{a}{a + p(y)} \cdot v_y
\]

\[
c_h = \tilde{c}_h = \frac{a}{a + p(z)} \cdot v_z
\]

Words \( x, y, z \) are so infrequent that the probability of them being produced by chance or by the common discourse vector is negligible; the likelihood of them being produced is therefore proportional to the inner product of the discourse and word vectors. Given that the words \( x, y, z \) have zero similarity, and given that the only word \( z \in h \) is identical to its discourse vector, we would expect:

\[
p(h|c_h) \gg p(g|c_g)
\]

However, (3) does not always hold. Suppose that the word embeddings lie in \( \mathbb{R}^2 \). Then any scalar \( k \) can be used to create a valid set of assignments for word embeddings \( v_x, v_y, v_z \) that satisfy (2):

\[
v_x = \begin{bmatrix} 2k \\ 0 \end{bmatrix}, \quad v_y = \begin{bmatrix} 0 \\ 2k \end{bmatrix}, \quad v_z = \begin{bmatrix} k \\ k \end{bmatrix}
\]

Assuming the words \( x, y, z \) have roughly the same frequency, they should have the same SIF-weight. Then the weighted averages, and by extension the discourse vectors (2), are the same:

\[
c_g = c_h = \frac{a}{a + p(x)} \begin{bmatrix} k \\ k \end{bmatrix}
\]

\[
\Rightarrow \left\langle c_g, x \right\rangle = \left\langle c_g, y \right\rangle = \left\langle c_h, z \right\rangle = \frac{a}{a + p(x)} \cdot 2k^2
\]

\[
\Rightarrow p(g|c_g) = p(h|c_h)
\]

Thus it is possible for \( g \) to be generated by discourse vector \( c_g \) with roughly the same probability.
as \( h \) by \( c_h \), contradicting (3). How is this possible, given that the words in \( g \) have zero similarity with each other while those in \( h \) are identical to each other? The answer can be found in the word vector lengths. Because \( ||v_x|| = \sqrt{2}||v_y||_2 \), and \( p(w|c_x) \) depends on the inner product of the word and discourse vectors (1), words with longer word vectors are more likely to be produced. In fact, if \( v_x \) and \( v_y \) were multiplied by some scalar greater than 1, then \( p(h|c_h) \) would be less than \( p(g|c_g) \).

**Generalizing Worst-Case Scenarios** By manipulating the word vector length, we can also come up with a more general class of assignments that can contradict (3):

\[
v_x = \begin{bmatrix} \beta k_1 \sigma \\ \beta k_2 (1 - \sigma) \end{bmatrix}, \quad v_y = \begin{bmatrix} \beta k_1 (1 - \sigma) \\ \beta k_2 \sigma \end{bmatrix}, \quad v_z = \begin{bmatrix} k_1 \\ k_2 \end{bmatrix}
\]

(5)

where \( \sigma \in [0, 1], \beta \in \mathbb{R}, \beta \geq 2 \). For convenience, we replace \( C \) with \( C \) below:

\[
c_g = C \begin{bmatrix} 1/2 k_1 \\ 1/2 k_2 \end{bmatrix}, \quad c_h = C \begin{bmatrix} k_1 \\ k_2 \end{bmatrix}
\]

For simplicity, we assume that the words \( x, y, z \) across the two sentences are so infrequent that the probability of them being generated by chance is zero. Then the conditional probabilities of the sentences being generated are:

\[
p(g|c_g) \propto \exp (\langle c_g, v_x \rangle + \langle c_g, v_y \rangle) \\
= \exp \left( \frac{1}{2} \beta^2 C (k_1^2 + k_2^2) \right)
\]

\[
p(h|c_h) \propto \exp (\langle c_h, v_x \rangle + \langle c_h, v_z \rangle) \\
= \exp \left( 2C (k_1^2 + k_2^2) \right)
\]

\[
\therefore \beta \geq 2 \Rightarrow p(g|c_g) \geq p(h|c_h)
\]

(6)

In this general formulation, not all scenarios are worst-case. This describes a spectrum of scenarios ranging from acceptable (e.g., \( v_x = v_y = v_z \) when \( \beta = 2, \sigma = 0.5 \)) to completely counter-intuitive (see (4)). Though these assignments only apply for word vectors in \( \mathbb{R}^2 \), they can easily be extended to higher-dimensional spaces.

The confound of vector length persists for longer, naturally occurring sentences. Ultimately, the underlying log-linear word production model (1) means that words with longer word vectors are more likely to be generated. Because this confound is due to model design, rather than the MLE, removing it requires redesigning the model. The exact degree of the confound varies across sentences, but in theory, it is unbounded.

### 3.3 An Angular Distance–Based Random Walk Model

To address the confounding effect of word vector length, we propose a random walk model where the probability of observing a word \( w \) at time \( t \) is inversely related to the angular distance between the time-variant discourse vector \( c_t \in \mathbb{R}^d \) and the word vector \( v_w \in \mathbb{R}^d \):

\[
p(w|c_t) \propto 1 - \frac{\arccos (\cos (v_w, c_t))}{\pi},
\]

where \( \cos (v_w, c_t) \) is the angular distance. For the intuition behind the use of this distance metric, note that the angular distance between two vectors is equal to the geodesic distance between them on the unit sphere. Thus the angular distance can also be interpreted as the length of the shortest path between the \( L_2 \) normalized word vector and the \( L_2 \) normalized discourse vector on the unit sphere. Since the angular distance lies in \([0, \pi]\), we divide it by \( \pi \) to bound it in \([0, 1]\). Our choice of angular distance – as opposed to, say, the exponentiated cosine similarity – is critical to avoiding hyperparameter tuning.

Assuming that the discourse vector \( c_t \) does not change much over the course of the sentence, the sequence of discourse vectors \( \{c_i\} \) across all time steps can be replaced with a single discourse vector \( c_s \) for the sentence \( s \). To model sentences more realistically, we allow words to be generated in two additional ways, as proposed in Arora et al. (2017): (a) by chance, with probability \( \alpha \cdot p(w) \), where \( \alpha \) is some scalar and \( p(w) \) is the frequency; (b) if the word is correlated with one of \( m \) common discourse vectors \( \{c_i^m\} \), which represent various types of frequent discourse, such as stopwords. The probability of a word \( w \) being generated by discourse vector \( c_s \) is then:

\[
p(w|c_s) = \alpha \cdot p(w) + (1 - \alpha) \cdot \frac{d(\bar{c}_s, v_w)}{Z_{\bar{c}_s}},
\]

where \( \bar{c}_s \equiv (1 - \beta) c_s + \beta \sum_{i=1}^{m} \lambda_i c_i' \), \( c_s \perp c_i' \)

\[
d(\bar{c}_s, v_w) \equiv 1 - \frac{\arccos (\cos (v_w, c_s))}{\pi},
\]

\[
\sum_{w \in \mathcal{V}} d(\bar{c}_s, v_w)
\]

(8)

where \( \alpha, \beta, \{\lambda_i\} \) are scalar hyperparameters, \( \mathcal{V} \) is the vocabulary, \( \bar{c}_s \) is a linear combination of the
discourse and common discourse vectors parameterized by \( \beta \) and \( \{ \lambda_i \} \), and \( Z_c \) is the partition function. Instead of searching for the optimal hyperparameter values over some large space, as Arora et al. (2017) did, we make some simple assumptions to directly compute them.

We define the sentence embedding for some sentence \( s \) to be the MAP estimate of the discourse vector \( c_s \) that generates \( s \). Assuming a uniform prior over possible \( c_s \), the MAP estimate is also the MLE estimate for \( c_s \). The log-likelihood of a sentence \( s \) is:

\[
\log p(s|c_s) = \sum_{w \in s} \log p(w|c_s)
\]

To maximize \( \log p(s|c_s) \), we can approximate \( \log p(w|c_s) \) using a first-degree Taylor polynomial:

\[
f_w(\tilde{c}_s) \approx \log p(w|\tilde{c}_s) + \nabla f_w(\tilde{c}_s) \cdot (c'_w - \tilde{c}_s)
\]

\[
\frac{\partial}{\partial \tilde{c}_s} \log p(w|\tilde{c}_s) = \frac{\partial}{\partial \tilde{c}_s} \frac{1}{\pi \cdot Z_c} \cdot \exp(f_w(\tilde{c}_s)) \cdot \frac{1}{\sqrt{1 - \cos^2(v_w, \tilde{c}_s)}}
\]

Where \( a \equiv (1 - \alpha)/(\alpha Z_c) \), \( C \) is a constant, and \( \tilde{c}_s \) is a vector orthogonal to \( v_w \) with length \( \| v_w \|^{-1} \):

\[
f_w(\tilde{c}_s) = \frac{1}{\pi} \cdot (p(w) + \frac{1}{2} \cdot a) \cdot v_w (\tilde{c}_s - v'_w)
\]

\[
= C + \frac{1}{\pi} \frac{a}{p(w) + \frac{1}{2} \cdot a} (\tilde{c}_s, v_w)
\]

The MLE for \( \tilde{c}_s \) on the unit sphere (ignoring normalization) is then approximately proportional to:

\[
\frac{1}{|s|} \sum_{w \in s} \frac{a}{p(w) + \frac{1}{2} \cdot a} \cdot v_w
\]

The MLE of \( \tilde{c}_s \) is approximately a weighted average of word vectors, where more frequent words are down-weighted. In fact, it very closely resembles the SIF weighting scheme (Arora et al., 2017)! However, there are two key differences. For one, as we show later in this subsection, we have derived this weighting scheme from a model that is robust to the confounding effect of word vector length. Secondly, in SIF, \( a \) is a hyperparameter that needs to be tuned on a validation set.

We now show that in our approach, we can calculate \( a \) directly as a function of the vocabulary \( \mathcal{V} \) and the number of words in the sentence, \( |s| \).

**Normalization** Before weighting the word vectors, we normalize them along each dimension: we construct a matrix \( [v_{w_1}, ..., v_{w_n}] \) and take the \( L_2 \) norm of each row, which corresponds to a single dimension in \( \mathbb{R}^d \). We then multiply this \( d \)-dimensional vector element-wise with every vector in the sentence. This helps reduce the difference in variance across the dimensions.

**Partition Function** To calculate \( Z_c \), we borrow the key assumption from Arora et al. (2017) that the word vectors \( v_w \) are roughly uniformly dispersed in the latent space. Then the expected geodesic distance between a latent discourse vector and a word vector on the unit sphere is \( \pi/2 \), so

\[
\mathbb{E}_{w \in \mathcal{V}}[d(\tilde{c}_s, v_w)] = \frac{1}{2}
\]

Then:

\[
Z_c = \sum_{w \in \mathcal{V}} d(\tilde{c}_s, v_w)
\]

\[
= |\mathcal{V}| \mathbb{E}_{w \in \mathcal{V}}[d(\tilde{c}_s, v_w)] = \frac{1}{2} |\mathcal{V}|
\]

**Odds of Random Production** \( \alpha \) is the probability that a word \( w \) will be produced by chance instead of by the discourse or common discourse vectors. To estimate \( \alpha \), we first consider the probability that a random word \( w \) will be produced by a discourse vector \( c_s \) at least once over \( n \) steps of a random walk:

\[
p(w|c_s^1, ..., c_s^n) = 1 - \prod_{i=1}^n \left[ 1 - \frac{1 - d(c'_s, v_w)}{Z_{c_s}} \right]
\]

\[
\mathbb{E}_{w \sim \mathcal{V}}[p(w|c_s^1, ..., c_s^n)] = 1 - \left( 1 - \frac{1}{|\mathcal{V}|} \right)^n
\]

The number of steps taken during the random walk is itself a random variable, so we let \( n = \mathbb{E}_{w \in \mathcal{V}}[s] \). We assume that if the frequency is greater than this expectation, then the word is always produced by chance; less than this expectation, and it is always produced by the discourse or common discourse vectors. \( \alpha \) is the proportion of the vocabulary with \( p(w) \) above this threshold:

\[
\alpha = \frac{\sum_{w \in \mathcal{V}} \mathbbm{1}[p(w) > \mathbb{E}_{w \sim \mathcal{V}}[p(w|c_s^1, ..., c_s^n)]]}{|\mathcal{V}|}
\]

(11)

Since we can directly calculate \( Z_{c_s} \) and \( \alpha \), we can also directly calculate \( a = (1 - \alpha)/(\alpha Z_{c_s}) \).

**Common Discourse Vectors** We estimate the \( m \) common discourse vectors as the first \( m \) singular vectors from the singular value decomposition of
the weighted average vectors. \{\lambda_i\} are the weights on the common discourse vectors. In reality, these are unique to the word for which \(p(w|c_i)\) is being evaluated. However, we let \(\lambda_i\) be: \[
\lambda_i = \frac{\sigma_i^2}{\sum_j \sigma_j^2}
\]
where \(\sigma_i\) is the singular value for \(c_i\), \(\lambda_i\) can be interpreted as the proportion of variance explained by \(\{c_1^i, ..., c_m^i\}\) that is explained by \(c_i^i\). If removing the common discourse vectors is a form of denoising (Arora et al., 2017), increasing \(m\), in theory, should improve results. Because the variance explained by a singular vector falls with every additional vector that is included, the choice of \(m\) is thus a trade-off between variance explained and computational cost. When \(m = 1\), this is equivalent to the removal in Arora et al. (2017). We fix \(m = 5\), as we find empirically that singular vectors beyond that do not explain much more variance. To get \(c_s\), we subtract from \(\tilde{c}_s\) the weighted projection on each singular vector:
\[
c_s = \tilde{c}_s - \sum_{i=1}^m \lambda_i \text{proj}_{c_i} \tilde{c}_s
\]

We call this piecewise common component removal. Because our weighting scheme requires no hyperparameter tuning, it is completely unsupervised. For this reason, we call it unsupervised smoothed inverse frequency (uSIF). The full algorithm is given in Algorithm 1.

Note that while it is certainly possible to tune the hyperparameters in our model to achieve optimal results, it is not necessary to do so, which allows our method to be used when there is no labelled data. By contrast, in Arora et al.’s model (2017), hyperparameter tuning is a necessity.

Confound of Vector Length To understand why this model is not prone to the confound of word vector length, we reconsider the class of assignments for \(v_x, v_y, v_z\) in (5) and the resulting values for \(\tilde{c}_s\) and \(\tilde{c}_h\). Recall that in our example, sentence \(g\) comprises words \(x, y\) and sentence \(h\) comprises two instances of the word \(z\). Under our new weighting scheme, \(C\) in (5) is replaced with \(C' = \frac{\sigma}{p(x) + \frac{\sigma}{2}}\). Note that we use \(p(x)\) in \(C'\) because of the simplifying assumption that \(p(x) = p(y) = p(z)\). Assuming again that \(p(x) \approx 0\) and that piecewise common component removal has negligible effect, we can see how \(p(g|c_g)\) and \(p(h|c_h)\) change in our random walk model:
\[
p(g|c_g) \propto \prod_{w \in \{x,y\}} \left(1 - \frac{\arccos (\cos (c_w, v_w))}{\pi}\right)
\]
\[
p(h|c_h) \propto \left(1 - \frac{\arccos (\cos (c_h, v_z))}{\pi}\right)^2 = 1
\]
Because \(p(g|c_g)\) is ultimately based on the cosine similarities between the discourse vector and word vectors, it is a function of the parameter \(\sigma \in [0, 1]\) that controls the degree of similarity between \(v_x\) and \(v_y\). For example, for the worst-case assignments (4), \(p(g|c_g) \approx 9/16\). Conversely, when \(v_x = v_y = v_z\), we get \(p(g|c_g) = p(h|c_h) \approx 1\). Recall that in Arora et al.’s model (2017), \(\beta \geq 2\) was sufficient to ensure the counter-intuitive result of \(p(g|c_g) \geq p(h|c_h)\) (6), where \(\beta\) was a scalar that controlled the word vector length. In contrast, in our random walk model, the effect of \(\beta\) – and thus the confound of vector length – is entirely absent; only the similarity between the word vectors is influential.
4 Results and Discussion

4.1 Textual Similarity Tasks

We test our approach on the SemEval semantic textual similarity (STS) tasks (2012-2015) (Agirre et al., 2012, 2013, 2014, 2015), the SemEval 2014 Relatedness task (SICK’14) (Marelli et al., 2014), and the STS Benchmark dataset (Cer et al., 2017). In these tasks, the goal is to determine the semantic similarity between a given pair of sentences; the evaluation criterion is the Pearson correlation coefficient between the predicted and actual similarity scores. To predict the similarity score, we simply encode each sentence and take the cosine similarity of their vectors. The individual scores for STS tasks are in Table 4 in the Appendix and the average scores are in Table 1. The STS benchmark scores are in Table 2. We compare our results with those from several methods, which are categorized by Cer et al. (2017) as ‘unsupervised’, ‘weakly supervised’, or ‘supervised’.

4.2 Experimental Settings

For a fair comparison with Arora et al. (2017), we use the unigram probability distribution used by them, based on the enwiki dataset (Wikipedia, 3B words). Our preprocessing of the sentences is limited to tokenization. We try our method with three types of word vectors: GloVe vectors (Pennington et al., 2014), PARAGRAM-SL999 (PSL) vectors (Wieting et al., 2015), tuned on the SimLex999 dataset, and ParaNMT-50 vectors (Wieting and Gimpel, 2017a), tuned on 51M English-English sentence pairs translated from English-Czech sentence pairs. The value of $n$ in (11) is $\mathbb{E}_{s \in S}|s| \approx 11$ and was estimated using sentences from all corpora. The value of $a$ in (9) is then $1.2 \times 10^{-3}$. Our results are denoted as $X+UP$, where $X \in \{ \text{GloVe}, \text{PSL}, \text{ParaNMT}\}$, $U$ denotes uSIF-weighting, and $P$ denotes piecewise common component removal.

4.3 Results

Our model outperforms Arora et al.’s by up to 44.4% on individual tasks (see GloVe+UP vs. GloVe+WR for the STS’12 MSRpar task in Table 4) and by up to 15.5% on yearly averages (see GloVe+UP vs. GloVe+WR for STS’12 in Table 1). Our approach proves most useful in cases where Arora et al. (2017) underperform others, such as for STS’12, where our models – GloVe+UP and PSL+UP – outperform their equivalents in Arora et al.’s results by 15.5% and 10.6% respectively. On average, our approach outperforms Arora et al.’s by around 7.6%, but the improvement is highly variable. This may be because the hyperparameter values we derived may be closer to the optima for some corpora more than others or because our other improvements – normalization and piecewise common component removal – are more effective for certain datasets.

Our best model, ParaNMT+UP, is also competitive with the state-of-the-art model, ParaNMT Trigram-Word, an average of trigram and word embeddings tuned on the ParaNMT-dataset. ParaNMT+UP outperforms ParaNMT Trigram-Word on STS’12, STS’13, and STS’14; it is narrowly outperformed on STS’15 and the STS benchmark. ParaNMT Trigram-Word’s inclusion of trigram embeddings gives it an edge over our model for out-of-vocabulary words (Wieting and Gimpel, 2017a). It should be noted that ParaNMT+UP outperforms both ParaNMT Word Avg. and ParaNMT BiLSTM Avg., implying that our model composes words better than both simple averaging and BiLSTMs. Similarly, our model PSL+UP outperforms PP-XXL (Wieting et al., 2016b), despite the latter using the same word vectors and a learned projection instead.

Ablation Study

On average, our weighting scheme alone is responsible for a roughly 4.4%
improvement over Arora et al. The piecewise common component removal alone is responsible for a roughly 5.1% improvement, and the normalization alone is responsible for a roughly 6.7% improvement. This suggests that the benefits of our individual contributions have much overlap. The choice of tuned word vectors (e.g., ParaNMT over GloVe) can also improve results by up to 11.2%.

### 4.4 Supervised Tasks

We also test our approach on three supervised tasks: the SICK similarity task (SICK-R), the SICK entailment task (SICK-E), and the Stanford Sentiment Treebank (SST) binary classification task (Socher et al., 2013). To a large extent, performance on these tasks depends on the architecture that is trained with the sentence embeddings. We take the embeddings that perform best on the textual similarity tasks, ParaNMT+UP, and follow the setup in Wieting et al. (2016b). As seen in Table 3, both SIF-weighting with common component removal (Arora et al., 2017) and uSIF-weighting with piecewise common component removal (ours) perform slightly better than simple word averaging, but not as well as more sophisticated models. Past work has found that tuning the word embeddings in addition to the parameters of the model yields much better performance (Wieting et al., 2016b), as does increasing the size of the hidden layer in the classifier (Arora et al., 2017). The results here, however, suggest that regardless of such changes, our approach would not be any more effective than Arora et al.’s on these tasks. Still, our approach retains the advantage of being a completely unsupervised method that can be used when there is no labelled data.

<table>
<thead>
<tr>
<th>Unsupervised</th>
<th>GloVe+UP</th>
<th>Charagram (Wieting et al., 2016a)</th>
<th>Paragraph-Phrase (Wieting et al., 2016b)</th>
<th>PSIF+UP</th>
<th>S2vec (Pagliardini et al., 2017)</th>
<th>Doc2Vec DBOW (Le and Mikolov, 2014)</th>
</tr>
</thead>
<tbody>
<tr>
<td>Score</td>
<td>71.5</td>
<td>77.6</td>
<td>73.2</td>
<td>74.8</td>
<td>75.8</td>
<td>64.9</td>
</tr>
</tbody>
</table>

Table 2: Results (Pearson’s $r \times 100$) on the STS Benchmark dataset. The highest score is in bold. The scores of our approaches are underlined.

<table>
<thead>
<tr>
<th>Weakly Supervised</th>
<th>GloVe+WR (Arora et al., 2017)</th>
<th>GRAN (Wieting and Gimpel, 2017b)</th>
<th>ParaNMT+UP</th>
<th>ParaNMT-UP</th>
<th>S2vec (Wieting and Gimpel, 2017a)</th>
</tr>
</thead>
<tbody>
<tr>
<td>Score</td>
<td>72.0</td>
<td>76.4</td>
<td>79.2</td>
<td>79.5</td>
<td>79.9</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>Supervised</th>
<th>Constiency-Tree-LSTM (Tai et al., 2015)</th>
<th>CNN (HCTi) (Shao, 2017)</th>
</tr>
</thead>
<tbody>
<tr>
<td>Score</td>
<td>71.9</td>
<td>78.4</td>
</tr>
</tbody>
</table>

Table 3: Results on the SST, SICK-R, and SICK-E tasks. The best score for each task is bolded. † indicates our implementation.

### 5 Future Work

There are several possibilities for future work. For one, the values we derived for $Z$, $\alpha$, $\alpha$, and $\{\lambda\}$ are not necessarily optimal. While they are based on reasonable assumptions, there are likely sentence-specific and task-specific values that yield better results. Hyperparameter search is one way of finding these values, but that would require supervision. It may be possible, however, to theoretically derive more optimal values.

### 6 Conclusion

We first showed that word vector length has a confounding effect on the log-linear random walk model of generating text (Arora et al., 2017), the basis of a strong baseline method for sentence embeddings. We then proposed an angular distance-based random walk model where the probability of a sentence being generated is robust to distortion from word vector length. From this model, we derived a simple approach for creating sentence embeddings: normalize the word vectors, compute a weighted average, and then modify it using SVD. Unlike in Arora et al., our approach does not require hyperparameter tuning – it is completely unsupervised and can therefore be used when there is no labelled data. Our approach outperforms Arora et al.’s by up to 44.4% on textual similarity tasks and is even competitive with state-of-the-art methods. Because our simple approach is tough-to-beat, robust, and unsupervised, it is an ideal baseline for computing sentence embeddings.
Acknowledgments

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References


Evaluating Word Embeddings in Multi-label Classification Using Fine-grained Name Typing

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Abstract

Embedding models typically associate each word with a single real-valued vector, representing its different properties. Evaluation methods, therefore, need to analyze the accuracy and completeness of these properties in embeddings. This requires fine-grained analysis of embedding subspaces. Multi-label classification is an appropriate way to do so. We propose a new evaluation method for word embeddings based on multi-label classification given a word embedding. The task we use is fine-grained name typing: given a large corpus, find all types that a name can refer to based on the name embedding. Given the scale of entities in knowledge bases, we can build datasets for this task that are complementary to the current embedding evaluation datasets in: they are very large, contain fine-grained classes, and allow the direct evaluation of embeddings without confounding factors like sentence context.

1 Introduction

Distributed representation of words, aka word embedding, is an important element of many natural language processing applications. The quality of word embeddings is assessed using different methods. Baroni et al. (2014) evaluate word embeddings on different intrinsic tests: similarity, analogy, synonym detection, categorization and selectional preference. Different concept categorization datasets are introduced. These datasets are small (<500) (Baroni et al., 2014; Rubinstein et al., 2015) and therefore measure the goodness of embeddings by the quality of their clustering. Usually cosine is used as the similarity metric between embeddings, ignoring subspace similarities.

Extrinsic evaluations are also used, cf. Li and Jurafsky (2015). In these tasks, embeddings are used in context/sentence representations with composition involved.

In this paper, we propose a new evaluation method. In contrast to the prior work on intrinsic evaluation, our method is supervised, large-scale, fine-grained, automatically built, and evaluates embeddings in a classification setting where different subspaces of embeddings need to be analyzed. In contrast to the prior work on extrinsic evaluation, we evaluate embeddings in isolation, without confounding factors like sentence contexts or composition functions.

Our evaluation is based on an entity-oriented task in information extraction (IE). Different areas of IE try to predict relevant data about entities from text, either locally (i.e., at the context-level), or globally (i.e., at the corpus-level). For example, local (Zeng et al., 2014) and global (Riedel et al., 2013) in relation extraction, or local (Ling and Weld, 2012) and global (Yaghoobzadeh and Schütze, 2015) in entity typing. In most global tasks, each entity is indexed with an identifier (ID) that usually comes from knowledge bases such as...
Freebase. Exceptions are tasks in lexicon generation or population like entity set expansion (ESE) (Thelen and Riloff, 2002), which are global but without entity IDs. ESE usually starts from a few seed entities per set and completes the set using pattern-based methods.

Here, we address the task of fine-grained name typing (FNT), a global prediction task, operating on the surface names of entities. FNT and ESE share applications in name lexicon population. FNT is different from ESE because we assume to have sufficient training instances for each type to train supervised models.

The challenging goal of FNT is to find the types of all entities a name can refer to. For example, "Washington" might refer to several entities which in turn may belong to multiple types, see Figure 1. In this example, "Washington" refers to "Washington DC (city)" or "Washington (state)", or "George Washington (president)". Also, each entity can belong to several types, e.g., "George Washington" is a POLITICIAN, a PERSON and a SOLDIER, or "Washington (state)" is a STATE and a LOCATION.

Learning global representations for entities is very effective for global prediction tasks in IE (cf., Yaghoobzadeh and Schütze (2015)). For our task, FNT, we also learn a global representation for each name. By doing so, we see this task as a challenging evaluation for embedding models. We intend to use FNT to answer the following questions: (i) How well can embeddings represent distinctive information, i.e., different types or senses? (ii) Which properties are important for an embedding model to do well on this task?

We build a novel large-scale dataset of (name, types) from Freebase with millions of examples. The size of the dataset enables supervised approaches to work, an important requirement to be able to look at different subspaces of embeddings (Yaghoobzadeh and Schütze, 2016). Also, in FNT names are—in contrast to concept categorization datasets—multi-labeled, which requires to look at multiple subspaces of embeddings.

In summary, our contributions are (i) introducing a new evaluation method for word embeddings (ii) publishing a new dataset that is a good resource for evaluating word embeddings and is complementary to prior work: it is very large, contains more different classes than previous word categorization datasets, and allows the direct evaluation of embeddings without confounding factors like sentence context\(^1\).

2 Related Work

Embedding evaluation. Baroni et al. (2014) evaluate embeddings on different intrinsic tests: similarity, analogy, synonym detection, categorization and selectional preference. Schnabel et al. (2015) introduce tasks with more fine-grained datasets. The concept categorization datasets used for embedding evaluation are mostly small (<500) (Baroni et al., 2014) and therefore measure the goodness of embeddings by the quality of their clustering. In contrast, we test embeddings in a classification setting and different subspaces of embeddings are analyzed. Extrinsic evaluations are also used (Li and Jurafsky, 2015; Köhn, 2015; Lai et al., 2015). In most tasks, embeddings are used in context/sentence representations with composition involved. In this work, we evaluate embeddings in isolation, on their ability to represent multiple senses.

Related tasks and datasets. Our proposed task is fine-grained name typing (FNT). A related task is entity set expansion (ESE): given a set of a few seed entities of a particular class, find other entities (Thelen and Riloff, 2002; Gupta and Manning, 2014). We can formulate FNT as ESE, however, there is a difference in the training data assumption. For our task, we assume to have enough instances for each type available, and, therefore, to be able to use a supervised learning approach. In contrast, for ESE, mostly only 3-5 seeds are given as training seeds for a set, which makes an evaluation like ours impossible.

Named entity recognition (NER) consists of recognizing and classifying mentions of entities locally in a particular context (Finkel et al., 2005). Recently, there has been increased interest in fine-grained typing of mentions (Ling and Weld, 2012; Yogatama et al., 2015; Ren et al., 2016; Shimaoka et al., 2016). One way of solving our task is to collect every mention of a name, use NER to predict the context-dependent types of mentions, and then take all predictions as the global types of the name. However, our focus in this paper is on how embedding models perform and propose this task as a good evaluation method. We leave the comparison to an NER-based approach for future work.

Corpus-level fine-grained entity typing is the

\(^1\)Our dataset is available at: \url{https://github.com/yyaghoobzadeh/name_typing}
task of predicting all types of entities based on their mentions in a corpus (Yaghoobzadeh and Schütze, 2015; Yaghoobzadeh and Schütze, 2017; Yaghoobzadeh et al., 2018). This is similar to our task, FNT, but in FNT the goals is to find the corpus-level types of names. Corpus-level entity typing has also been used for embedding evaluation (Yaghoobzadeh and Schütze, 2016). However, they need an annotated corpus with entities. For FNT, however, pretrained word embeddings are sufficient for the evaluation.

Finally, there exists some previous work on FNT, e.g., Chesney et al. (2017). In contrast to us, they do not explicitly focus on the evaluation of embedding models, such that their dataset only contains a limited number of types. In contrast, we use 50 different types, making our dataset suitable for the type of evaluation intended.

3 Multi-label Classification of Word Embeddings

Word embeddings are global representations of word properties learned from the context distribution of words. Words are usually ambiguous and belong to multiple classes, e.g., multiple part-of-speech tags or multiple meanings. A good word embedding should represent all information about the word, including its multiple classes. Our evaluation methodology is based on this hypothesis and tries to test this through multi-label classification of word embeddings. Here, we focus on the semantic property of nouns and entity names. We try to find all categories or types of a noun given its embedding.

Multi-label classification of embedding has multiple advantages over current evaluation methods: (i) large datasets can be created without much human annotation; (ii) more fine-grained analysis of the results is possible through analyzing classification performance; (iii) it allows the direct evaluation of embeddings without confounding factors like sentence context.

4 Fine-grained Name Typing

We assume to have the following: a set of names \( N \), a set of types \( T \) and a membership function \( m : N \times T \rightarrow \{0, 1\} \) such that \( m(n, t) = 1 \) iff name \( n \) has type \( t \); and a large corpus \( C \). In this problem setting, we address the task of fine-grained name typing (FNT): we want to infer from the corpus for each pair of name \( n \) and type \( t \) whether \( m(n, t) = 1 \) holds.

For example, for the name “Hamilton”, we should find all of the following: location, organization, person, city, sports_team and soldier, since “Hamilton” can describe entities belonging to those types. Another example is “Falcon” which is used for animal, airplane, software, art. FNT sheds light on to which level these fine-grained types can be inferred from a corpus using embeddings.

4.1 Embedding-based Model

We aim to find \( P(t|n) \), i.e., the probability of name \( n \) having type \( t \). Given sufficient training instances for each type \( t \), we can formulate the problem as a multi-label classification task. As input, we use a representation for \( n \), learned from the corpus \( C \). Distributional representations have shown to capture various types of information about a word, especially their categories or types (Yaghoobzadeh and Schütze, 2015).

After learning an embedding for \( n \), we train two kinds of binary classifiers for each type \( t \) to estimate \( P(t|n) \): (i) linear: logistic regression (LR) with stochastic gradient decent; and (ii) non-linear: multi-layer perceptron (MLP) with one hidden layer and ReLU as the non-linearity. We use the Scikit-learn (Pedregosa et al., 2011) toolkit for training our classifiers.

5 Dataset

Using Freebase (Bollacker et al., 2008), we first retrieve the set of all entities \( E_n \) for each name \( n \).\(^2\) Then, we consider the types of all \( e \in E_n \) the types of \( n \). See Figure 1 for an example: all of the shown types belong to the name “Washington”.

Since some of the about 1,500 Freebase types have very few instances, we map them first to the FIGER (Ling and Weld, 2012) type-set, which contains 113 types. We then further restrict our set to the top 50 most frequent types. See Table 5 for the list of types.

In order to be able to evaluate each embedding on its own, we divide our dataset into single-word (891,241 names) and multi-word (8,907,715 names). In this work, the multi-word set is not used. We then set a frequency threshold of 100 in our lowercased Wikipedia corpus \(^3\) and select

\(^2\)What we call “names” here are either names or aliases in the Freebase terminology.

\(^3\)Our Wikipedia dump is from 2014.
We choose four different Embedding models.

**6 Experiments**

6.1 FNT for Embedding Evaluation

Embedding models. We choose four different embedding models for our comparisons: (i) SkipGram (henceforth SKIP) (skipgram bag-of-words model) (Mikolov et al., 2013), (ii) CBOW (continuous bag-of-words model) (Mikolov et al., 2013), (iii) Structured SkipGram (henceforth SSKIP) (Ling et al., 2015), (iv) CWindow (henceforth CWIN) (continuous window model) (Ling et al., 2015), and SSKIP and CWIN are order-aware, i.e., they take the order of the context tokens into account, while SKIP and CBOW are bag-of-words models.

Results and analysis. We report the results for all embedding models using LR and MLP in Table 3. We use the following evaluation measures, which are used in entity typing (Yaghoobzadeh and Schütze, 2015): (i) ACC (accuracy): percentage of test examples where all predictions are correct, (ii) Micro-F1: the global F1 computed over all the predictions.

Models in lines 1-5 in Table 3 are trained on the Wikipedia corpus. We set the min frequency in corpus to 100. Window size = 3; negative sampling with $n = 10$. Based on the results of LR, order-aware architectures are better than their bag-of-words counterparts, i.e., SSKIP > SKIP and CWIN > CBOW. Overall, SSKIP is the best using LR classification. In MLP results, however, CBOW works best on micro-F1 measure and SSKIP and SKIP are bests on accuracy. There is no significant difference between CBOW and CWIN, or SSKIP and SKIP, respectively. Overall, the nonlinear classifier (MLP) with one hidden layer outperforms the linear classifier (LR) substantially, emphasizing that the encoded information about different types is easier to extract with stronger models.

Analysis on the number of name types. As a separate analysis, we measure how the classification performance depends on the $N$ number of types of a name. To do so, we group test names based on their number of types. We keep the groups that have more than 100 members. Then, we plot the F1 results of CBOW and CWIN models trained using MLP classifier in Figure 2.

As it is shown, both models get their best results on names with $N = 2$. We suppose that the bad performance of $N = 1$ is related to the fact that one-type names have missing types in our dataset due to the incompleteness of Freebase. The worse F1 of $N >= 3$ compared to $N = 2$ is expected since bigger $N$ means that the models need to predict more types from the name embeddings. From $N = 4$, somewhat surprisingly the F1 increases as $N$ increases. This is perhaps related to the frequency of names in the corpus, and its relation to the number of names types: as $N$ increases, the frequency of words increases and the embedding has a better quality. However, this is only a hypothesis and more investigation is required. The other observation is in the trend of CBOW and CWIN results. CBOW is worse for $N <= 2$, but

<table>
<thead>
<tr>
<th></th>
<th>ACC</th>
<th>Micro-F1</th>
<th>ACC</th>
<th>Micro-F1</th>
</tr>
</thead>
<tbody>
<tr>
<td>CBOW</td>
<td>19.2</td>
<td>47.8</td>
<td>24.9</td>
<td>54.6</td>
</tr>
<tr>
<td>SKIP</td>
<td>22.6</td>
<td>49.3</td>
<td>25.2</td>
<td>53.5</td>
</tr>
<tr>
<td>CWIN</td>
<td>22.6</td>
<td>49.8</td>
<td>25.1</td>
<td>54.2</td>
</tr>
<tr>
<td>SSKIP</td>
<td>23.4</td>
<td>50.5</td>
<td>25.2</td>
<td>53.6</td>
</tr>
</tbody>
</table>

Table 3: Accuracy and micro-F1 results on FNT for different embedding models using two classifiers (LR and MLP). Best result in each column is bold.

Table 1: List of the 50 types in our FNT dataset.

<table>
<thead>
<tr>
<th>type</th>
<th>count</th>
</tr>
</thead>
<tbody>
<tr>
<td>train</td>
<td>dev</td>
</tr>
<tr>
<td>50,000</td>
<td>20,000</td>
</tr>
<tr>
<td>avg #types per name</td>
<td>3.77</td>
</tr>
</tbody>
</table>

Table 2: Some statistics (number of names; average number of types per name) for our name typing dataset.
it works clearly better for $N > 2$. This shows that the embedding models behave differently for different number of classes they belong to. This could also be related to the frequency of words. Analysis of the reasons would be interesting. We leave it for the future work.

7 Conclusion

We proposed multi-label classification of word embeddings using the task of fine-grained typing of entity names. The dataset we built is a resource that is complementary to prior work in embedding evaluation: it is very large, its examples are multi-labeled with very fine-grained classes, and it allows the direct evaluation of embeddings without the need for context. We analyzed the performance of different embedding models on this dataset, showing differences in their performance as well as some of their limits in representing types accurately and completely.

More analysis and evaluation is necessary though, but we believe by using this kind of dataset, we are able to do much more than what we could do before with the small manually built word similarity and categorization benchmarks.

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NOTICE OF RETRACTION

The authors, Ade Romadhony, Alfan Farizki Wicaksono, Ayu Purwarianti and Dwi Hendratmo Widiantoro, of the paper “A Dense Vector Representation for Open-Domain Relation Tuples”, have agreed to have their already printed paper to be retracted from any form of publication, and thus should never be cited.

Reason for retraction: this paper was accepted as an extended abstract paper that should not have appeared in the proceedings but was mistakenly included.

The authors have the consent of the publication chairs, Hongyuan Mei and Isabelle Augenstein of Representation Learning for NLP, to have the paper retracted from the published proceedings.

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July 27 2018

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Exploiting Common Characters in Chinese and Japanese to Learn Cross-lingual Word Embeddings via Matrix Factorization

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Abstract

Learning vector space representation of words (i.e., word embeddings) has recently attracted wide research interests, and has been extended to cross-lingual scenario. Currently most cross-lingual word embedding learning models are based on sentence alignment, which inevitably introduces much noise. In this paper, we show in Chinese and Japanese, the acquisition of semantic relation among words can benefit from the large number of common characters shared by both languages; inspired by this unique feature, we design a method named CJC targeting to generate cross-lingual context of words. We combine CJC with GloVe based on matrix factorization, and then propose an integrated model named CJ-Glo. Taking two sentence-aligned models and CJ-BOC (also exploits common characters but is based on CBOW) as baseline algorithms, we compare them with CJ-Glo on a series of NLP tasks including cross-lingual synonym, word analogy and sentence alignment. The result indicates CJ-Glo achieves the best performance among these methods, and is more stable in cross-lingual tasks; moreover, compared with CJ-BOC, CJ-Glo is less sensitive to the alteration of parameters.

1 Introduction

Word representation is critical to various NLP tasks, and the traditional one-hot representation, despite its simplicity, suffers from at least two aspects: the vector dimensionality increases with vocabulary size, leading to “curse of dimensionality”; more importantly, it fails to capture the semantic relation among words.

Due to the defects of one-hot representation, the majority of research interests now have switched to distributed word representation (also known as “word embedding”), which represents word as a real-valued vector. Represented as vectors, the semantics of words are better reflected, as the relatedness of words can be quantified using vector arithmetic.

To efficiently train word embeddings, a range of models have been proposed, most of them targeting to train monolingual word embedding. Though word embedding is often discussed under monolingual scenario, cross-lingual embedding can serve as a useful tool in several NLP tasks including machine translation (Wu et al., 2016), word sense disambiguation (Chen et al., 2014), and so on. This is because cross-lingual word embeddings map words from two languages into one vector space, thereby making it possible to measure the semantic relation among words from different languages. However, compared with the bulk of works studying monolingual word embedding, cross-lingual word embedding is still at its initial stage, with no learning model being widely accepted.

In this paper, we present a method named CJC (Chinese-Japanese Common Character)
aiming to extract cross-lingual context of words from sentence aligned Chinese-Japanese corpus. Given the large amount of common characters shared by both languages and the rich semantic connections thereof, we exploit them to acquire potential word level alignment. The acquired cross-lingual contexts can be flexibly integrated with various models; in this paper, CJC is mainly integrated with a matrix factorization model called GloVe (Pennington et al., 2014), and the integrated model is thus called CJ-Glo.

To evaluate the performance of CJ-Glo, we take 2 sentence aligned models respectively based on CBOV(Mikolov et al., 2013a) and GloVe, and CJ-BOC model (based on Common Character + CBOV) (Wang et al., 2016) as contrast, and compare the trained word embeddings of these methods using three typical NLP tasks, including cross-lingual synonym, word analogy and sentence alignment. According to the experiment results, the acquired word embeddings by using CJ-Glo have better quality than those of the other models; moreover, CJ-Glo performs more stably than its competitors, and is less sensitive to parameter alteration.

2 Related work

Word embedding was initiated by Hinton (1986), which essentially encodes word using a real-valued vector. With word embeddings, the intrinsic relatedness among words can be explicitly measured as the distances or angles between word pairs. This favorable feature of word embedding soon led to its popularity in industry and academia in past decades. Specifically, word embedding has found its applications in machine translation (Wu et al., 2016; Lample et al., 2017), word sense disambiguation (Chen et al., 2014; Guo et al., 2014), information retrieval (Vulić and Moens, 2015) and so on.

To efficiently acquire high-quality word embeddings, vast research efforts have therefore emerged. A representative framework to learn word embeddings is Neural Network Language Model (NNLM) proposed by Bengio et al. (2003), which adopts back-propagation when training word embeddings and parameters for the model. Another typical approach is matrix factorization, whose basic idea is to approximate original matrices with low-rank matrices by leveraging statistic information. For example, GloVe (Pennington et al., 2014) explicitly factorizes the co-occurrence matrix, training only non-zero elements instead of an entire spare matrix.

Traditionally, word embedding was studied under monolingual setting, and then naturally extended to bilingual scenario. Compared with monolingual word embeddings, bilingual word embedding reveals the internal relation among words of different languages; and such capability makes bilingual word embeddings a powerful tool to assist machine translation, or even serves as a substitute for word mapping matrix and dictionary in previous machine translation methods. A range of works have been proposed to learn bilingual word embeddings, such as (Mikolov et al., 2013b), which attempts to map separately trained word embeddings into one vector space, and acquire bilingual word embeddings. BilBOWA is a model proposed in (Gouws et al., 2015), whose most notable merit is the whole training process does not require word alignment or dictionary. word alignment or dictionary. (Shi et al., 2015) is another work that utilizes matrix factorization in word embeddings learning. Ruder et al. (2017) provides a detailed survey, which enumerates the input format and basic principles of various bilingual word embedding learning methods.

When it comes to non-alphabet-based language like Chinese and Japanese, an essential difference from alphabet-based languages is that each character in a word contains abundant information, and makes sense itself. In addition to this, an underlying correlation between Chinese and Japanese is the large portion of shared characters in both languages; with the help of these characters, Chu et al. (2014) extracted texts from Wikipedia web pages of Chinese and Japanese version, based on which they then constructed a Chinese-Japanese parallel corpus. A natural conjecture about the common characters is the semantic similarity or even equivalence among them. In light of this, we proposed CJ-BOC model in our previous work (Wang et al., 2016) to learn Chinese-Japanese bilingual word embed-
dings, which outperforms sentence-alignment approaches in terms of embedding quality. To our knowledge, our previous work is the first attempt to learn Chinese-Japanese word embeddings using common Chinese characters.

3 Chinese-Japanese Common Character

Historically, Chinese character has spread to a group of countries in East Asia as a major carrier of Chinese culture, thereby influencing the writing systems in these countries. Traditional Chinese, Simplified Chinese and Japanese Kanji are now being used, all developing from Traditional Chinese; and given the same root of them, these three writing systems actually share a large portion of common characters: for a certain character in one of them, we can find its counterparts in the other two, with minor variation or even of the same shape. Chu et al. (2012) proposed a Chinese character table comparing traditional Chinese, simplified Chinese and Japanese. As summarized in Table 1, the glyphs of such common characters can be 1) the same in all these three writing systems; 2) consistent in two of them; 3) different in all these three.

And with regard to their semantics, simplified and traditional Chinese are only two written forms of the same language, and therefore common characters within them are semantically equivalent. For Japanese Kanji, most characters are semantically equivalent or relevant to their counterparts in Chinese.

We in our previous work (Wang et al., 2016) quantified such semantic relatedness from the view of information theory using mutual information (MI) and conditional mutual information (CMI). By repeating the experiments in this paper, we acquired the results in Table 2. All these 5 characters have multiple meanings in both Chinese and Japanese, and their respective meanings differ to some extent in both languages. Normally CMI should be larger than MI, which indicates that in a translation-sentence pair, if 2 words from each sentence share a common character, they are likely to form a translation word pair. The results of shown in Table 2 are no exception, providing theoretical root for our model which will be proposed in section 4.

4 Model

4.1 Context of Word and CJC Method

Before delving into the learning models, we should first clarify the concept of context. In natural language processing, a widely adopted semantic representation model is Bag-of-Words (Zhang et al., 2010). The fundamental assumption of this model is: within a given sentence or paragraph, the target word is prone to have the most intimate semantic relation with its closest context words. Formally define a sentence \( S \) with \( l \) words as an ordered sequence: \( S = \langle w_0, w_1, ..., w_l \rangle \), and context function \( Ctx(\cdot) \) is often formulated as:

\[
Ctx(w, S) = \{w_k | i - K \leq k \leq i + K\}. \quad (1)
\]

In cross-lingual scenario, besides two monolingual corpora of both languages, a parallel corpus is often required in most models, which is aligned in either word-level (Guo et al., 2016) or sentence-level. Some recent works attempted to learn embeddings without using parallel corpus, such as (Artetxe et al., 2017).

Now try to consider bilingual context of a given target word in aligned parallel corpus. Let \( \langle S_{zh}, S_{ja} \rangle \) be a sentence pair, then define:

\[
Ctx(w_{zh,i}, S_{zh}) \cup Ctx(w_{zh,i}, S_{ja}). \quad (2)
\]

As formulated above, the context of a target word is the union of its contexts in both sentences. Therefore in word-aligned parallel corpus, let \( \langle w_{zh,i}, w_{ja,j} \rangle \) be a pair of aligned words, and the cross-lingual context \( Ctx_w(w_{zh,i}, S_{ja}) \) is equal to \( Ctx_w(w_{ja,j}, S_{ja}) \), since contexts in both languages are taken into account in this definition. In sentence-aligned parallel corpus, the cross-lingual context \( Ctx_s(w_{zh,i}, S_{ja}) \) is defined as the set of all the words in the respective sentence.

In real applications, sentence alignment data are usually easier to acquire. For example, Chu et al. (2014) proposed an approach to align Chinese-Japanese cross-lingual wiki corpus, using the common characters between both languages.

According to the analysis in Section 3, given an aligned Chinese-Japanese sentence pair, word alignment can be performed upon word pairs that share common characters. Based on
As the name implies, GloVe utilizes the global information of the corpus for vector training. GloVe and CBOW, two words in a pair share their respective context during training.

There are two corresponding word pairs denoted by common characters: “The weather is nice, let’s take a walk”. Two words in a pair share their respective context during training.

This conclusion, using common characters, we can now give a definition for context similar to context in sentence-align corpus.

Define a character matching function $CC(\cdot)$ that generates a set of words in which each word has at least one common character with target word $w_{zh,i}$:

$$CC(w_{zh,i}, S_{ja}) = \{w_{ja} | w_{ja} \in S_{ja}, c \in w_{zh,i}, c \in w_{ja}\}.$$  (3)

Thus parallel context $Ctx_c(w_{zh,i}, S_{ja})$ can be acquired via common character matching:

$$Ctx_c(w_{zh,i}, S_{ja}) = \{w | w \in Ctx(w_{ja}, S_{ja}), w_{ja} \in CC(w_{zh,i}, S_{ja})\}.$$  (4)

Hence, when multiple words in the corresponding sentence have common characters with the target word, all of them will be included in $Ctx_c(w_{zh,i}, S_{ja})$. However, such case rarely occurs during our experiments.

For example, “天気/不错/一起/去/散步/吧” and “天气/が/良好/から/散步/しま/しょう” are a parallel sentence-pair, meaning “The weather is nice, let’s take a walk”.

Table 2: Corresponding examples and percentages(%) of common characters in Simplified Chinese (SC), Traditional Chinese (TC), and Japanese Kanji (KJ).

<table>
<thead>
<tr>
<th>Type</th>
<th>Example of Characters with Unicode</th>
<th>Percentage</th>
</tr>
</thead>
<tbody>
<tr>
<td>1 - AAA</td>
<td>人 (U+4EBA) 人 (U+4EBA) 人 (U+4EBA)</td>
<td>People 56.55</td>
</tr>
<tr>
<td>2 - AAB</td>
<td>窗 (U+7A97) 窗 (U+7A97) 窗 (U+7A93)</td>
<td>Window 4.63</td>
</tr>
<tr>
<td>3 - ABA</td>
<td>国 (U+56FD) 国 (U+570B) 国 (U+56FD)</td>
<td>Country 3.45</td>
</tr>
<tr>
<td>4 - ABB</td>
<td>習 (U+4E60) 習 (U+7FD2) 習 (U+7FD2)</td>
<td>Study 29.17</td>
</tr>
<tr>
<td>5 - ABC</td>
<td>図 (U+56FE) 図 (U+5716) 図 (U+56F3)</td>
<td>Picture 6.19</td>
</tr>
</tbody>
</table>

Table 2: Estimated MI and CMI of 5 Common Characters.

<table>
<thead>
<tr>
<th></th>
<th>MI</th>
<th>CMI</th>
</tr>
</thead>
<tbody>
<tr>
<td>天</td>
<td>0.3369</td>
<td>30.3057</td>
</tr>
<tr>
<td>地</td>
<td>0.5804</td>
<td>87.4515</td>
</tr>
<tr>
<td>人</td>
<td>0.8942</td>
<td>151.0069</td>
</tr>
<tr>
<td>中</td>
<td>0.7337</td>
<td>138.9676</td>
</tr>
<tr>
<td>学</td>
<td>0.4173</td>
<td>119.8921</td>
</tr>
</tbody>
</table>

We name this method as CJC (Chinese-Japanese Common Character) which uses $CC(\cdot)$ to determine context. Different from our previous work (Wang et al., 2016) which exploited common characters to facilitate only CBOW, this CJC method is more of a generalized scheme that can be integrated with various models including CBOW, Skip-Gram, GloVe etc.

4.2 CBOW-like Models

CBOW was a model proposed by Mikolov et al. in (Mikolov et al., 2013a), whose optimization goal is maximizing a probabilistic language model. In cross-lingual especially Chinese-Japanese scenario, the objective function for training $w_{zh,i}$ is:

$$L(S_{zh}) = \frac{1}{N} \sum_{i=1}^{N} \left\{ P_{zh,i,zh} \right\} + \lambda \cdot P_{zh,i,ja,c} + \mu \cdot P_{zh,i,ja,s},$$  (5)

where $P_{zh,i,zh}$, $P_{zh,i,ja,c}$, and $P_{zh,i,ja,s}$ are softmax function of the target word $w_{zh,i}$ to its corresponding monolingual context, sentence aligned cross-lingual context, and CJC context. Both $\lambda$ and $\mu$ here are parameters of the model. If $\lambda = 0$, this is a trial sentence aligned CBOW model, otherwise it is a CJC+CBOW model; the CJC-BOC model in our previous work (Wang et al., 2016) used similar approach, and would be used as a baseline in our experiments.

4.3 GloVe-like Models

4.3.1 GloVe

GloVe model was originally proposed by Pennington et al. (2014). As the name implies, GloVe utilizes the global information of the corpus for vector training. GloVe and CBOW,
as commonly adopted learning models, however differ a lot in terms of mathematical models, as they are respectively based on matrix factorization and neural network. The process of GloVe is as follows:

First, construct a word-word cooccurrence matrix $M = (m_{ij})_{n \times n}$, where $n$ is the size of the corpus, and $m_{ij}$ represents the number of occurrence of $w_j$ in the context of $w_i$ in all the sentences $S$.

The learning problem of GloVe can then be transformed into the optimization of function $F(\cdot)$, such that for any word embeddings $x_i$, $x_j$ and probe word embedding $\tilde{x}_k$, the objective function is defined below:

$$L = \sum_{i,j=1}^{n} f(m_{ij})(x_i^T \tilde{x}_j + b_i + b_j - \log m_{ij})^2,$$  

(6)

$$f(m) = \begin{cases} \left(\frac{m}{m_{\text{max}}}\right)^\alpha & \text{if } m < m_{\text{max}} \\ 1 & \text{otherwise.} \end{cases}$$  

(7)

In this function, both $b_i$ and $b_j$ are bias, and $f$ is a weighing function aiming to mitigate the impact of dataset size on training results. In GloVe, $m_{\text{max}}$ is set to 100 and $\alpha$ to $\frac{3}{4}$.

4.3.2 Cross-lingual GloVe and CJ-Glo

To fit GloVe in cross-lingual scenario, one should first expand the word-word cooccurrence matrix. Suppose two languages respectively contain $n$ and $t$ words, the new matrix would have a size of If $w_i$ and $w_j$ belong to the same language, $m_{ij}$ can be computed using exactly the same way as in GloVe; otherwise, suppose $(S_{zh}, S_{ja})$ is a pair of parallel sentences, $w_i \in S_{zh}$, $w_j \in S_{ja}$, and we have:

$$m_{ij} = \sum_{(S_{zh}, S_{ja})} (\lambda \cdot C_{ij,c} + \mu \cdot C_{ij,s}),$$  

(8)

$$C_{ij,c} = \text{Cnt}(w_j, Ctx_c(w_i, S_{ja})), \quad C_{ij,s} = \text{Cnt}(w_j, Ctx_s(w_i, S_{ja})).$$  

(9)

$\text{Cnt}(\cdot)$ counts the frequency of $w_j$ in certain context of $w_i$, either sentence aligned context or CJC context. Once the cross-lingual word-word cooccurrence matrix is obtained, the following optimization unfolds similarly with the monolingual GloVe model, using the objective function (6) and weighting function (7) to train.

Similar to Cross-lingual CBOW model, if the CJC learning rate $\lambda = 0$ in equation (8), this is a sentence aligned cross-lingual GloVe model. Otherwise, it is a CJC-enhanced model, and is thus called CJ-Glo.

Figure 1 demonstrates the operational principle of CJ-Glo: the square in this figure is a cross-lingual word co-occurrence matrix, in which the green square is a Chinese monolingual co-occurrence sub-matrix, and the orange square is for Japanese. The blue sections are cross-lingual sub-matrices and elements in them are calculated using equation (8). When two parallel sentences each contain a word sharing common characters, each word would be taken as a co-occurrence in the context of the other. Every point crossed by dotted lines and dotted rectangles represents an element to increment when processing the sentence pair.

5 Experiments and Analysis

5.1 Evaluation Methods

To evaluate the quality of cross-lingual word embeddings obtained from various models, we conducted three groups of experiments: 1) the straightforward cross-lingual synonym comparison; 2) cross-lingual word analogy; 3) sentence alignment.

Cross-lingual synonym comparison.

In monolingual scenario, the word embeddings of a pair of synonyms should have a high cosine similarity. This property is also applicable in cross-lingual word embeddings, i.e., the cosine similarity between a word embedding and its translated counterpart should also
be high. In real applications, the correspondence between words in source language and words in target language can be one-to-one, one-to-many, or vice versa. To effectively eliminate ambiguity, we picked 200 one-to-one corresponding word pairs \( \langle w_{zh}, w_{ja} \rangle \) at random, then for each word pair, calculated the cosine similarity between \( w_{zh} \) and \( w_{ja} \), denoted as \( d \), and computed the rank of \( d \) among the cosine similarities from \( w_{zh} \) to every Japanese word in corpus \( V_{ja} \). Use the rank to calculate its relative rate among all words:

\[
rate = (1 - \frac{rank - 1}{total\_word\_num}) \times 100\%. \quad (10)
\]

Conducted the same operation for \( w_{ja} \) and all words in corpus \( V_{zh} \). Calculate the average rate for all the 200 word pairs, and acquire the average rate of \( w_{zh} \rightarrow w_{ja} \) and \( w_{ja} \rightarrow w_{zh} \) respectively. Ambiguity is eliminated in all these word pairs, so a large rate is therefore favored.

**Cross-lingual word analogy.**

Word analogy is probably the most widely adopted task to evaluate the performance of word embeddings, because it depicts the connection between trained vector space and word semantics. Both CBOW(Mikolov et al., 2013a) and GloVe(Pennington et al., 2014) used a dataset with 19,544 queries for evaluation.

Given several related words from different languages, cross-lingual analogical reasoning works as follows: \( y = v(\text{はは}) - v(\text{ちち}) + v(\text{男男}) \), we hope that the relatedness between Japanese words “はは (mother)” and “ちち (father)” could help us find the Chinese word “女女 (girl)” and Japanese “女の子 (girl)” through Chinese word “男男 (boy)”.

More formally, the cross-lingual analogy task was undertaken as follows:
1. Input a quadruple of word embeddings \( \langle w_1: w_2: w_3: w_4 \rangle \), where each word could be either Chinese or Japanese;
2. Compute the target vector \( u = w_2 - w_1 + w_3 \), acquire the corresponding rank and rate as in cross-lingual synonym comparison for \( u \rightarrow w_4 \);
3. Based on the ratio of Chinese word count to Japanese word count in the quadruple \( \langle w_1: w_2: w_3: w_4 \rangle \), the word analogy task is divided into 5 subtasks, whose ratio are \((0:4), (1:3), (2:2), (3:1)\) and \((4:0)\), and their respective query amount is 420, 1680, 2520, 1680, and 420 in our experiment;
4. Calculate the average rate on every subtask.

Also, the average rate here is expected to be as large as possible.

**Sentence alignment.**

The above experiments respectively evaluated the direct similarity and cross-lingual feature of word embeddings. And now we consider a more complicated task: sentence alignment. In the dataset from (Chu et al., 2014), other than training data, a manual test dataset was also attached, which are 198 sentence pairs. Using this dataset, we conduct this experiment as follows:
1. For a Chinese sentence \( S_{zh,i} \), calculate its average vector \( U_{zh,i} \) and all \( U_{ja} \) of all sentences \( S_{ja,i} \), and compute the cosine similarity.
2. Sort all the cosine similarities in step 2, and acquire the rank of the average vector \( U_{ja,i} \) of \( S_{ja,i} \) (the parallel sentence of \( S_{zh,i} \)).
3. Transform rank into rate using formula 10, where total number is 198.
4. Compute the average rate \( S_{zh} \rightarrow S_{ja} \);
5. Follow the same steps above to generate \( S_{ja} \rightarrow S_{zh} \).

Compared with the previous experiments, which evaluate only the relation between individual word embeddings, sentence alignment is a comprehensive task using word embedding, and is a critical indicator for the overall quality of the trained word embeddings.

**5.2 Dataset and Training Details**

As mentioned previously, (Chu et al., 2014) generated a parallel corpus including Chinese-Japanese sentence pairs from Wikipedia; train.ja and train.zh in this dataset were used throughout our empirical study, both containing 126,811 lines of text. Concretely, every single line in these two files is a complete sentence, which is parallel to its counterpart in the other file. As the preprocessing for datasets, both files were segmented using MeCab\(^1\) and Jieba\(^2\) for Japanese and Chinese, respectively. During the preprocessing, we assured the segmentation on Chinese and

\(^2\)https://github.com/fxsjy/jieba, commit number: cb0dc2973d2fahaaf67a0245a14206d8be70db515.
Table 3: Parameters of CJC and sentence learning rates in each models.

<table>
<thead>
<tr>
<th>Model</th>
<th>$\lambda$</th>
<th>$\mu$</th>
</tr>
</thead>
<tbody>
<tr>
<td>SenBow</td>
<td>0</td>
<td>0.2</td>
</tr>
<tr>
<td>CJ-BOC</td>
<td>0.4</td>
<td>0.2</td>
</tr>
<tr>
<td>SenGlo</td>
<td>0</td>
<td>0.2</td>
</tr>
<tr>
<td>CJ-Glo</td>
<td>0.4</td>
<td>0.2</td>
</tr>
</tbody>
</table>

Table 4: Cross-lingual synonym comparison results on 200 one-to-one word pairs, the average rates(%) of each models.

<table>
<thead>
<tr>
<th>Model</th>
<th>$w_{zh} \rightarrow w_{ja}$</th>
<th>$w_{ja} \rightarrow w_{zh}$</th>
</tr>
</thead>
<tbody>
<tr>
<td>SenBow</td>
<td>83.97</td>
<td>83.76</td>
</tr>
<tr>
<td>CJ-BOC</td>
<td>96.75</td>
<td>97.61</td>
</tr>
<tr>
<td>SenGlo</td>
<td>91.17</td>
<td>90.05</td>
</tr>
<tr>
<td>CJ-Glo</td>
<td><strong>97.97</strong></td>
<td><strong>98.80</strong></td>
</tr>
</tbody>
</table>

Japanese were approximately grained, by tuning parameters.

Four models in total are put into comparison in our experiment:

1. *SenBow* model is the bilingual CBOW model applying sentence-aligned method;
2. *CJ-BOC* model from (Wang et al., 2016), considered as a CJC+CBOW model;
3. *SenGlo* model applies sentence-aligned method to GloVe model;
4. *CJ-Glo* model is our CJC method enhanced GloVe model.

The parameters of CJC learning rate $\lambda$ and sentence learning rate $\mu$ are showed in Table 3. Both SenGlo and CJ-Glo have a $m_{max}$ of 100, and an $\alpha$ of $\frac{3}{4}$. The thread count is 16 in the implementations of all these four models, the output vector dimensionality is 100, and the training process is iterated 15 times. We set the parameters to the above values, since these models achieved the optimal performances under such settings in our evaluation. All models are implemented using C language, and the code can be found on GitHub\(^3\).

Figure 2 summarizes the results of the cross-lingual word analogy task, whose X-axis represents the ratio of Chinese word count to Japanese word count. In the figure, the leftmost point represents the result of pure Japanese word analogy, and the rightmost is the pure Chinese word analogy. We can see that all 4 models achieve fair performances in pure Chinese/Japanese word analogy. However, when it comes to the cross-lingual word analogy, CJ- models outperform Sen- models, and GloVe-like models generally beat CBOW-like ones. Another noticeable fact is that CJ-Glo performs approximately good under all 5 ratios, showing basically no difference between cross-lingual and monolingual word analogy.

We display the sentence alignment results in Table 5. Similarly, we still find CJ- models outperform Sen-, and GloVe-like models beat CBOW-like ones. Again, CJ-Glo has the best performance.

According to the above experiments, we can see compared with typical sentence-aligned methods, Common Character enhanced mod-

5.3 Results

The result of cross-lingual synonym comparison is shown in Table 4, from which we can see the integration of Common Character leads to obvious performance improvement for both CBOW-like and GloVe-like models, compared with sentence-aligned models, and CJ-Glo achieve the best result.

Figure 2: Cross-lingual word analogy experiment result. X-axis is the number ratio of Chinese words and Japanese words in the analogy query ($w_1 : w_2 :: w_3 : w_4$).

\(^3\)https://github.com/jileiwang/CJC, commit number: a10592d200bc15f7b33d81a8f895e7de9ef8676d.
Table 5: Sentence alignment results on 198 parallel sentence pairs, the average rates(%) of each models.

<table>
<thead>
<tr>
<th>Model</th>
<th>$S_{zh} \rightarrow S_{ja}$</th>
<th>$S_{ja} \rightarrow S_{zh}$</th>
</tr>
</thead>
<tbody>
<tr>
<td>SenBow</td>
<td>79.14</td>
<td>74.63</td>
</tr>
<tr>
<td>CJ-BOC</td>
<td>86.39</td>
<td>83.14</td>
</tr>
<tr>
<td>SenGlo</td>
<td>90.33</td>
<td>84.90</td>
</tr>
<tr>
<td>CJ-Glo</td>
<td><strong>91.57</strong></td>
<td><strong>86.00</strong></td>
</tr>
</tbody>
</table>

models are superior in learning Chinese-Japanese cross-lingual word embeddings, as it achieves obvious performance boost in various tasks. Moreover, CJ-Glo performs better than CJ-BOC, and is non-sensitive in cross-lingual tasks.

5.4 Model Analysis: CJC Learning Rate

CJC learning rate here refers to the multiplying factor of CJC context $C(t_c(\cdot))$, which is $\lambda$ in CJ-BOC and CJ-Glo. It worths discussion that how would CJC learning rate affects the performance of our proposed models. To explore this issue, we conduct a simple experiment: fixing the other parameters as set in section 5.2, we only change CJC learning rate, and apply the acquired word embeddings to synonym $w_{zh} \rightarrow w_{ja}$ tasks. The results are displayed in Figure 3, in which we can find as $\lambda$ increases in CJ-BOC, the accuracy declines after an increase, showing a obvious local optimal. While in CJ-Glo, the accuracy keeps improving with the increase of $\lambda$. Note that both parameters should be less than 1, because otherwise the impact of cross-lingual context would dominate the learning process, obviously resulting in overfit. CJ-Glo is more stable during the change of CJC learning rate, this interesting difference between both models is related to the their underlying learning mechanisms.

6 Conclusion and Future Work

In this paper, we quantified the semantic connection among common characters shared by Chinese and Japanese, and utilized it as the theoretical root to propose our cross-lingual context extracting method CJC. CJC makes use of common characters of both languages to assist the acquisition of parallel contexts. The effectiveness of CJC enhanced matrix factorization model CJ-Glo was verified via a series of tasks including cross-lingual synonym, word analogy and sentence alignment. As the experiment result shows, models like CBOW and GloVe achieved notable performance gain after integrated with CJC. Furthermore, CJ-Glo performed the best among all evaluated state-of-the-art methods, and showed its stability on cross-lingual tasks and non-sensitiveness of training parameter changing.

Below are several directions we may work on in the future: 1) The idea of training character and word embeddings jointly (Chen et al., 2015) is applicable to Chinese-Japanese word embedding training. Meanwhile, we can also align common characters and train cross-lingual character embeddings to further improve the quality of trained word embeddings. 2) A recent work (Lai et al., 2016) indicates that the performances of a model may vary given different tasks. Therefore, we shall study the performance fluctuation of CJ-Glo with more tasks including machine translation.

Acknowledgments

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Reference


Ivan Vulić and Marie-Francine Moens. 2015. Monolingual and cross-lingual information retrieval models based on (bilingual) word embeddings. In Proceedings of the 38th International ACM SIGIR Conference on Research and Development in Information Retrieval, pages 363–372. ACM.


Abstract

Semantic networks and semantic spaces have been two prominent approaches to represent lexical semantics. While a unified account of the lexical meaning relies on one being able to convert between these representations, in both directions, the conversion direction from semantic networks into semantic spaces started to attract more attention recently. In this paper we present a methodology for this conversion and assess it with a case study. When it is applied over WordNet, the performance of the resulting embeddings in a mainstream semantic similarity task is very good, substantially superior to the performance of word embeddings based on very large collections of texts like word2vec.

1 Introduction

The study of lexical semantics has been at the core of the research on language science and technology as the meaning of linguistic forms results from the meaning of their lexical units and from the way these are combined (Pelletier, 2016). How to represent lexical semantics has thus been a central topic of inquiry. Three broad families of approaches have emerged in this respect, namely those advocating that lexical semantics is represented as a semantic network (Quillian, 1966), a feature-based model (Minsky, 1975; Bobrow and Norman, 1975), or a semantic space (Harris, 1954; Osgood et al., 1957).

In terms of data structures, under a semantic network approach, the meaning of a lexical unit is represented as a node in a graph whose edges between nodes encode different types of semantic relations holding among the units (e.g. hypernymy, meronymy, etc.). In a feature-based model, the semantics of a lexicon is represented by a hash table where a key is the lexical unit of interest and the respective value is a set of other units denoting typical characteristics of the denotation of the unit in the key (e.g. role, usage or shape, etc.). Under a semantic space perspective, in turn, the meaning of a lexical unit is represented by a vector in a high-dimensional space, where each component is based on some frequency level of co-occurrence with the other units in contexts of language usage.

The motivation for these three families of lexical representation is to be found in their different suitability and success in explaining a wide range of empirical phenomena, in terms of how these are manifest in ordinary language usage and how they are elicited in laboratory experimentation. These phenomena are related to the acquisition, storage and retrieval of lexical knowledge (e.g. the spread activation effect (Meyer and Schvaneveldt, 1971), the fan effect (Anderson, 1974), among many others) and to how this knowledge interacts with other cognitive faculties or tasks, including categorization (Estes, 1994), reasoning (Rips, 1975), problem solving (Holyoak and Koh, 1987), learning (Ross, 1984), etc.

In the scope of the formal and computational modeling of lexical semantics, these approaches have inspired a number of initiatives to build repositories of lexical knowledge. Popular examples of such repositories are, for semantic networks, WordNet (Fellbaum, 1998), for feature-based models, Small World of Words (De Deyne et al., 2013), and for the semantic space, word2vec (Mikolov et al., 2013a), among many others. Interestingly, to achieve the highest quality, repositories of different types typically resort to different empirical sources of data. For instance, WordNet is constructed on the basis of systematic lexical intuitions handled by human experts; the informa-
tion encoded in Small World of Words is evoked from laypersons; and word2vec is built on the basis of the co-occurrence frequency of lexical units in a collection of documents.

Even when motivated in the first place by psycholinguistic research goals, these repositories of lexical knowledge have been extraordinarily important for language technology. They have been instrumental for major advances in language processing tasks and applications such as word sense disambiguation, part-of-speech tagging, named entity recognition, sentiment analysis (e.g. (Li and Jurafsky, 2015)), parsing (e.g. (Socher et al., 2013)), textual entailment (e.g. (Baroni et al., 2012)), discourse analysis (e.g. (Ji and Eisenstein, 2014)), among many others.1

The proliferation of different types of representation for the same object of research is common in science, and searching for a unified rendering of a given research domain has been a major goal in many disciplines. To a large extent, such search focuses on finding ways of converting from one type of representation into another. Once this is made possible, it brings not only the theoretical satisfaction of getting a better unified insight into the research object, but also important instrumental rewards of reapplying results, resources and tools that had been obtained under one representation to the other representations, thus opening the potential for further research advances.

This is the case also in what concerns the research on lexical semantics. Establishing whether and how any given lexical representation can be converted into another representation is important for a more unified account of it. On the language science side, this will likely enhance the plausibility of our empirical modeling about how the mind-brain handles lexical meaning. On the language technology side, in turn, this will permit to reuse resources and find new ways to combine different sources of lexical information for better application results.

In the present paper, we seek to contribute towards a unified account of lexical semantics. We report on the methodology we used to convert from a semantic network based representation of lexical meaning into a semantic space based one, and on the successful evaluation results obtained when applying that methodology. We resorted to Princeton WordNet version 3 as a repository of the lexical semantics of the English language, represented as a semantic graph, and converted a subgraph of it with half of its concepts into wnet2vec, a collection of vectors in a high-dimension space. These WordNet embeddings were evaluated under the same conditions that semantic space based repositories like word2vec are, namely under the processing task of determining the semantic similarity between pairs of lexical units. The evaluation results obtained for wnet2vec are around 15% superior to the results obtained for word2vec with the same mainstream evaluation data set SimLex-999 (Hill et al., 2016).

2 Distributional vectors from ontological graphs

For a given word \( w \), its distributional representation \( \vec{w} \) (aka word embedding) is a high dimension vector whose elements \( \vec{w}_i \) record real valued scores expressing the strength of the semantic affinity of \( w \) with other words in the vocabulary. The usual source of these scores, and ultimately the empirical base of word embeddings, has been the frequency of co-occurrence between words taken from large collections of text.

The goal here instead is to use semantic networks as the empirical source of word embeddings. This will permit that the lexical knowledge that is encoded in a semantic graph be re-encoded as an embeddings matrix compiling the distributional vectors of the words in the vocabulary.

To determine the strength of semantic affinity of two words from their representation in a semantic graph, we follow this intuition: the larger the number of paths and the shorter the paths connecting any two nodes the stronger is their affinity.

To make this intuition operative we resort to the following procedure, to be refined later on. First, the semantic graph \( G \) is represented as an adjacency matrix \( M \) such that iff two nodes of \( G \) with words \( w_i \) and \( w_j \) are related by an edge representing a direct semantic relation between them, the element \( M_{ij} \) is set to 1 (to 0 otherwise).

Second, to enrich \( M \) with scores that represent the strength of semantic affinity of nodes not directly connected with each other by an edge, the following cumulative iteration is resorted to

\[
M^{(n)}_G = I + \alpha M + \alpha^2 M^2 + \ldots + \alpha^n M^n \quad (1)
\]

where \( I \) is the identity matrix; the \( n \)-th power of

---

1For the vast number of applications of WordNet, see http://lit.csci.unt.edu/∼wordnet
the transition matrix, $M^n$, is the matrix where each $M_{ij}$ counts the number of paths of length $n$ between nodes $i$ and $j$; and $\alpha < 1$ is a decay factor determining how longer paths are dominated by shorter ones.

Third, this iterative procedure is pursued until it converges into matrix $M_G$, which is analytically obtained by an inverse matrix operation given by

$$M_G = \sum_{e=0}^{\infty} (\alpha M)^e = (I - \alpha M)^{-1} \quad (2)$$

3 WordNet embeddings

In order to assess this procedure, we use it to convert a mainstream ontological graph into an embeddings matrix. We use Princeton WordNet (Fellbaum, 1998) as our working semantic network. This is a lexical ontology for English with over 120k concepts that are related by over 25 types of semantic relations and comprise over 155k words (lemmas), from the categories Noun (with 117k words), Verb, Adjective and Adverb.

The quality of the resulting semantic space (based on a semantic network) is assessed by resorting to the mainstream procedure to evaluate semantic spaces: (i) it is used to solve the task of determining the semantic similarity between words in a mainstream test data set used in the literature; (ii) its performance is compared to the performance of a mainstream semantic space (based on a text collection), namely word2vec (Mikolov et al., 2013b), which serves as our baseline.

The base data set was obtained by extracting a sub-graph from WordNet that supports a 60k word distributional matrix. All parts of speech in WordNet were considered.

The nodes in WordNet are related by different types of semantic relations (e.g. hypernymy, merononymy, etc.). Relations of different types were taken into account with identical weight for the sake of the conversion of the graph into a matrix.

Upon applying the conversion procedure by resolving equation (2), its outcome $M_G$ was subject to the Positive Point-wise Mutual Information transformation (PMI+) seeking to reduce the eventual bias introduced by the conversion towards words with more senses.

\[\text{Evaluate word pairs using } \text{Gensim package (Rehurek and Sojka, 2010) to determine the performance of both semantic spaces, the wnet2vec and the word2vec embeddings.}\]

\[\begin{array}{|c|c|}
\hline
\text{Model} & \text{Similarity} \\
\hline
\text{wnet2vec} & 0.50 \\
\text{word2vec} & 0.44 \\
\hline
\end{array}\]

Table 1: Performance in semantic similarity task over SimLex-999 given by Spearman’s coefficient (higher score is better).

For the sound application of the conversion, each line in $M_G$ was normalized, using L2-norm, so that it corresponds to a vector whose scores sum to 1, corresponding to a transition matrix.

Finally, we used Principal Component Analysis (PCA) (Wold et al., 1987) to transform the matrix, reducing the size of the vectors and setting to 850 the dimension of the encoded semantic space.

To assess the quality of the resulting semantic space, we resorted to the test data set SimLex-999 (Hill et al., 2016), containing a list of 999 pairs of words. Each pair is associated with a score, on a 0-10 scale, that indicates the strength of the semantic similarity between the words in that pair. For each pair, with the resulting embedding matrix, the cosine between the vectors of the words in that pair is calculated and mapped into the 0-10 scale. The outcome is compared to the gold standard scores in SimLex-999 resorting to Spearman’s rank correlation coefficient. The respective scores are displayed in Table 1.

4 Discussion

These results indicate a clear advantage of around 15% of the WordNet embeddings, scoring 0.50, over the word2vec embeddings, scoring 0.44. This indicates that the proposed conversion procedure is very effective.

WordNet embeddings is a semantic space empirically based on an internal language resource: on a systematic elicitation and recording of the semantic relations between words, thus being closely aligned with the lexical knowledge in the minds of speakers. Word2vec, in turn, is a semantic space empirically based on an external language resource: on records of contingent language usage, namely some texts that were produced by a population of language users and happened to be
collected together. Hence, while words related by some semantic relation are likely to be linked in WordNet, they may happen to rarely or never occur in relevant context windows, as practical constraints on the production and usage of language may not favor that. This may help to explain the advantage of wnet2vec over word2vec.5

The conversion procedure is composed by a number of steps where each may receive a range of configurations. This opens a large experimental space of which the experiment in Section 3 instantiates one set of coordinates. In the remainder of the present section we justify the eventual empirical settings used and discuss the lessons learned by exploring this experimental space. The conversion procedure will be revisited in a backwards fashion, from its final to its initial steps, with the experiments being performed over the 60k subset identified in Subsection 4.3.

4.1 Matrix manipulation

Vector dimension: There have been studies indicating the positive effect of the reduction of the dimensionality of the semantic space (e.g. (Underhill et al., 2007; Grünauer and Vincze, 2015)). We experimented with a range of final vector dimensions, namely sizes 100, 300, 850, 1000 and 3000, also over evaluation data sets other than just SimLex-999.6 Results obtained consistently indicated that size 850 leads to better performance.7

Dimensionality reduction: We compared two different techniques for dimensionality reduction, PCA (Wold et al., 1987) and a neural network approach. For the neural solution, the encoder-decoder architecture with a Sigmoid activation function was employed. The model was trained using a Nadam optimizer with binary cross entropy as loss metric. Experimentation consistently indicated that PCA is substantially more successful.

Normalization and bias: We contrasted the performance of the WordNet embeddings obtained with and without normalization of the distributional vectors. Results consistently indicated the advantage of doing normalization, even if for a small margin, with a delta of around 0.08.

Ablation tests were done also with respect to PMI+, which indicated a clear advantage of applying it.

4.2 Graph manipulation

Decay factor: The best results were achieved with $\alpha = 0.75$, after experimenting with values in the range 0.65 to 0.85.

Picking semantic relations: Concepts in WordNet are connected via semantic relations of different types. The relations of Hypernymy/Hyponymy, Synonymy and Antonymy play an essential role in structuring a semantic network, as without them the network could not exist. We undertook experiments where all semantic relations or only these kernel relations were taken into account for the conversion procedure, with results indicating a clear advantage for using all relations.

Weighting semantic relations: In the definition of a semantic network, some types of relations appear as necessary (e.g. Hypernymy), while other appears as more secondary (e.g. Meronymy). It might thus happen that the conversion of a semantic network into a semantic space might be optimized if different weights were assigned to different relations accordingly. We ran an experiment where different weights were assigned to different relations, namely hypernymy, hyponymy, antonymy and synonymy got 1, meronymy and holonymy 0.8 and other relations 0.5; and another experiment where all types of semantic relation were assigned the same weight. Better results were obtained with the latter.

4.3 Base data sets

Subgraphs: The conversion procedure relies on equation (2), whose complexity is dominated by the calculation of the inverse matrix, which is of exponential order. For the Princeton WordNet graph, with over 120k concepts, given the size of the adjacency matrix $M^1$ is over $120k \times 120k$, its calculation and the overall conversion of the ontological graph into the final embeddings matrix faces substantial challenges in terms of the memory footprint. To cope with this issue, we resorted to initial subgraphs of manageable size.8

5Naturally, the comparative advantage between a semantic space based on a semantic network and another based on a collection of texts depends also on the sizes of the network and of the collection. The training corpus of word2vec-GoogleNews-vectors we used is one of the largest, with an impressive amount of 100 billion tokens, and a vocabulary of 3 million types, which differently from the vocabulary units in WordNet, are wordforms, not lemmas (Mikolov et al., 2013a).

6More on evaluation data sets in Section 4.4

7The vector size in word2vec embeddings is 300.

8To invert a 60k matrix, numpy used all memory available in a machine with 32 CPUs/2.50GHz and 430Gb RAM.
We reduced the size of $M^1$ by eliminating more sparse rows (rows with more zero elements), corresponding to eliminating words in concepts with lower number of outgoing edges in WordNet. Rows were ordered by decreasing sparsity, with rows with identical level of sparsity (identical number of zero elements) randomly ordered among themselves. The first 25k, 30k, 45k and 60k rows were extracted and used in the conversion process. To maximize overlap with test set SimLex-999, its words in WordNet were retained. The performance scores of the resulting models are displayed in Table 2.

<table>
<thead>
<tr>
<th>Random subgraphs</th>
<th>25k</th>
<th>30k</th>
<th>45k</th>
<th>60k</th>
</tr>
</thead>
<tbody>
<tr>
<td>Semantic similarity</td>
<td>0.45</td>
<td>0.47</td>
<td>0.49</td>
<td>0.50</td>
</tr>
</tbody>
</table>

Table 2: Performance of wnet2vec in similarity task over SimLex-999 (Spearman’s coefficient).

The larger the size of the WordNet subgraph the better is the performance of the resulting embeddings. As they contain more concepts, which on average are closer to each other, larger subgraphs tend to be denser and generate less sparse adjacency matrices. This supports semantic spaces with distributional vectors with more discriminative information on the semantic affinity of a word with respect to others.

The progression of scores in Table 2, for subgraphs with matrices in the range 25k-60k, supports the conjecture that when enough computational means are available and the full 155k word WordNet be used, the performance of the resulting embeddings may still improve by a substantial margin over the result now observed for the 60k matrix, with less than half of the words.

Additionally, we experimented with two specific subgraphs that were not randomly extracted from WordNet, namely: the subgraph supporting the matrix with the 13k most frequent words of English,$^9$ and the subgraph supporting the matrix with the 13k words used in (De Deyne et al., 2016),$^{10}$ which have been selected to act as cue words in psycholinguistic experiments for eliciting associated words from subjects. The performance results of the resulting models are displayed in Table 3.

<table>
<thead>
<tr>
<th>Specific subgraphs</th>
<th>13k most frequent</th>
<th>13k cue words</th>
</tr>
</thead>
<tbody>
<tr>
<td>Similarity</td>
<td>0.47</td>
<td>0.50</td>
</tr>
</tbody>
</table>

Table 3: Performance of wnet2vec in similarity task over SimLex-999 given by Spearman’s coefficient. First row indicates the sizes of the matrices supported by specific subgraphs.

These matrices have less than 1/4 of the size of the 60k matrix, and yet they show a better than expected approximation to its performance, taking into account the progression registered in Table 2. These results indicate that larger size is not the only factor improving the performance of WordNet embeddings. Very interestingly, they seem to indicate that words more commonly used may support semantic spaces that are more accurate to discriminate semantic similarity.

Frequency of occurrence in texts plays no direct role in the conversion of semantic networks into semantic spaces by equation (2). Hence this effect likely results from the fact captured by one of the Zipf word distributions, that on average more frequent words are more ambiguous than less frequent ones: On average more frequent words express more concepts — that is, they occur in more WordNet synsets — and thus enter in more outgoing edges in the semantic network, and this should support less sparse vectors in the semantic space.

This explanation is empirically supported by the fact that the word ambiguity rates are 2.7 and 2.8, in the subgraphs with 13k cue words and with 13k most frequent words, respectively, while there is a lower word ambiguity rate of around 1.5 for the random graph with 60k words.$^{11}$

**Parts of Speech:** Princeton WordNet covers nouns, adjectives, verbs and adverbs. Nouns (117k) are the largest portion of all words (155k) in the graph and, among the different POS, they support the most dense subgraph of semantic relations. We run experiments with words from all POS categories, and where only Nouns were considered. While results obtained with Nouns only (0.44) are not that distant from the results obtained with all POS (0.50), the latter setting consistently showed better performance.

---

$^9$To reach 13k, we used the 10k most common English words, as determined by n-gram frequency analysis of the Google’s Trillion Word Corpus, from (Kaufman, 2017), supplemented with non repeating words from Wiktionary frequency lists (Wiktionary, 2017).

$^{10}$Available from https://smallworldofwords.org/en

$^{11}$This is obtained by counting $n$ lemmas for a word that enters WordNet under $n$ POS categories. Word ambiguity rate of the whole WordNet is 1.3.
4.4 Testing data and metrics

To assess the robustness of the results obtained, experiments were undertaken with: (i) yet another evaluation metric, namely Pearson’s correlation coefficient; (ii) further evaluation data sets for semantic similarity, namely RG1965 (Rubenstein and Goodenough, 1965) and Wordsim-353-Similarity (Agirre et al., 2009); (iii) and testing over another task, namely semantic relatedness, with the evaluation data sets Wordsim-353-Relatedness (Agirre et al., 2009), MEN (Bruni et al., 2012) and MTurk-771 (Halawi et al., 2012). In these experiments we used our best settings, with a random 60k subgraph, and our second best settings, with the best model with a specific 13k subgraph, cf. Subsection 4.3.

Additional metric: The evaluation scores obtained over SimLex-999 with the Pearson’s coefficient are basically aligned with the scores already obtained with Spearman’s coefficient, confirming the superiority of the WordNet embeddings.

Additional data sets: Even with a number of test pairs much lower than the pairs in SimLex-999 and built under less standard procedure, and thus supporting less reliable results, we evaluated our models over the Wordsmith353-S and RG1965 data sets. Wnet2vec showed competitive performance when put side by side with word2vec even though their scores were not superior. With these smaller alternative data sets, the results for the specific 13k model were slightly superior to the results for the random 60k model.

Additional task: The relation “semantic relatedness” is broader and less well defined than the relation “semantic similarity”. Experiments with a second task of determining semantic relatedness showed that word2vec performs clearly better on this task than on the task of semantic similarity, while wnet2vec in general performs worst on it. Wnet2vec is thus less prone than word2vec to get fooled by words that are just semantically related by not necessarily similar. This indicates that the superiority of wnet2vec in the similarity task results from an enhanced discriminative capacity, with it being better both at judging as similar, words that are actually similar, and at judging as non similar, not only words that may be clearly non similar but also words that are semantically related, and thus may be close to be similar.

The results obtained with these experiments are displayed in Table 4.12

5 Related work

From semantic spaces to semantic networks: There has been a long research tradition on semantic networks enhanced with information extracted from text, including distributional vectors, which in the limit may encompass semantic networks obtained from semantic spaces. As a way of illustration, among many others, this includes the work on semantic relations determined from patterns based on regular expressions, either hand crafted (Hearst, 1992), or learned from corpora (Snow et al., 2005); work on semantic relations predicted by classifiers running over distributional vectors (Baroni et al., 2012; Roller et al., 2014; Weeds et al., 2014); work on semantic relations obtained with deep learning that integrates distributional information and patterns of grammatical dependency relations (Shwartz et al., 2016), including the hard task of distinguishing synonymy from antonymy (Nguyen et al., 2017); etc. While being highly relevant for a unified account of lexical semantics, this line of research addresses the conversion direction, from semantic spaces to semantic networks, that is not the major focus of this paper.

From semantic networks to semantic spaces: Work towards the conversion direction that is of interest here is more recent. As a way of illustration, among others, one can mention (Faruqui et al., 2015), which explored retrofitting to refine distributional representations using relational information, and (Yu and Dredze, 2014), which focused also on refining word embeddings with lexical knowledge, but which are not addressing the goal of obtaining semantic spaces solely on the basis of semantic networks as we do here.

That is the aim also of recent work like (Camacho-Collados et al., 2015) who improve the embeddings built from data sets made of selected Wikipedia pages by resorting to the local, one-edge relations of each relevant word in the WordNet graph.

Further recent works worth mentioning include (Vendrov et al., 2015) that resorted to order embeddings, which however do not preserve distance and/or do not preserve directionality under...
the relevant semantic relations; (Nickel and Kiela, 2017) that experimented with computing embeddings not in Euclidean but in hyperbolic space, namely the Poincaré ball model. A shortcoming with these proposals is that their outcome is not easily plugged into neural models. Also they are not fit to evaluation on external tasks, like the semantic similarity task, with their evaluation being rather based on their ability to complete missing edges from ontological graphs. In contrasts, an example of the suatability of wnet2vec to be plugged into neural models and of its application in a downstream task is reported in (Rodrigues et al., 2018), where these embeddings support the prediciton of brain activation based on neural networks.

There has been also a long tradition of research on learning vector embeddings from multi-relational data of which, among many others, one can refer (Bordes et al., 2013), (Lin et al., 2015), and (Nickel et al., 2016). Though to a large extent these are generic approaches for graph to vectors conversion, also here the major focus has been on exploring these models on their ability to complete missing relations in knowledge bases rather than to experiment them on natural language processing and lexical semantics.

Other related approaches worth of note are (De Deyne et al., 2016) and (Goikoetxea et al., 2015). While being based also on the iterative conversion procedure used here, the first concentrates however on converting, not a semantic network, but a fragment of the lexicon represented under a feature-based approach into a semantic space.

While seeking to obtain WordNet embeddings, the second resorts, however, not to a genuine conversion procedure, but to a lossy intermediate “textual” representation: it generates sequences of words by concatenating words visited by random walks over the WordNet; this “artificial text” is a partial and contingent reflection of the semantic network and is used to obtain distributional vectors by resorting to typical word embeddings techniques based on text.

**Distances in a semantic graph:**

The task of determining the semantic similarity between two words can be performed not only on the basis of the distance of their respective vectors in a semantic space, but also on the basis of the distance of the respective concepts in a lexical semantic network, like WordNet. There has been a long research tradition on this issue whose major proposals include (Jiang and Conrath, 1997), (Lin, 1998), (Leacock and Chodorow, 1998), (Hirst and St-Onge, 1998), (Resnik, 1999), among others, which received nice comparative assessments in (Ferlez and Gams, 2004) and (Budanitsky and Hirst, 2006), including their correlation with human judgments.

In this context, it is worth of note the work by (Hughes and Ramage, 2007), which resorts to random graph walks over WordNet edges. Differently from our approach, its goal is to obtain word-specific stationary probability distributions — such that the semantic affinity of two words is based on the similarity of their probability distributions —, rather than to obtain vectorial representations for words in a shared distributional se-

<table>
<thead>
<tr>
<th>data set</th>
<th>task</th>
<th>size</th>
<th>overlap %</th>
<th>w2vec Spearman coef</th>
<th>Pearson coef</th>
</tr>
</thead>
<tbody>
<tr>
<td>SimLex-999</td>
<td>simil</td>
<td>999</td>
<td>99.8</td>
<td>0.44</td>
<td>0.45</td>
</tr>
<tr>
<td>RG1965</td>
<td>simil</td>
<td>65</td>
<td>100.0</td>
<td>0.75</td>
<td>0.75</td>
</tr>
<tr>
<td>Wordsim353-S</td>
<td>simil</td>
<td>203</td>
<td>98.0</td>
<td>0.74</td>
<td>0.73</td>
</tr>
<tr>
<td>Wordsim353-R</td>
<td>relat</td>
<td>252</td>
<td>97.6</td>
<td>0.61</td>
<td>0.58</td>
</tr>
<tr>
<td>MEN</td>
<td>relat</td>
<td>3000</td>
<td>44.9</td>
<td>0.70</td>
<td>0.68</td>
</tr>
<tr>
<td>MTURK-771</td>
<td>relat</td>
<td>771</td>
<td>99.7</td>
<td>0.66</td>
<td>0.63</td>
</tr>
</tbody>
</table>

Table 4: Performance of different models in the semantic similarity (simil) and relatedness (relat) tasks over different data sets measured by Spearman’s and Pearson’s coefficients. Models used: w2vec (w2vec); wnet2vec with the random 60k subgraph (n2vec 60k r); and wnet2vec with the best specific 13k subgraph (n2vec 13k s), cf. Subsection 4.3. Overlap with the vocabulary of wnet2vec 60k random appears in the fourth column.

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mantic space.

The focus of the present paper is on an effective method to convert a semantic network into a semantic space, with the graph-based affinity obtained by the chaining of "local" one-edge distances ensured by the iteration in (1)-(2) being central for that goal.

It will be interesting to understand whether it will be possible to consider, as an alternative, those graph-based metrics of semantic similarity for any two nodes anywhere in the graph — resorting to the "non-local" multi-edge distance between the two input words. It remains to be understood whether they can be resorted to as the basis of an "all vs. all" type of procedures for an exhaustive screening of the graph that are computationally tractable — thus aiming at keeping up with an effective method for graph to matrix conversion of an entire lexical semantic network that resists the eventual exponential explosion.

6 Conclusions

In this paper, we offer a contribution towards a unified account of lexical semantics. We propose a methodology to convert from semantic networks, that are encoded in ontological graphs and empirically based on systematic linguistic intuitions (in their higher quality incarnations), to semantic spaces, that are encoded in distributional vectors and empirically based on very large collections of texts (in their higher quality implementations). This conversion methodology relies on a straightforward yet powerful intuition — the larger the number of paths and the shorter the paths connecting two nodes in an ontological graph the stronger is their semantic affinity —, with iteration (1) making it operative in order to generate a distributional matrix from an ontological graph.

We report also on the results of assessing this conversion methodology with a case study, namely by applying it to a subgraph of WordNet with less than half of its words (60k), randomly selected from the ones whose senses have a larger number of outgoing edges. The resulting distributional vectors wnet2vec were evaluated under the mainstream task of determining the semantic similarity of words arranged in pairs, against the mainstream gold standard SimLex-999, with very good results. The performance of wnet2vec was around 15% superior to the performance of word2vec, trained on a 100 billion token collection of texts. This indicates that the proposed conversion procedure is very effective and that the WordNet embeddings are competitive when compared to text based embeddings.

It is nevertheless worth underlying that the research goal of this paper was not to search for word embeddings that outperform all previous proposals known in the literature in terms of intrinsic evaluation tasks, like semantic similarity, etc., or when they are embedded in larger systems. Its research goal was rather to demonstrate that it is feasible to create very effective word embeddings from semantic networks with a straightforward and yet powerful method of conversion from semantic networks to semantic spaces that, given its simplicity, offer the promise to generalize very well for more types of lexical networks and ontologies other than just WordNet, which was the case study used here.

The fact that less than half of the words in WordNet were used in the reported experiment reinforces this positive expectation with respect to the strength of the proposed approach, and point towards future work that will seek to use larger portions of WordNet, as computational limitation can be overcome.

The results reported in this paper thus hint at very promising research avenues, including, among others, experiments with further ontologies of different domains, empirical origins, etc.; with cross-lingual triangulation with aligned WordNets and aligned embeddings; with reciprocal reinforcement of ontological graphs and distributional vectors; with other metrics of semantic affinity in a graph, etc.

The wnet2vec data and software and their future updates are distributed at https://github.com/nlx-group/WordNetEmbeddings

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Knowledge Graph Embedding with Numeric Attributes of Entities

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Abstract

Knowledge Graph (KG) embedding projects entities and relations into low dimensional vector space, which has been successfully applied in KG completion task. The previous embedding approaches only model entities and their relations, ignoring a large number of entities’ numeric attributes in KGs. In this paper, we propose a new KG embedding model which jointly model entity relations and numeric attributes. Our approach combines an attribute embedding model with a translation-based structure embedding model, which learns the embeddings of entities, relations, and attributes simultaneously. Experiments of link prediction on YAGO and Freebase show that the performance is effectively improved by adding entities’ numeric attributes in the embedding model.

1 Introduction

Recently, a number of Knowledge Graphs (KGs) have been created, such as DBpedia (Lehmann, 2015), YAGO (Mahdisoltani et al., 2015), and Freebase (Bollacker et al., 2008). KGs encode structured information of entities in the form of triplets (e.g. ⟨Microsoft,isLocatedIn,UnitedStates⟩), and have been successfully applied in many real-world applications. Although KGs contain a huge amount of triplets, most of them are incomplete. In order to further expand KGs, much work on KG completion has been done, which aims to predict new triplets based on the existing ones in KGs. A promising group of research for KG completion is known as KG embedding. KG embedding approaches project entities and relations into a continuous vector space while preserving the original knowledge in the KG. KG embedding models achieve good performance in KG completion in terms of efficiency and scalability. TransE is a representative KG embedding approach (Bordes et al., 2013), which projects both entities and relations into the same vector space: if a triplet (head entity, relation, tail entity) (denoted as (h, r, t)) holds, TransE wants that h + r ≈ t. The embeddings are learned by minimizing a margin-based ranking criterion over the training set. TransE model is simple but powerful, and it gets promising results on link prediction and triple classification problems. There are several enhanced model of TransE, including TransR (Lin et al., 2015), TransH (Wang et al., 2014) and TransD (Ji et al., 2015) etc. By introducing new representations of relational translation, later approaches achieve better performance at the cost of increasing model complexity. Recent surveys (Wang et al., 2017; Nickel et al., 2016) give detailed introduction and comparison of various KG embedding approaches.

However, most of the existing KG embedding approaches only model relational triplets (i.e. triplets of entity relations), while ignoring a large number of attributive triplets (i.e. triplets of entity attributes, e.g. ⟨Microsoft,wasFoundedOnDate,1975⟩) in KGs. Attributive triplets describe various attributes of entities, such as ages of people or areas of a city. There are a huge number of attributive triplets in real KGs, and we believe that information encoded in these triplets is also useful for predicting entity relations. Having the above motivation, we propose a new KG embedding approach that jointly model entity relations and entities’ numeric attributes. Our approach consists of two component models,
structure embedding model and attribute embedding model. The structure embedding model is a translational distance model that preserves the knowledge of entity relations; the attribute embedding model is a regression-based model that preserves the knowledge of entity attributes. Two component models are jointly optimized to get the embeddings of entities, relations, and attributes. Experiments of link prediction on YAGO and Freebase show that the performance is effectively improved by adding entities’ numeric attributes in the embedding model.

2 Our Approach

To effectively utilize numeric attributes of entities in KG embedding, we propose TransEA, which combine a new attribute embedding model with the structure embedding model of TransE. Two component models in TransEA share the embeddings of entities, and they are jointly optimized in the training process.

2.1 Structure Embedding

The structure embedding directly adopts the translation-based method in TransE to model the relational triplets in KGs. Both Entities and relations in a KG are represented in the same vector space $\mathbb{R}^d$. In a triplet $\langle h, r, t \rangle$, the relation is considered as a translation vector $r$, which connects the vector of entities $h$ and $t$ with low error, i.e. $h + r \approx t$. The score function of a given triplet $\langle h, r, t \rangle$ is defined as

$$f_r(h, t) = -||h + r - t||_{1/2}$$

(1)

$||x||_{1/2}$ denotes either the $L1$ or $L2$ norm. For all the relational triplets in the KG, the loss function of the structure embedding is defined as:

$$L_R = \sum_{\langle h, r, t \rangle \in S} \sum_{\langle h', r', t' \rangle \in S'} \left[ \gamma + f_r(h, t) - f_r(h', t') \right]_+$$

(2)

where $[x]_+ = \max(0, x)$, $S'$ denotes the set of negative triplets constructed by corrupting $\langle h, r, t \rangle$, i.e. replacing $h$ or $t$ with a randomly chosen entity in KG; $\gamma > 0$ is a margin hyper-parameter separating positive and negative triplets.

2.2 Attribute Embedding

Attribute embedding model takes all the attributive triplets in a KG as input, and learns embeddings of entities and attributes. Both entities and attributes are represented as vectors in space $\mathbb{R}^d$. In an attributive triplet $\langle e, a, v \rangle$, $e$ is an entity, $a$ is an attribute, and $v$ is the value of the entity’s attribute. In our approach, we only consider attributive triplets containing numeric values or values can be easily converted into numeric ones. For a triplet $\langle e, a, v \rangle$, we define a score function as

$$f_a(e, v) = -||a^T \cdot e + b_a - v||_{1/2}$$

(3)

where $a$ and $e$ are vectors of attribute $a$ and entity $e$, $b_a$ is a bias for attribute $a$. The idea of this score function is to predict the attribute value by a linear regression model of attribute $a$; the vector $a$ and bias $b_a$ are the parameters of the regression model. For all the attributive triplets in the KG, the loss function of the attribute embedding is defined as:

$$L_A = \sum_{\langle e, a, v \rangle \in T} f_a(e, v)$$

(4)

where $T$ is the set of all attributive triplets with numeric values in the KG.

2.3 Joint Model

To combine the above two component models, TransEA minimizes the following loss function:

$$L = (1 - \alpha) \cdot L_R + \alpha \cdot L_A$$

(5)

where $\alpha$ is a hyper-parameter that balances the importance of structure and attribute embedding. In the joint model, we let the embeddings of entities shared by two component models. Entities, relations, and attributes are all represented by vectors in $\mathbb{R}^d$. We implement our approach by using TensorFlow\(^1\), and the loss function is minimized by performing stochastic gradient descent.

3 Experiments

3.1 Datasets

The following two datasets are used in the experiments, Table 1 shows their detail information.

**YG58K.** YG58K is a subset of YAGO3 (Mahdisoltani et al., 2015) which contains about 58K entities. YG58K is built by removing entities from YAGO3 that appear less than 25 times or have no attributive triplets. All the remaining triplets are then randomly split into training/validation/test sets.

\(^1\)https://www.tensorflow.org
FB15K. FB15K is a subset of triplets extracted from Freebase\(^2\). This subset of Freebase was originally used in (Bordes et al., 2013), and then widely used for evaluating KB completion approaches. Since our approach consumes attributive triplets, we extract all the attributive triplets of entities in FB15K from Freebase to build the evaluation dataset.

![Table 1: Statistics of datasets](https://everest.hds.utc.fr/doku.php?id=en:transe)

<table>
<thead>
<tr>
<th>Datasets</th>
<th>YG58K</th>
<th>FB15K</th>
</tr>
</thead>
<tbody>
<tr>
<td># Relational Triplets</td>
<td>497783</td>
<td>592213</td>
</tr>
<tr>
<td># Attribution Triplets</td>
<td>130278</td>
<td>24034</td>
</tr>
<tr>
<td># Entities</td>
<td>58130</td>
<td>14951</td>
</tr>
<tr>
<td># Relations</td>
<td>59071</td>
<td>483142</td>
</tr>
<tr>
<td># Attributes</td>
<td>1345</td>
<td>399480</td>
</tr>
<tr>
<td># Train Sets</td>
<td>399480</td>
<td>483142</td>
</tr>
<tr>
<td># Valid Sets</td>
<td>49171</td>
<td>59071</td>
</tr>
<tr>
<td># Test Sets</td>
<td>49132</td>
<td>50000</td>
</tr>
</tbody>
</table>

3.2 Experimental setup

In the experiments, Mean Rank (the mean rank of the original correct entity), Hits@\(k\) (the proportion of the original correct entity to the top \(k\) entities), and MRR (the mean reciprocal rank) are used as evaluation metrics. Given a testing triplet \((h, r, t)\), we replace the head \(h\) by every entity in the KGs and calculate dissimilarity measures according to the score function \(f_r\). Ranking the scores in ascending order, then we get the rank of the original correct triplet to compute the evaluation metrics. And we repeat the procedure when removing the tail \(t\) instead of the head \(h\). We name the evaluation setting as “Raw”. While corrupted triplets that appear in the train/valid/test sets (except the original correct one) may underestimate the metrics, we also filter out those corrupted triplets before getting the rank of each testing triplet and we call this process “Filter”.

Because our approach is built based on TransE, we compare our approach with TransE to see whether adding attribute embedding in the model improves the performance of link prediction. For TransE and TransEA, we consider the learning rate \(\lambda\) among \([0.1, 0.01, 0.001]\), the margin \(\gamma\) among \([1, 2, 4, 10]\), the dimensions of embedding \(d\) among \([20, 50, 100, 150]\), the types of norm in two score functions among \([L1, L2]\), and \(\alpha\) among \([0.2, 0.3, 0.4, 0.5, 0.6]\). Based on the mean rank in validation set, we select the best configurations for two approaches. On the YG58K dataset, the best parameter configuration for TransE is \((\lambda = 0.1, \gamma = 4, d = 50, f_r = L1, f_a = L1)\), and for TransEA is \((\lambda = 0.001, \gamma = 4, d = 50, f_r = L1, f_a = L1, \alpha = 0.6)\). On the FB15K dataset, the best parameter configuration for TransE is \((\lambda = 0.01, \gamma = 1, d = 50, f_r = L1, f_a = L1)\), and for TransEA is \((\lambda = 0.001, \gamma = 2, d = 100, f_r = L1, f_a = L1, \alpha = 0.3)\).

3.3 Results

Table 2 shows the results of link prediction on YG58K and FB15K datasets. The results of predicting head and tail entities are outlined separately, and we also report the overall results by considering prediction of both head and tail entity. According to the overall results, TransEA outperforms TransE on both two datasets in terms of all the three metrics. TransEA gets lower Mean Ranks by about 13 on YG58K dataset; the MRR and Hits@\(k\) of two approaches are very close, TransEA gets slightly better results, the improvements of MRR and Hits@\(k\) are 0.1-0.2% and 0-0.3%. On FB15K dataset, TransEA gets lower Mean Ranks by 13, and it also gets better results than TransE according to MRR, Hits@10 and Hits@3.

Table 3 shows the results of different relational categories. In general, TransEA has superiority on two datasets, except one-to-many relation for replacing head entity on YG58K. And the improvements on FB15K are larger than YG58K.

In order to figure out which relations are predicted more accurately by TransEA, Table 4 lists the top 5 improved relations in terms of Hits@10 on YG58K. It shows the best improvement of Hits@10 is 25% for the relation isInterestedIn. The second one is 12.5% for hasAcademicAdvisor, and the third is 6.3% for worteMusicFor. Entities of these three relations have plenty of numeric attributes (wasBornOnDate, diedOnDate) describing people, we believe they are helpful to improving the embeddings of entity relations. Entities in relational triplets about livesIn, (e.g. (HankAzaria,livesIn,NewYork)), also have some numeric attributes (hasLatitude, hasLongitude, hasNumberOfPeople, etc), therefore TransEA gets a 5% improvement of Hits@10.

On FB15K dataset, five relations have 100%
Table 2: Link prediction results

<table>
<thead>
<tr>
<th>Dataset</th>
<th>Entity</th>
<th>Model</th>
<th>Mean Rank</th>
<th>MRR(%)</th>
<th>Hits@10(%)</th>
<th>Hits@3(%)</th>
<th>Hits@1(%)</th>
</tr>
</thead>
<tbody>
<tr>
<td>YG58K</td>
<td>Head</td>
<td>TransE</td>
<td>930</td>
<td>3.1</td>
<td>9.1</td>
<td>4.1</td>
<td>1.0</td>
</tr>
<tr>
<td></td>
<td></td>
<td>TransEA</td>
<td>944</td>
<td>3.1</td>
<td>9.4</td>
<td>4.1</td>
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</tr>
<tr>
<td></td>
<td>Tail</td>
<td>TransE</td>
<td>230</td>
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<td>27.0</td>
<td>12.2</td>
<td>4.5</td>
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<tr>
<td></td>
<td></td>
<td>TransEA</td>
<td>229</td>
<td>8.5</td>
<td>27.6</td>
<td>12.4</td>
<td>4.7</td>
</tr>
<tr>
<td>All</td>
<td>TransE</td>
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<td>18.0</td>
<td>8.2</td>
<td>2.8</td>
<td></td>
</tr>
<tr>
<td></td>
<td>TransEA</td>
<td>482</td>
<td>7.7</td>
<td>24.3</td>
<td>8.2</td>
<td>2.9</td>
<td></td>
</tr>
<tr>
<td>FB15K</td>
<td>Head</td>
<td>TransE</td>
<td>230</td>
<td>14.5</td>
<td>47.0</td>
<td>26.2</td>
<td>11.8</td>
</tr>
<tr>
<td></td>
<td></td>
<td>TransEA</td>
<td>229</td>
<td>14.5</td>
<td>49.5</td>
<td>28.0</td>
<td>11.8</td>
</tr>
<tr>
<td></td>
<td>Tail</td>
<td>TransE</td>
<td>168</td>
<td>17.6</td>
<td>54.8</td>
<td>32.8</td>
<td>16.4</td>
</tr>
<tr>
<td></td>
<td></td>
<td>TransEA</td>
<td>157</td>
<td>17.6</td>
<td>57.5</td>
<td>34.5</td>
<td>16.3</td>
</tr>
<tr>
<td>All</td>
<td>TransE</td>
<td>204</td>
<td>16.0</td>
<td>50.9</td>
<td>71.9</td>
<td>29.5</td>
<td></td>
</tr>
<tr>
<td></td>
<td>TransEA</td>
<td>191</td>
<td>26.7</td>
<td>53.5</td>
<td>77.3</td>
<td>31.3</td>
<td></td>
</tr>
</tbody>
</table>

Table 3: Hits@10(%) by relational category in the filtered evaluation setting. (N. stand for MANY)

<table>
<thead>
<tr>
<th>Relation</th>
<th>TransE</th>
<th>TransEA</th>
</tr>
</thead>
<tbody>
<tr>
<td>isInterestedIn</td>
<td>50.0</td>
<td>75.0</td>
</tr>
<tr>
<td>hasAcademicAdvisor</td>
<td>31.3</td>
<td>43.8</td>
</tr>
<tr>
<td>wroteMusicFor</td>
<td>12.5</td>
<td>18.8</td>
</tr>
<tr>
<td>livesIn</td>
<td>23.8</td>
<td>28.8</td>
</tr>
<tr>
<td>hasNeighbor</td>
<td>48.1</td>
<td>52.8</td>
</tr>
</tbody>
</table>

Table 4: Top 5 relations of promoted Hits@10 and their Hits@10(%) on YG58K

<table>
<thead>
<tr>
<th>Relation</th>
<th>TransE</th>
<th>TransEA</th>
</tr>
</thead>
<tbody>
<tr>
<td>business/brand/company</td>
<td>24</td>
<td>2</td>
</tr>
<tr>
<td>base/celebrity/restaurant</td>
<td>249</td>
<td>4</td>
</tr>
<tr>
<td>base/celebrity/product</td>
<td>24</td>
<td>2</td>
</tr>
<tr>
<td>music/artists_supported</td>
<td>44</td>
<td>3</td>
</tr>
<tr>
<td>sports/competition/country</td>
<td>24</td>
<td>4</td>
</tr>
</tbody>
</table>

Table 5: Top 5 relations of promoted Hit@10 and their Mean Rank on FB15K

<table>
<thead>
<tr>
<th>Relation</th>
<th>TransE</th>
<th>TransEA</th>
</tr>
</thead>
<tbody>
<tr>
<td>organization/dateFounded</td>
<td>50.0</td>
<td>75.0</td>
</tr>
<tr>
<td>music/artists supported</td>
<td>31.3</td>
<td>43.8</td>
</tr>
<tr>
<td>person/dateOfBirth</td>
<td>12.5</td>
<td>18.8</td>
</tr>
<tr>
<td>person/heightMeters</td>
<td>23.8</td>
<td>28.8</td>
</tr>
<tr>
<td>hasNeighbor</td>
<td>48.1</td>
<td>52.8</td>
</tr>
</tbody>
</table>

4 Conclusion

In this paper, we propose TransEA, an embedding approach which jointly models relational and attributive triplets in KGs. TransEA combines an attribute embedding model with the translation-based embedding model in TransE. Experiments on YAGO and Freebase show that TransEA achieves better performance than TransE in link prediction task. In the future, we will study how to predict missing attribute values in KGs based on KG embedding.

Acknowledgments

The work is supported by the National Key Research and Development Program of China (No. 2017YFC0804004) and the National Natural Science Foundation of China (No. 61772079).
References


Injecting Lexical Contrast into Word Vectors
by Guiding Vector Space Specialisation

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Abstract

Word vector space specialisation models offer a portable, light-weight approach to fine-tuning arbitrary distributional vector spaces to discern between synonymy and antonymy. Their effectiveness is drawn from external linguistic constraints that specify the exact lexical relation between words. In this work, we show that a careful selection of the external constraints can steer and improve the specialisation. By simply selecting appropriate constraints, we report state-of-the-art results on a suite of tasks with well-defined benchmarks where modeling lexical contrast is crucial: 1) true semantic similarity, with highest reported scores on SimLex-999 and SimVerb-3500 to date; 2) detecting antonyms; and 3) distinguishing antonyms from synonyms.

1 Introduction

Representation models grounded in the distributional hypothesis (Harris, 1954) generally fail to distinguish highly contrasting words (antonyms) from highly similar ones (synonyms), due to similar word co-occurrence signatures in text corpora (Turney and Pantel, 2010; Mohammad et al., 2013). In addition to antonymy and synonymy being fundamental lexical relations that are central to the organisation of the mental lexicon (Miller and Fellbaum, 1991; Murphy, 2010), this undesirable property of distributional word vector spaces has grave implications on their application in NLP reasoning and understanding tasks. As shown in prior work (Pham et al., 2015; Mrkšić et al., 2016; Kim et al., 2016; Nguyen et al., 2017b; Mrkšić et al., 2017, i.a.), explicitly modeling the lexical contrast benefits text entailment, dialogue state tracking, spoken language understanding, language generation, etc.

A popular solution to address the limitation concerning lexical contrast is to move beyond stand-alone unsupervised learning. Post-processing procedures have been designed that leverage external lexical knowledge available in human- and automatically-constructed lexical resources (e.g., PPDB, WordNet): these methods fine-tune input word vectors to satisfy linguistic constraints from the external resources (Faruqui et al., 2015; Jauhar et al., 2015; Rothe and Schütze, 2015; Wieting et al., 2015; Mrkšić et al., 2016; Mrkšić et al., 2017; Vulić et al., 2017b, i.a.). This process has been termed retrofitting or vector space specialisation.

As one advantage, the post-processing methods are applicable to arbitrary input vector spaces. They are also “light-weight”, that is, they do not require large corpora for (re-)training, as opposed to joint specialisation models (Yu and Dredze, 2014; Kiela et al., 2015; Pham et al., 2015; Nguyen et al., 2016) which integrate lexical knowledge directly into distributional training objectives.

The main driving force of the retrofitting models are the external constraints, which specify which words should be close to each other in the specialised vector space (i.e., the so-called ATTRACT constraints), and which words should be far apart in the space (REPEL). By manipulating the constraints, one can steer the specialisation goal: e.g., Vulić et al. (2017a) use verb relations from VerbNet (Kipper, 2005) to accentuate VerbNet-style syntactic-semantic relations in the vector space.

1 As pointed out by Cruse (1986), antonyms have a paradoxical nature: on the one hand, they constitute the two opposites of a meaning continuum, and therefore could be seen as semantically remote; on the other hand, they are paradigmatically similar, having almost identical distributions.

2 Using a simple example, users asking for a cheap pub in northern Seattle do not want a virtual personal assistant to recommend an expensive restaurant in southern Portland.

3 An additional advantage of post-processors is their better overall performance across a range of tasks when compared to the “heavy-weight” joint models (Mrkšić et al., 2016).
Take a mini-batch of ATTRACT and REPEL pairs... For each pair, find two pseudo-negative examples... and fine-tune the vectors so that ATTRACT pairs are closest... and REPEL pairs furthest away from each other.

Figure 1: An illustration of specialisation for lexical contrast with toy examples. The specialisation model operates with two sets of external linguistic constraints: 1) ATTRACT word pairs, which have to be as close as possible in the fine-tuned vector space (e.g., irritating and annoying); and 2) REPEL word pairs, which have to be as far away from each other as possible (e.g., expensive and inexpensive).

Contributions. In this work, we investigate how different constraints affect specialisation. We show that a careful selection of external constraints can guide specialisation models to emphasise lexical contrast in the fine-tuned vector space: e.g., we indicate that direct (i.e., 1-step) WordNet hypernymy-hyponymy pairs are useful for boosting lexical contrast. Our specialised word vector spaces yield state-of-the-art results on a range of tasks where modeling lexical contrast is crucial: 1) true semantic similarity; 2) antonymy detection; and 3) distinguishing antonyms from synonyms. Our SimLex-999 (Hill et al., 2015) and SimVerb-3500 (Gerz et al., 2016) scores are the highest reported results on these datasets to date: the result on SimLex-999 is the first result on the dataset surpassing the ceiling of mean inter-annotator agreement.

2 Methodology

Specialisation Model. Post-processing models are generally guided by two broad sets of constraints: 1) ATTRACT constraints (AC) specify which words should be close to each other in the fine-tuned vector space; 2) REPEL (RC) constraints describe which words should be pulled away from each other. The nomenclature is adopted from Mrkšić et al. (2017). Earlier post-processors (Faruqui et al., 2015; Jauhar et al., 2015; Wieting et al., 2015) operate only with ATTRACT constraints, and are therefore not suited to model both aspects of lexical contrast. In this work, we employ the state-of-the-art specialisation model of Mrkšić et al. (2017) which integrates both sets of constraints into its fine-tuning process. Here, we provide only a high-level description of the model, also illustrated by Figure 1, while we refer the interested reader to the original paper for a full (technical) description.

In short, the model trains over batches of ATTRACT and REPEL pairs and contains three terms in its objective function. First, the ATTRACT term pushes two words from each ATTRACT constraint closer to each other (in terms of the cosine similarity) than to any other word present in the current batch by a margin $\delta_{att}$. Second, the REPEL term pulls away two words from each REPEL constraint so that they are further away from each other than from any other word present in the current batch (again, by a margin $\delta_{rpl}$): see Figure 1 again. Third, a regularisation term is used to preserve the useful semantic content originally present in the distribu-

<table>
<thead>
<tr>
<th>syn (AC)</th>
<th>hyp1 (AC)</th>
<th>antexp (RC)</th>
</tr>
</thead>
<tbody>
<tr>
<td>(outburst, outbreak)</td>
<td>(discordance, dissonance)</td>
<td>(smooth, shake)</td>
</tr>
<tr>
<td>(safe, secure)</td>
<td>(postmen, deliverymen)</td>
<td>(clear, obscurity)</td>
</tr>
<tr>
<td>(cordial, warmhearted)</td>
<td>(employee, worker)</td>
<td>(relief, pressure)</td>
</tr>
<tr>
<td>(answer, response)</td>
<td>(swap, exchange)</td>
<td>(half, full)</td>
</tr>
</tbody>
</table>

Table 1: Examples of linguistic constraints.
Linguistic Constraints. The constraints are in fact word pairs \((x_i, x_j), x_i, x_j \in V\), where \(V\) is the vocabulary represented in the input distributional space. First, the conflation of synonymy and antonymy relations in the input space can be obviously mitigated by assigning synonymy pairs (\(\text{syn}\)) to the ATTRACT set, and antonymy pairs (\(\text{ant}\)) to the REPEL set. Further, similar to Ono et al. (2015), it is possible to extend the (typically less exhaustive) list of antonyms by combining the available knowledge from \(\text{syn}\) and \(\text{ant}\) word pairs. If \((x_i, x_j)\) are a pair of synonyms, and \((x_i, x_k)\) are a pair of antonyms, one can add another pair \((x_j, x_k)\) to the expanded list of antonyms: this yields a larger set \((\text{antexp})\) to serve as REPEL constraints.

Finally, as the analysis of Hill et al. (2015) shows, the taxonomic hypernymy-hyponymy \(\text{IS}\)-\(\text{A}\) relation is often mistaken by true synonymy by humans. Therefore, we also experiment with direct (i.e. 1-step) \(\text{IS}\)-\(\text{A}\) pairs (\(\text{hyp1}\)) from Wordnet as another set included in the ATTRACT pairs for lexical contrast specialisation. To the best of our knowledge, the \(\text{hyp1}\) pairs were not used before for lexical contrast modeling. A selection of constraints from different sets is shown in Table 1. In what follows, we test how these different configurations of constraints influence the specialisation process.

3 Experimental Setup

Training Setup and Constraints. We train the state-of-the-art specialisation model of Mrkšić et al. (2017) using suggested settings.⁴ Adagrad (Duchi et al., 2011) is used for stochastic optimisation, batch size is 50, and we train for 15 epochs. To emphasise lexical contrast in the specialised space, we set the respective ATTRACT and REPEL margins \(\delta_{\text{att}}\) and \(\delta_{\text{rpl}}\) to the same value: 1.0. We use large 300-dim skip gram vectors with bag-of-words contexts and negative sampling (SGNS-GN) (Mikolov et al., 2013), pre-trained on the 100B Google News corpus. As all other components of the model are kept fixed, the difference in performance can be attributed to the difference in the constraints used.

We experiment with external constraints employed in prior work (Zhang et al., 2014; Ono et al., 2015): these were extracted from WordNet (Fellbaum, 1998) and the Roget thesaurus (Kipfer, 2009), and comprise 1,023,082 synonymy \((\text{syn})\) pairs and 380,873 ant pairs. The expanded \(\text{antexp}\) set of antonyms contains a total of 10,334,811 word pairs. Finally, the \(\text{hyp1}\) set extracted from WordNet contains 326,187 word pairs.

We evaluate all specialised spaces in three standard tasks with well-defined benchmarks where modeling lexical contrast is beneficial: 1) semantic similarity, 2) antonymy detection, and 3) distinguishing antonyms from synonyms. For each task, we compare against a representative selection of baselines, currently holding peak scores on the respective benchmarks. Due to a large space of models in our comparison, we refer the interested reader to the original papers for their full descriptions.

Task 1: Word Similarity. We evaluate all models on the SimLex-999 dataset (Hill et al., 2015), and SimVerb-3500 (Gerz et al., 2016), a recent verb pair similarity dataset with 3,500 verb pairs. The evaluation metric is Spearman’s \(\rho\) rank correlation.

Task 2: Antonymy Detection. For this task, we rely on the widely used Graduate Record Examination (GRE) dataset (Mohammad et al., 2008, 2013). The task, given an input cue word, is to select the best antonym from five options. Given a word vector space, we take the word with the largest cosine distance to the cue as the best antonym. The GRE dataset contains 950 questions in total. We report balanced \(F_1\) scores on the entire dataset.

Task 3: Synonymy vs. Antonymy. In this binary classification task, the system must decide whether the relation between two words is synonymy or antonymy. We use the recent dataset of Nguyen et al. (2017b), comprising 1,020 noun (N) test pairs, 908 verb (V) pairs, and 1,986 adjective (A) pairs, with the equal number of synonymy and antonymy pairs in each test subset. A classification threshold decides on the relation: all word pairs with their cosine similarity above the threshold are considered synonyms, all the others are antonyms.⁶

⁴https://github.com/nmrksic/attract-repel

⁵Unlike WordSim-353 (Finkelstein et al., 2002) or MEN (Bruni et al., 2014), SimLex and SimVerb provide explicit guidelines to discern between true semantic similarity and (more broad) conceptual relatedness, so that related but non-similar words (e.g. tiger and jungle) have a low rating.

⁶Similar to the work on hypernymy detection (Santus et al., 2014; Nguyen et al., 2017a; Vulić and Mrkšić, 2018), we tune the threshold on a validation set of 200 N pairs, 182 V pairs, and 398 A pairs, also used by Nguyen et al. (2017b).
Table 2: Task 1. Results on two word similarity benchmarks (Spearman’s ρ). Best-scoring baseline models from the literature are reported. The dashed line separates purely distributional models from the ones leveraging external lexical knowledge.

Table 3: Task 2. Results ($F_1$ scores) on the full GRE multiple-choice antonymy detection dataset.

4 Results and Discussion

Task 1: Word Similarity. A summary of the results is provided in Table 2. The most striking findings are new state-of-the-art correlation scores on both benchmarks: both are obtained by combining syn and hyp1 into ATTRACT constraints, and using the unexpanded list of antonyms as REPEL constraints. This suggests that: 1) both ATTRACT and REPEL constraints are required to provide the synergistic effect during specialisation; 2) a larger (and noisier) set of antonymy pairs is not necessarily more effective; 3) the hyp1 pairs are useful for modeling lexical contrast. When included as ATTRACT constraints, these pairs lead to small but consistent gains across all three tasks (see also Tables 3-4).

Table 4: Task 3. Results ($F_1$) on the synonymy-antonymy evaluation set (Nguyen et al., 2017b).

The reported high score on SimLex of 0.791 is the first correlation score moving beyond mean human performance on the dataset (0.779), thus questioning the further usability of the benchmark in semantic modeling evaluation. The gain on SimVerb is even more substantial: from the previous high score of 0.674 (Mrkšić et al., 2017) to 0.770. The difference is again attributed to the use of high-quality constraints: Mrkšić et al. (2017) relied on a noisier and smaller set from BabelNet, verifying the importance of guiding specialisation by the correct choice of constraints. In short, the specialisation model simply encodes the provided external knowledge into the input vector space, and as such it is critically tied to the constraints.

Task 2: Antonymy Detection. A summary of the results is provided in Table 3. The results suggest that antonymous REPEL constraints are more beneficial for this task, which is easily explained by the nature of the task, but the synergistic effect is again observed: both types of constraints are essential to boost the scores. The best performing configuration of constraints outperforms two strong baselines (Zhang et al., 2014; Ono et al., 2015) which also rely on the same external lexical knowledge (minus hyp1 pairs). Importantly, the results also suggest that the specialisation model indeed learns useful relationships in the specialised space beyond a simple baseline model that lookups into constraints: large gains over this baseline are reported with a variety of configurations. Distributional SGNS-GN vectors coalesce antonymy and synonymy: as a consequence, they are not a competitive baseline in any of the three evaluation tasks.
The model which uses a large set of ANTEXP again cannot match performance of the model which relies on the original ANT. We see this as an interesting finding which suggests that the massive expansion of lexical constraints decreases the strength of originally provided word relationships, which were hand-crafted by linguistic experts.

Task 3: Synonymy vs. Antonymy. A summary of the results with strongest baselines from prior work is provided in Table 4: specialisation again outperforms the competitors. The score differences between best-performing configurations are not as pronounced as in the other two tasks: we attribute this to the reduced task complexity. However, the results again indicate that: 1) both types of constraints are important for distinguishing between the coalesced relations of synonymy and antonymy, with the synergistic effect again observed; 2) the noisy and large ANTEXP set of antonyms falls short of the smaller, more accurate ANT set; and 3) the same configuration as in the two other tasks (AC: SYN+HYP1, RC: ANT) again leads to peak performance.

5 Conclusion

We have demonstrated that post-processing specialisation models serve as a powerful tool for injecting lexical contrast knowledge into distributional word vector spaces. We have verified the hypothesis that a careful selection of external constraints is crucial for guiding the specialisation by improving state-of-the-art scores on three standard tasks used for evaluation of lexical contrast modeling: detecting antonyms, distinguishing antonyms from synonyms, and word similarity.

The post-processing specialisation models such as ATTRACT–REPEL fine-tune only vectors of words present in the external constraints. In the follow-up work, we have proposed a method which can propagate the useful external signal also to the full vocabulary (Vulić et al., 2018), leading to additional gains with specialised vectors in downstream language understanding applications. In future work, we will further investigate the full-vocabulary specialisation approaches.

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References


Characters or Morphemes: How to Represent Words?

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Abstract

In this paper, we investigate the effects of using subword information in representation learning. We argue that using syntactic subword units effects the quality of the word representations positively. We introduce a morpheme-based model and compare it against to word-based, character-based, and character n-gram level models. Our model takes a list of candidate segmentations of a word and learns the representation of the word based on different segmentations that are weighted by an attention mechanism. We performed experiments on Turkish as a morphologically rich language and English with a comparably poorer morphology. The results show that morpheme-based models are better at learning word representations of morphologically complex languages compared to character-based and character n-gram level models since the morphemes help to incorporate more syntactic knowledge in learning, that makes morpheme-based models better at syntactic tasks.

1 Introduction

The distributional hypothesis of Harris (1954) has been used to motivate work on vector space models to learn word representations. Deep learning models learn another kind of vector space model for building word representations, which shows superior performance in representing words.

Although deep neural networks have been very successful in representing words via such vectors, those models have not been very successful at estimating the representations of rare words since they do not appear often enough to allow us to collect reliable statistics about their context. Morphologically complex words are also rare by definition. Cao and Rei (2016) state that a word like unbelievableness does not exist in the first 17 million words of Wikipedia. Some methods have been proposed to deal with the sparsity issue in learning word representations. One approach is to utilize the subword information such as characters, character n-grams, or morphemes rather than learning distinct word representations without considering the inner structure of words.

Character-based models usually learn better word representations compared to word-based models since they capture the regularities inside the words so that it mitigates the sparsity in representation learning. However, those models learn the representations through the characters that do not correspond to a syntactic or semantic unit. In Turkish, two words can have similar word representations under a character-based model just because of their common suffixes. For example, character-based models such as (Bojanowski et al., 2017) generate similar word representations for words that have common character n-grams such as kitaplardan (from the books) and kasaplardan (from the butchers) (where lar and dan are suffixes, kitap and kasap are the roots) although the two words are semantically not related at all.

Another problem we observed for the character-based models is that such models estimate distant representations for words that are semantically related but involve different forms of the same morpheme so called allomorphs. This is one of the consequences of vowel harmony in some languages like Turkish. We observed this through several semantic similarity tasks performed on semantically similar but orthographically different words by using the word representations obtained from character n-gram level models such as fasttext (Bojanowski et al., 2017). For example, Turkish words such as mavililerinki (of the ones with
the blue color) and sarılıın (of the ones with the yellow color) with allomorphs li and li; ler and lar; in and in are asserted to be distant from each other in regard to their word representations under a character n-gram level model such as fasttext (Bojanowski et al., 2017), although the two words are semantically similar and both referring to colors.

In this paper, we argue that learning word representations through morphemes rather than characters lead to more accurate word vectors especially in morphologically complex languages. Such character-based models are strongly affected by the orthographic commonness of words, that governs orthographically similar words to have similar word representations.

We introduce a model to learn morpheme and word representations especially for morphologically very complex words without using an external supervised morphological segmentation system. Instead, we use an unsupervised segmentation model to initialize our model with a list of candidate morphological segmentations of each word in the training data. We do not provide a single segmentation per word like others (Botha and Blunsom, 2014; Qiu et al., 2014), but instead we provide a list of potential segmentations of each word. Therefore, our model relaxes the requirement of an external segmentation system in morpheme-based representation learning. To our knowledge, this will be the first attempt in co-learning of morpheme representations and word representations in an unsupervised framework without assuming a single morphological segmentation per word.

Our model is mostly similar to that of Lazaridou et al. (2013) and Botha and Blunsom (2014) since we also aim to learn morpheme and word representations. Our model is akin to that of Pinter et al. (2017) from the training perspective since they infer the out-of-vocabulary word embeddings from pre-trained word embeddings. Here, we also try to mimic the word2vec (Mikolov et al., 2013) embeddings (i.e., that are the expected outputs of the model) to learn the rare word representations with a complex morphology.

Our model shows some architectural similarities to that of Cao and Rei (2016). Both models use the attention mechanism to up-weight the correct morphological segmentation of a word. However, their model is character-based and our model is morpheme-based where different segmentations of each word contribute to the resulting vector. It should be noted that our main concern is to investigate what character-based models cannot learn that the morpheme-based models learn. As for the experimental setting, we have chosen Turkish language that has a complex morphology and severe allomorphy.

The results show that a morpheme-based model is better at estimating word representations of morphologically complex words (with at least 2-3 suffixes) compared to other word-based and character-based models. We present experimental results on Turkish as an agglutinative language and English as a morphologically poor language.

2 Related Work

Classical word representation models such as word2vec (Mikolov et al., 2013) have been successful in learning word representations for frequent words. Since these classical models are based on collecting contextual information in a very large corpus, they estimate deficient word representations for rare words due to insufficient contextual information. This has a negative consequence in some natural language processing tasks that make use of the word representations.

One approach to overcome this deficiency in estimating rare word representations is to apply compositional methods. Each word comprises of different subword units, such as characters, character n-grams, or morphemes. Lazaridou et al. (2013) apply compositional methods by having the stem and affix representations in order to estimate the distributional representation of morphologically complex words. Bojanowski et al. (2017) introduce an extension to word2vec (Mikolov et al., 2013) by representing each word in terms of the vector representations of its n-grams, which was earlier applied by Schütze (1993) that learns the representations of fourgrams by applying singular value decomposition (SVD). Analogously, Alexandrescu and Kirchhoff (2006) represent each character n-gram with a vector representation and words are estimated by the summation of the subword representations. Their results show that compositional methods that are originally proposed for estimating the meaning of phrases can also be used for estimating the meaning of a word by combining the information coming from different subword units. Botha and Blunsom (2014) introduce
Figure 1: The neural network architecture of the morph2vec model.

a log-bilinear language model that integrates morphology with compositional methods, that is applied to translation task for morphologically rich languages.

Compositional models that use character-level features show that the representations of rare words can be estimated more accurately (in both semantic and syntactic tasks) than the word-based models since the character-level models share more features across different words that helps to mitigate sparsity. Cotterell and Schütze (2015) encode morphological tags within word embeddings by using a log-bilinear model, thereby leading morphologically similar words to have closer word representations in the embedding space. Luong et al. (2013) learn word representations based on morphemes that are obtained from an external morphological segmentation system. Collobert et al. (2011) enhance word vectors with some character-level features such as capitalization. Bhatia et al. (2016) incorporate morphological information as a prior distribution to improve word embeddings. They use Morfessor (Creutz and Lagus, 2002) as an external morphological segmentation system to extract the inner structure of words.

3 The morph2vec Model

In our morpheme-based model, a word is encoded by a sequence of morphemes. Each word \( w_{s_i} \) with a particular morphological segmentation \( s_i \) is represented by a list of morphemes \( m = \{m_0, m_1, \ldots, m_n\} \) as follows:

\[
 w_{s_i} = m_0m_1 \ldots m_n
\]

We assume that the correct morphological segmentation of a word is not known a priori by assuming a completely unsupervised learning model. We use an unsupervised neural segmentation algorithm (Üstün and Can, 2016) that generates a list of candidate segmentations for a given word (see Section 4 for the details).

Each distinct morpheme is defined by a column vector in a morpheme embedding matrix \( W_m \in \mathbb{R}^{d_{\text{morph}} \times |M|} \) where \( d_{\text{morph}} \) is the vector dimension for the morphemes and \( M \) is the set of all pseudo morphemes.

Word representations are coupled with a particular morphological segmentation of each word. In other words, each segmentation of a single word has its own representation. Word representation for each particular segmentation is learned by a sequential function \( f \) that takes a sequence of morphemes and generates the word representation with a dimension of \( d_{\text{word}} \). The word embedding that is to be estimated compositionally via its morphemes that belong to segmentation \( s_i \) is denoted by \( v_{s_i} \) and estimated by a function \( f \) as follows:

\[
 v_{s_i} = f(w_{s_i}) = f(v_{m_0}, v_{m_1}, \ldots, v_{m_n})
\]  

where \( v_{m_0} \) denotes the vector of \( m_0 \).

We use bidirectional LSTMs (Bi-LSTM) (Hochreiter and Schmidhuber, 1997) to estimate a trainable function \( f \) in our neural network architecture that is illustrated in Figure 1. In the forward LSTMs, morphemes from the beginning till the end of the word are given sequentially, whereas in the backward LSTMs, morphemes from the end till the beginning of the word are given in the reverse order. Each output of Bi-LSTM which is the concatenation of the outputs of the forward and backward LSTMs represents a particular segmentation of a given word.

Therefore, we train the model with a list of potential segmentations of each word in training data. Since a word is represented by different morpheme sequences that refer to different segmentations of the same word, we use an attention model over these sequences that are learned by the Bi-LSTMs. Attention model learns a weight \( \alpha_i \) for each segmentation, such that \( \sum_i \alpha_i = 1 \) where \( S_w \) denotes all potential morphological segmentations of \( w \). The final word representation is the
weighted sum of the embeddings of all candidate segmentations:

\[ f_{\text{att}}(w) = \sum_i \alpha_i v_{s_i} \]  

(2)

where \( v_{s_i} \) is the vector for segmentation \( s_i \) that is the output of a Bi-LSTM. The weight \( \alpha_i \) is estimated as follows (Bahdanau et al., 2014):

\[ \alpha_i = \frac{\exp(v_i^T \tanh(W \cdot v_{s_i}))}{\sum_j \exp(v_j^T \tanh(W \cdot v_{s_j}))} \]  

(3)

Here, a feed-forward layer is used with a softmax function that is applied over the outputs of Bi-LSTMs. \( W \cdot v_{s_i} \) denotes the corresponding column in the weight matrix of the feed-forward layer in the attention.

For training, we use the pre-trained word2vec (Mikolov et al., 2013) vectors in order to minimize the cost between the learned word representations. Since all possible morpheme boundaries are not generated in training, if the segmentation algorithm generates an unknown morpheme in testing, the representation for that word is preserved especially through inflection involving the unknown suffix.

### Neural Morphological Segmentation

Although it is possible to train the model by providing all the potential segmentations of each word, we utilize an unsupervised segmentation algorithm to make the model computationally more efficient by reducing the search space. The segmentation algorithm is based on the neural model by Üstün and Can (2016) that uses the semantic similarity between substrings of a word to detect the potential morpheme boundaries. This algorithm is based on the idea that the meaning of a word is preserved especially through inflection and it benefits from the word representations to utilize this preservation. The parent-child relations such as (respect,respectful) are defined similar to that of Narasimhan et al. (2015).

The algorithm begins by generating all possible segmentations where there are at most \( K \) segments\(^1\) (see Table 1). Then, the algorithm checks the semantic similarity at each split point (between the parent and its child) whether it is greater than a threshold\(^2\). If the condition is satisfied for all split points in a segmentation, the segmentation is added to the segmentations list that will be passed to a Bi-LSTM. Figure 2 illustrates an example for the segmentation algorithm on the Turkish word abalarımın (of my cars). \# denotes a function that takes two words and returns true if the cosine similarity between two substrings is above the threshold value. The cosine similarities between the substrings of the word abalarımın are given in Table 2.

![Diagram](image)

Table 1: Some candidate segmentations of the Turkish word abalarımın. The bold one is the correct segmentation for this word.

<table>
<thead>
<tr>
<th>Parent</th>
<th>Child</th>
<th>Sim.</th>
</tr>
</thead>
<tbody>
<tr>
<td>ababa</td>
<td>-lar-ım-ın</td>
<td>0.41</td>
</tr>
<tr>
<td>abalar</td>
<td>-arabalarım-ın</td>
<td>0.65</td>
</tr>
<tr>
<td>abalarım</td>
<td>-arabalarım-ın</td>
<td>0.26</td>
</tr>
</tbody>
</table>

Table 2: The cosine similarities between the substrings (parent-child) of the Turkish word abalarımın (of my cars). 0.25 is assigned for the cosine similarity threshold and only the splits above the threshold are listed.

\[ s_0 \quad \text{ababa - lar - im - in} \]
\[ s_1 \quad \text{abalar - lar - im - in} \]
\[ s_2 \quad \text{abalar - im - in} \]
\[ s_3 \quad \text{abalar - lar - im - in} \]

\^1\(^K\) is defined as 4 in all experiments.

\^2\(^\text{The threshold is assigned 0.25 in all experiments.}\)
representation, we use an external supervised segmentation system for only testing purposes. Another reason is that due to incorrect segmentations suggested by the unsupervised segmentation algorithm, two words (semantically related) involving the same set of suffixes cannot benefit from the syntactic similarity and therefore the representations of those words might diverge in testing.

5 Experiments

We performed several experiments to assess the quality of our morpheme and word embeddings. We did experiments on Turkish as a highly agglutinative language with a very complex morphology and English with a comparably poor morphology.

5.1 Experimental Setting

In all experiments, morpheme vectors have a dimension of $d_{\text{morph}} = 75$, while the forward and backward LSTMs have a dimension of $d_{\text{LSTM}} = 300$. Since the output of the Bi-LSTMs is the concatenation of the forward and backward LSTMs, the Bi-LSTM output has a dimensionality of $d_{\text{Bi-LSTM}} = 600$. The output of the Bi-LSTMs is reduced to half after feeding the output through a feed-forward layer that results with a word vector dimension of $d_{\text{word}} = 300$. Our model is implemented in Keras, and publicly available.

For the pre-trained word vectors, we used the word vectors of dimension 300 that were obtained by training word2vec (Mikolov et al., 2013). For Turkish, we trained word2vec on Boun corpus (Sak et al., 2008) that contains 361 million word tokens. For English, we used the Google’s pre-trained word2vec model that was trained on 100 billion words with a vocabulary size of 3M. For training of our model, we used the most frequent 200K words from the pre-trained vocabularies to filter out the noise for both languages.

In order to compare the quality of our embeddings against the embeddings obtained from character n-gram level model fasttext (Bojanowski et al., 2017), we used the pre-trained word vectors trained on Wikipedia (Bojanowski et al., 2017) and we used the Google’s pre-trained word vectors. In order to compare our model with the character-based model by Cao and Rei (2016), we used Text8 corpus.

<table>
<thead>
<tr>
<th>Model</th>
<th>en</th>
<th>tr</th>
</tr>
</thead>
<tbody>
<tr>
<td>word2vec (Mikolov et al., 2013)</td>
<td>0.69</td>
<td>0.483</td>
</tr>
<tr>
<td>fasttext (Bojanowski et al., 2017)</td>
<td><strong>0.71</strong></td>
<td>0.208</td>
</tr>
<tr>
<td>morph2vec</td>
<td>0.38</td>
<td><strong>0.529</strong></td>
</tr>
</tbody>
</table>

Table 3: The comparison of the Spearman correlation between the human judgments and the word similarities obtained by computing the cosine similarity between the learned word embeddings for English and Turkish.

Only for testing reasons, we used PC-KIMMO (Koskenniemi, 1984) for English and the two-level Turkish morphology (Akin and Akin, 2007) for Turkish in order to segment test sets to obtain the actual morphemes for generating word representations from the morpheme vectors that are learned in a fully unsupervised setting. Unsupervised segmentation system also could be used for the evaluation step, but we wanted to minimize the effect of incorrect segmentations to be able to evaluate the embeddings properly. Yet, we discuss the effect of the supervised vs unsupervised segmentations in Section 5.5.

We did only intrinsic evaluation with a set of experiments that assess the quality of the word and morpheme representations.

5.2 Evaluation of Word Representations: Word Similarity Results

In order to evaluate the quality of the word vectors, we did experiments on a list of word pairs. We computed the cosine similarity between the learned vectors of each word pair and compared the similarity scores against to human judgments.

We used the Set 2 in WordSim353 dataset (Finkelstein et al., 2001) for the semantic similarity experiments that already involves the human judgment scores from 1 to 10 for 200 English word pairs. Since there is no available word-pair list for Turkish, we prepared WordSimTr that involves 138 word pairs and asked 15 human annotators to judge how similar two words are on a fixed scale from 1 to 10 where 1 shows a poor semantic similarity between the two words. Our Turkish word pair list involves two groups of words. The first group involves 81 semantically similar words that have at least two suffixes (possibly allomorphs). An example pair is televizyonlarda (on the televi-
Table 4: The comparison of the Spearman correlation between the human judgments and the word similarities obtained by computing the cosine similarity between the learned word embeddings for English on Text8 corpus.

<table>
<thead>
<tr>
<th>Model</th>
<th>WordSim353</th>
<th>RW</th>
</tr>
</thead>
<tbody>
<tr>
<td>char2vec (Cao and Rei, 2016)</td>
<td>0.345</td>
<td>0.284</td>
</tr>
<tr>
<td>morph2vec</td>
<td>0.386</td>
<td>0.297</td>
</tr>
</tbody>
</table>

We performed experiments for the analogy task in order to test whether the suffixes make a linear numerical change on the word vectors in the embedding space. The analogy experiments are usually performed for a triple of words such that A is to B so C is to ?, where A-B+C is expected to be equal to the questioned word. The analogy can be syntactical such as cat is to meow, so dog is to bark, or syntactic such as go is to gone, so have is to had.

Here, we tested only the syntactic analogy on a list of word tuples since our focus is especially morphologically complex languages. For English, we used the syntactic relations section provided in the Google analogy dataset (Mikolov et al., 2013) that involves 10675 questions. Since there is no analogy dataset for Turkish, we prepared a Turkish analogy set SynAnalogyTr with 206 syntactic questions that involves inflected word forms. The syntactic word tuples are judged by 40 human annotators in a scale from 1 to 10, where 1 shows a weak word analogy. Most words involve more than one suffix to test the morphological regularity in the analogy task.

The results are given in Table 6 and Table 7 for English and Turkish. The results show that our model outperforms both word2vec (Mikolov et al., 2013) and fasttext (Bojanowski et al., 2017) on both Turkish and English languages. Additionally, some examples to analogy results are given in Table 9 and the nearest neighbors of the Turkish word kitap-lar-dan-miş (it was from the books) are given in Table 8.

5.4 Evaluation of Morpheme Representations: Allomorph Results

In addition to the evaluation of the word vectors, we also evaluated the morpheme vectors that are the input embeddings to the neural network to be estimated during training. In order to evaluate how well our morpheme vectors represent the morphemes, we used the allomorphs. Allomorphs can be considered as true synonyms as they convey the same meaning with each other but with a different orthography.
In Turkish, there is a common use of allomorphs due to the vowel and consonant harmony in the language. For example, the morpheme ُل (dî) has got 8 allomorphs one of which is chosen depending on the last vowel and the consonant in the word, that are ُل (dî), ُع (du), ُئ (di), ُت (ti), ُج (tu), ُغ (ti), and ُل (ti). For example, -ت (the past tense of the third person singular) is chosen, for the verb ُت (git-(mek) (to go), whereas ُع is chosen for the verb ُت (solu-(mak) (to breathe).

We prepared a Turkish dataset that involves 108 morphemes that are allomorphs of 33 unique morpheme types including tense and case markers as well as derivations. For the evaluation of allomorphs, we used the MAP metric that is often used in information retrieval tasks. For each allomorph set in the data, we calculated the MAP@k where k is the number of allomorphs for the given morpheme. If the allomorph of a morpheme exists in the k nearest neighbours, then it is regarded as correct, otherwise it is incorrect. We averaged the MAP@k scores for all allomorph sets. The results are given in Table 10. Our model can learn the morpheme representations better than fasttext (Bojanowski et al., 2017) since allomorphs in our model are closer to each other in the embedding space compared to fasttext. Some of the allomorphs obtained from our model and fasttext (Bojanowski et al., 2017) are given in Table 11. As seen on the table, our model can capture the allomorphs better than fasttext (Bojanowski et al., 2017). Additionally, all Turkish allomorphs learned by our model are given in Figure 3. As can be seen from the figure, the allomorphs fall into similar regions in the vector space. Apart from some infrequent morphemes, the rest has similar representations.

### 5.5 The Effect of Supervision

In our experiments, the model training is performed in a fully unsupervised setting in terms of...
Table 9: Example Turkish analogy questions and the cosine similarities between the expected words and the learned word representations obtained from morph2vec, word2vec and fasttext.

<table>
<thead>
<tr>
<th>Word 1</th>
<th>Word 2</th>
<th>Word 3</th>
<th>Expected</th>
<th>morph2vec</th>
<th>word2vec</th>
<th>fasttext</th>
</tr>
</thead>
<tbody>
<tr>
<td>gel-dim</td>
<td>gel-me-dim</td>
<td>duy-dum</td>
<td>duy-ma-dim</td>
<td>0.80</td>
<td>0.43</td>
<td>0.63</td>
</tr>
<tr>
<td>(I came)</td>
<td>(I did not come)</td>
<td>(I heard)</td>
<td>(I did not hear)</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>çöz-müş</td>
<td>çöz-müş-tü</td>
<td>bul-müş</td>
<td>bul-muş-tu</td>
<td>0.73</td>
<td>0.18</td>
<td>0.66</td>
</tr>
<tr>
<td>(she solved)</td>
<td>(she had solved)</td>
<td>(she found)</td>
<td>(she had found)</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>aç-mak</td>
<td>aç-il-mak</td>
<td>ört-mek</td>
<td>ört-un-mek</td>
<td>0.89</td>
<td>0.36</td>
<td>0.52</td>
</tr>
<tr>
<td>(to open)</td>
<td>(to be opened)</td>
<td>(to cover)</td>
<td>(to cover himself)</td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

Table 10: MAP scores for the allomorph coverage in fasttext (Bojanowski et al., 2017) and the morph2vec.

<table>
<thead>
<tr>
<th>Model</th>
<th>MAP</th>
</tr>
</thead>
<tbody>
<tr>
<td>fasttext (Bojanowski et al., 2017)</td>
<td>0.504</td>
</tr>
<tr>
<td>morph2vec</td>
<td>0.618</td>
</tr>
</tbody>
</table>

Table 11: Some allomorphs of the given morpheme on the left that are found by our model and fasttext (Bojanowski et al., 2017). The bold font indicates the non-allomorphs for the given morpheme type.

<table>
<thead>
<tr>
<th>morph</th>
<th>morph2vec</th>
<th>fasttext</th>
</tr>
</thead>
<tbody>
<tr>
<td>iyor</td>
<td>iyor iyor iyor</td>
<td>iyor iyor iyor</td>
</tr>
<tr>
<td>mı</td>
<td>mı mı mı</td>
<td>mı mı mı</td>
</tr>
<tr>
<td>di</td>
<td>di di tu di tu duk</td>
<td>di di di inttır inttır</td>
</tr>
<tr>
<td>muş</td>
<td>muşmuş</td>
<td>muş yor duk</td>
</tr>
</tbody>
</table>

6 Conclusion and Future Work

Recent work shows that character level models learn more representative word embeddings for rare words (including morphologically complex words) compared to word level models, which is a sign that incorporating subword information improves the word representations. However, in this paper, we argued that morpheme-based representation models can learn better word embeddings (especially for the syntactic tasks) since they incorporate the syntactic and semantic information through the morphemes better compared to character level models. We pointed to the poor representation of allomorphs in complex words where the character-level models estimate a low word similarity between semantically similar words with different forms of the same morpheme, i.e. allomorphs. Moreover, we pointed to the character level models that assign a high word similar-
We introduce a morpheme-based representation model that learns word embeddings through the morphemes that are obtained from a list of morphological segmentations for each word. Therefore, our work introduces the idea of releasing the need for using an external morphological segmentation system in such representation learning models that are based on subword information. Our morpheme-based model morph2vec learns better word representations for morphologically complex words compared to the word-based model word2vec (Mikolov et al., 2013), character-based model char2vec (Cao and Rei, 2016), and the character n-gram level model fasttext (Bojanowski et al., 2017). Our results are also competitive for the English language.

We leave other languages and experiments such as morphological segmentation task for the future work. Another goal is to perform extrinsic evaluation on a different task such as part-of-speech tagging using the learned word embeddings.

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References


Abstract

We propose a hierarchical model for sequential data that learns a tree on-the-fly, i.e., while reading the sequence. In the model, a recurrent network adapts its structure and reuses recurrent weights in a recursive manner. This creates adaptive skip-connections that ease the learning of long-term dependencies. The tree structure can either be inferred without supervision through reinforcement learning, or learned in a supervised manner. We provide preliminary experiments in a novel Math Expression Evaluation (MEE) task, which is explicitly crafted to have a hierarchical tree structure that can be used to study the effectiveness of our model. Additionally, we test our model in a well-known propositional logic and language modelling tasks. Experimental results show the potential of our approach.

1 Introduction

Many kinds of sequential data such as language or math expressions naturally come with a hierarchical structure. Sometimes the structure is hidden deep in the semantics of the sequence, like the syntax tree in natural language; Other times the structure is more explicit, as in math expressions, where the tree is determined by the parentheses.

Recurrent neural networks (RNNs) have shown tremendous success in modeling sequential data, such as natural language (Mikolov et al., 2010; Merity et al., 2017). However, RNNs process the observed data as a linear sequence of observations: the length of the computational path between any two words is a function of their position in the observed sequence, instead of their semantic or syntactic roles, leading to the appearance of difficult-to-learn long-term dependencies and stimulating research on strategies to deal with that (Bengio et al., 1994; El Hihi and Bengio, 1996; Hochreiter and Schmidhuber, 1997). Hierarchical, tree-like structures may alleviate this problem by creating shortcuts between distant inputs and by simulating compositionality of the sequence, compressing the sequence into higher-level abstractions. Models that use tree as prior knowledge (e.g. (Socher et al., 2013; Tai et al., 2015; Bowman et al., 2016)) have shown improved performances over sequential models, validating the value of tree structure. For example, TreeLSTM (Tai et al., 2015) learns a bottom-up encoder, but requires the model to have access to the entire sentence as well as its parse tree before encoding it, which limits its application in some cases, e.g., language modeling. There has been various efforts to learn the tree structure as a supervised training target (Dyer et al., 2016; Socher et al., 2010; Zhou et al., 2017; Zhang et al., 2015), which free the model from relying on an external parser.

More recent efforts learn the best tree structure without supervision, by minimizing the log likelihood of the observed corpus, or by optimizing over a downstream task (Williams et al., 2017). These models usually take advantage of a binary tree assumption on the inferred tree, which imposes restrictions on the flexibility of inferred tree structure, for example, Gumbel TreeLSTM (Choi et al., 2017; Yogatama et al., 2016).

We propose a model that reads sequences using a hierarchical, tree-structured process (Fig. 1): it creates a tree on-the-fly, in a top-down fashion. Our model sits in between fully recursive models that have access to the whole sequence, such as
TreeLSTMs (Tai et al., 2015), and vanilla recurrent models that “flatten” input sequence, such as LSTMs (Schmidhuber, 1992). At each time-step in the sequence, the model chooses either to create a new sub-tree (split), to return and merge information into the parent node (merge), or to predict the next word in the sequence (recur). On split, a new sub-tree is created which takes control on which operation to perform. Merge operations end the current computation and return a representation of the current sub-tree to the parent node, which composes it with the previously available information on the same level. Recurrent operations use information from the siblings to perform predictions. Operations at every level in the tree use shared weights, thus sharing the recursive nature of TreeLSTMs. In contrast to TreeLSTMs however, the tree is created on-the-fly, which establishes skip-connections with previous tokens in the sequence and forms compositional representations of the past. The branching decisions can either be trained through supervised learning, by providing the true branching signals, or by policy gradients (Williams, 1992) which maximizes the log-likelihood of the observed sequence. As opposed to previous models, these three operations constantly change the structure of the model in an online manner.

Experimental evaluation aims to analyze various aspects of the model such as: how does the model generalizes on sequences of different lengths than those seen during training? how hard is the tree learning problem? To answer those questions, we propose a novel multi-operation math expression evaluation (MEE) dataset, with a standard set of tasks with varying levels of difficulty, where the difficulty scales up with respect to the length of the sequence.

2 Model

Similar to a standard RNN, our model modifies a hidden state $h_t$ for each step of the input sequence $x = \{x_1, \ldots, x_N\}$ by means of split, merge and recurrent operations. Denote the sequence of operations by $z = \{z_1, \ldots, z_L\}$, where $L$ may be greater than the number of tokens $N$ since only recurrent operations consume input tokens (see Fig. 1). Each operation is parametrized by a different “cell”. A policy network controls which operation to perform during sequence generation.

**split (S)** The split cell creates a sub-tree by taking the previous state $h_{t-1}$ as input and generating two outputs $h_t$ and $h_{down}$. $h_t$ is used for further computation, while $h_{down}$ (the small blue rectangle in Fig. 1(d)) is pushed into a stack for future use. In our model, $h_{down} = F_1(h_{t-1}, x_t)$ and $h_t = F_2(h_{t-1}, x_t)$ where the $F_1$ and $F_2$ are LSTM units (Hochreiter and Schmidhuber, 1997), and $x_t$ is the current input.

**recurrent (R)** This cell is a standard LSTM unit that takes as input the previous state $h_{t-1}$ and the current token $x_t$, and outputs the hidden state $h_t$, which will be used to predict the next output $\hat{x}_{t+1}$. After application of this cell, the counter $t$ is incremented and input $x_t$ is consumed.
merge (M) The merge cell closes a sub-tree and returns control to its parent node. It does so by merging the previous hidden state $h_{t-1}$ with the top of the stack $h_{down}$ into a new hidden state $h_m = MLP(h_{t-1}, h_{down})$ (Fig. 1(c)). $h_m$ is then used as input to another LSTM unit ($F_3$) to yield $h_t = F_3(x_t, h_m)$, the new hidden state of the overall network. Intuitively, $h_{t-1}$ summarizes the contents within the sub-tree. This is merged with information obtained before the model entered the sub-tree $h_{down}$ into the new state $h_t$.

Policy Network We consider the decision at each timestep $z_t \in \{S, M, R\}$ as a categorical variable sampled from a policy network $p_z$, conditioned on the hidden state $h_t$ and the input embedding $e_t$ of the current input $x_t$. In the supervised setting, $p_z(z_t|e_t, h_t)$ is trained by maximizing the likelihood of the true branching labels, while in the unsupervised setting, we resort to the REINFORCE algorithm using $-\log p(y_t|C)$ as a reward, where $y_t$ is the task target (i.e. the next word in language modeling), and $C$ is the representation learnt by the model up until time $t$.

3 Experimental Results

We conduct our experiments on a math induction task, a propositional logic inference task (Bowman et al., 2016) and language modelling. First of all, we aim to investigate whether a) our hierarchical model may help in tasks that explicitly exhibit hierarchical structure, and then b) whether the trees learned without supervision correspond to the ground-truth trees, c) how our model fare with respect to hierarchical models that have access to the whole sequence with a pre-determined tree structure and finally, d) are there any limitations for models that are not capable of learning hierarchical structures on-the-fly.

3.1 Math Induction

Our math expression evaluation dataset (MEE) consists of parenthesized mathematical expressions and their corresponding evaluations. The math expressions contain bracketing symbols ("("), four different kinds of operations, "+,-,*,%", where "/" is the modulo operation, and digits from 0 to 9. The “length” of an expression is the number of operations in the expression and its result is restricted to be a positive, two-digit integer (Table 1). We randomly generate expressions of different lengths and for each length the resulting sub-set is divided into 100,000, 10,000 and 10,000 expressions as training, valid and test sets. We make sure that there is no overlap between the splits and every expression is made unique across the whole dataset.

We use an encoder-decoder approach where the encoder reads the characters in the expression and produces the encoding as input to the decoder, which in turn sequentially generates 2 digits as the predicted value. We experiment on various encoders, including our model, and compare their performances. We use the same decoder architecture to ensure a fair comparison. The output of the encoder is provided as the initial hidden state of the decoder LSTM. To test the generalization of our model, for all the experiments shown in this subsection, we train the model on expressions of length 4 and 5, and evaluate on expressions of length 4 to 7 in the test sets.

For this task, our baseline is a simple LSTM encoder (which corresponds to our model with only recur operations). We compare two versions of our RRNet encoder. In the supervised setting, we force the model to split and merge when it reads "(" and ")", respectively, and recur otherwise. This gives us an idea on how well the model would perform if it had access to the ground-truth tree. In the unsupervised setting, we learn the tree using policy gradient, where the reward is the accuracy of the math result prediction. The results are in Table 2. The supervised RRNet yields the best performance showing that (a) it is important to exploit the hierarchical structure of the observed data, corroborating previous work (Williams et al., 2017), and (b) our model is effective at capturing that information. The unsupervised RRNet model also outperforms the baseline LSTM: the model learn to exploit branching operations to achieve better performance. We observe that the trees produced by the model do not correspond to the ground-truth trees. In order to assess whether the additional parameters of split and merge operations, rather than the learned tree

<table>
<thead>
<tr>
<th>Length</th>
<th>Expression</th>
<th>Value</th>
</tr>
</thead>
<tbody>
<tr>
<td>4</td>
<td>((9+(2+6)+((1*3)))</td>
<td>20</td>
</tr>
<tr>
<td>5</td>
<td>(((7-2)%((3%1)+6))*9)</td>
<td>45</td>
</tr>
<tr>
<td>6</td>
<td>(((3-0)+(7-6))*(0+9))-7</td>
<td>29</td>
</tr>
<tr>
<td>7</td>
<td>((4*(6+(7*(2*8))))%9+(3+7)))</td>
<td>16</td>
</tr>
</tbody>
</table>

Table 1: Sample expressions from MEE dataset
Table 2: Prediction accuracy on MEE dataset.

<table>
<thead>
<tr>
<th>Model</th>
<th>Train</th>
<th>L=4</th>
<th>L=5</th>
<th>L=6</th>
<th>L=7</th>
</tr>
</thead>
<tbody>
<tr>
<td>LSTM</td>
<td>75.80</td>
<td>81.32</td>
<td>67.65</td>
<td>52.70</td>
<td>41.35</td>
</tr>
<tr>
<td>Uns RRNet</td>
<td>86.00</td>
<td>89.42</td>
<td>77.96</td>
<td>61.34</td>
<td>50.46</td>
</tr>
<tr>
<td>Sup RRNet</td>
<td>93.70</td>
<td>93.28</td>
<td>86.69</td>
<td>79.09</td>
<td>72.70</td>
</tr>
</tbody>
</table>

Figure 2: Test accuracy of the models, trained on sequences of length $\leq 6$ in logic data. The horizontal axis indicates the length of the sequence, and the vertical axis indicates the accuracy of model’s performance on the corresponding test set.

structure, produce better results, we measured the performance of our model trained with “random” deterministic policies (associating each of the input characters to either a split, merge or recur operation). We see that “random” policies perform worse than a “learnt” policy on the task, effectively similar to the baseline LSTM. In turn, the model with the learnt policy underperforms the model trained with ground-truth trees. Even in this seemingly easy task, it has appeared difficult for the model to learn the optimal branching policy.

3.2 Logical inference

In the next task, we analyze performance on the artificial language as described in Bowman et al. (2015b). This language has six word types \{a, b, c, d, e, f\} and three logical operations \{or, and, not\}. There are seven mutually exclusive logical relations that describe the relationship between two sentences: entailment ($\sqsubset$, $\sqsubseteq$), equivalence ($\equiv$), exhaustive and non-exhaustive contradiction ($\wedge$, $\|$, |), and two types of semantic independence ($\#$, $\dashv$). The train/dev/test dataset ratios are set to 0.8/0.1/0.1 as described \(^1\) with the number of logical operations ranging from 1 to 12.

From Figure 2, we report the performance of our model when trained with ground-truth trees as input. It is encouraging to see that our recurrent-recursive encoder improves performance over Transformer (FAN) (Tran et al., 2018) and LSTM, especially for long sequences. The best performance on this dataset is given by TreeLSTM, which has access to the whole sequence (Bowman et al., 2015b) and does not encode sequences on-the-fly.

3.3 Language Modelling

In language modeling, architectures such as TreeLSTM aren’t directly applicable since their structure isn’t computed on-the-fly, while reading the sentence. We perform preliminary experiments using the Penn Treebank Corpus dataset (Marcus et al., 1993), which has a vocabulary of 10,000 unique words and 929k, 73k and 82k words in training, validation and test set respectively. Our cells use one layer and the hidden dimensionality is 350. Our model yields test perplexity of 107.28 as compared to the LSTM baseline which gets 113.4 (Dyer et al., 2016). This preliminary result shows that the endeavor to exploit explicit hierarchical structures for language modeling, although challenging, may be promising.

4 Final Considerations

In this work, we began exploring properties of a recurrent-recursive neural network architecture that learns to encode the sequence on-the-fly, i.e. while reading. We argued this may be an important feature for tasks such as language modeling. We additionally proposed a new mathematical expression evaluation dataset (MEE) as a toy problem for validating the performance of sequential models to learn from hierarchical data. We empirically observed that, in this task, our model performs better than a standard LSTM architecture with no explicit structure and also outperforms the baseline LSTM and FAN architectures on the propositional logic task.

We hope to further study the properties of this model by either more thorough architecture search (recurrent dropouts, layer norm, hyper-parameter sweeps), different variation of RL algorithms such

\(^1\)https://github.com/sleepinyourhat/vector-entailment
as deep Q-learning (Mnih et al., 2013) and employing this model on various other tasks such as SNLI (Bowman et al., 2015a) and semi-supervised parsing.

Acknowledgement

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Limitations of cross-lingual learning from image search

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Abstract

Cross-lingual representation learning is an important step in making NLP scale to all the world’s languages. Previous work on bilingual lexicon induction suggests that it is possible to learn cross-lingual representations of words based on similarities between images associated with these words. However, that work focused (almost exclusively) on the translation of nouns only. Here, we investigate whether the meaning of other parts-of-speech (POS), in particular adjectives and verbs, can be learned in the same way. Our experiments across five language pairs indicate that previous work does not scale to the problem of learning cross-lingual representations beyond simple nouns.

1 Introduction

Typically, cross-lingual word representations are learned from word alignments, sentence alignments, from aligned, comparable documents (Levy et al., 2017), or from monolingual corpora using seed dictionaries (Ammar et al., 2016). However, for many languages such resources are not available.

Bergsma and Van Durme (2011) introduced an alternative idea, namely to learn bilingual representations from image data collected via web image search. The idea behind their approach is to represent words in a visual space and find valid translations between words based on similarities between their visual representations. Representations of words in the visual space are built by representing a word by a set of images that are associated with that word, i.e., the word is a semantic tag for the images in the set.

Kiela et al. (2015) improve performance for the same task using a feature representation extracted from convolutional networks. However, both works only consider nouns, leaving open the question of whether learning cross-lingual representations for other POS from images is possible.

In order to evaluate whether this work scales to verbs and adjectives, we compile wordlists containing these POS in several languages. We collect image sets for each image word and represent all words in a visual space. Then, we rank translations computing similarities between image sets and evaluate performance on this task.

Another field of research that exploits image data for NLP applications is the induction of multi-modal embeddings, i.e. semantic representations that are learned from textual and visual information jointly (Kiela et al., 2014; Hill and Korhonen, 2014; Kiela and Bottou, 2014; Lazaridou et al., 2015; Silberer et al., 2017; Kiela et al., 2016; Vulić et al., 2016). The work presented in our paper differs from these approaches, in that we do not use image data to improve semantic representations, but use images as a resource to learn cross-lingual representations. Even though lexicon induction from text resources might be more promising in terms of performance, we think that lexicon induction from visual data is worth exploring as it might give insights in the way that language is grounded in visual context.

1Recent work by Lample et al. (2018) introduces unsupervised bilingual lexicon induction from monolingual corpora, however, it was shown that this approach has important limitations (Søgaard et al., 2018).

2Kiela et al. (2016) induce English-Italian word translations from image data for the Simlex-999 dataset which contains adjectives and verbs, but they do not evaluate the performance for these POS compared to nouns.
1.1 Contributions

We evaluate the approaches by Bergsma and Van Durme (2011) and Kiela et al. (2015) on an extended data set, which apart from nouns includes both adjectives and verbs. Our results suggest that none of the approaches involving image data are directly applicable to learning cross-lingual representations for adjectives and verbs.

2 Data

Wordlists We combined 3 data sets of English words to compile the wordlists for our experiments: the original wordlist used by Kiela et al. (2015), the Simlex-999 data set of English word pairs (Hill et al., 2014) and the MEN data set (Bruni et al., 2014). Whereas the first wordlist contains only nouns, the latter two datasets contain words of three POS classes (nouns, adjectives and verbs). We collect all distinct words and translate the final wordlist into 5 languages (German, French, Russian, Italian, Spanish) using the Google translation API. Choosing the most frequent translation with the respective POS tag. Table 1 shows the POS distribution in the datasets.

<table>
<thead>
<tr>
<th></th>
<th>MEN</th>
<th>Simlex</th>
<th>Bergsma</th>
<th>Combined</th>
</tr>
</thead>
<tbody>
<tr>
<td>N</td>
<td>656</td>
<td>751</td>
<td>500</td>
<td>1406</td>
</tr>
<tr>
<td>V</td>
<td>38</td>
<td>170</td>
<td>0</td>
<td>206</td>
</tr>
<tr>
<td>A</td>
<td>57</td>
<td>107</td>
<td>0</td>
<td>159</td>
</tr>
</tbody>
</table>

Table 1: Distribution of POS tags in the datasets used to compile the final wordlist.

Image Data Sets We use the Google Custom Search API to represent each word in a wordlist by a set of images. We collect the first 50 jpeg images returned by the search engine when querying the words specifying the target language. This way, we compile image data sets for 6 languages. Figure 1 shows examples for images associated with a word in two languages.

3 Approach

The assumption underlying the approach is that semantically similar words in two languages are associated with similar images. Hence, in order to find the translation of a word, e.g. from English to German, we compare the images representing the English word with all the images representing German words, and pick as translation the German word represented by the most similar images. To compute similarities between images, we compute cosine similarities between their feature representations.

3.1 Convolutional Neural Network Feature Representations

Following Kiela et al. (2015), we compute convolutional neural network (CNN) feature representations using a model pre-trained on the ImageNet classification task (Russakovsky et al., 2015). For each image, we extract the pre-softmax layer representation of the CNN. Instead of an AlexNet (Krizhevsky et al., 2012) as used by Kiela et al. (2015), we use the Keras implementation of the VGG19 model as described in Simonyan and Zisserman (2014), which was shown to achieve similar performance for word representation tasks by Kiela et al. (2016). Using this model, we represent each image by a 4069-dimensional feature vector.

Similarities Between Individual Images Bergsma and Van Durme (2011) determine similarities between image sets based on similarities between all individual images. For each image in image set 1, the maximum similarity score for any image in image set 2 is computed. These maximum similarity scores are then either averaged (AVGMAX) or their maximum is taken (MAXMAX).

Similarities Between Aggregated Representations In addition to the above described methods, Kiela et al. (2015) generate an aggregated representation for each image set and then compute the similarity between image sets by computing the similarity between the aggregated representations. Aggregated representations for image sets are generated by either taking the component-wise average (CNN-MEAN) or the component-wise maximum (CNN-MAX) of all images in the set.

K-Nearest Neighbor For each image in an image set in language 1, we compute its nearest
neighbor across all image sets in language 2. Then, we find the image set in language 2 that contains the highest number of nearest neighbors. The image word is translated into the image word that is associated with that image 2 set. Ties between image sets containing an equivalent number of nearest neighbors are broken by computing the average distance between all members. We refer to the method as KNN. Whereas the other approaches described above provide a ranking of translations, this method determines only the one translation that is associated with the most similar image set.

Clustering Image Sets
As we expect the retrieved image sets for a word to contain images associated with different senses of the word, we first cluster images into $k$ clusters. This way, we hope to group images representing different word senses. Then, we apply the KNN method as described above. We refer to this method as KNN-C.

3.2 Evaluation Metrics

Ranking performance is evaluated by computing the Mean Reciprocal Rank (MRR) as $MRR = \frac{1}{M} \sum_{i=1}^{M} \frac{1}{\text{rank}(w_s, w_t)}$. $M$ is the number of words to be translated and $\text{rank}(w_s, w_t)$ is the position the correct translation $w_t$ for source word $w_s$ is ranked on.

In addition to MRR, we also evaluate the cross-lingual representations by means of precision at $k$ ($P@k$).

4 Experiments and Results

We run experiments for 5 language pairs English–German, English–Spanish, English–French, English–Russian and English–Italian. We evaluate the representations computed from image data and compare the different methods for similarity computation described in 3. For each English word, we rank all the words in the corresponding target languages based on similarities between image sets and evaluate the models’ ability to identify correct translations, i.e. to rank the correct translation on a position near the top. We compare 4 settings that differ in the set of English words that are translated. In the setting ALL, all English words in the wordlist are translated. NN, VB and ADJ refer to the settings where only nouns, verbs and adjectives are translated.

4.1 Results

Comparison of similarity computation methods for visual representations

Table 2 displays results averaged over all language pairs.\(^7\) First, comparing the different methods to compute similarities between image sets, $\text{AVGMAX}$ outperforms the other methods in almost all cases. Most importantly, we witness a very significant drop in performance when moving from nouns to verbs and adjectives. For verbs, we rarely pick the right translation based on the image-based word representations. This behavior applies across all methods for similarity computation. Further, we see small improvements if we cluster the image sets prior to applying the KNN method, which might indicate that the clustering helps in finding translations for polysemous words.

4.2 Analysis

If we try to learn translations from images, integrating verbs and adjectives into the dataset worsens results compared to a dataset that contains only nouns. One possible explanation is that images associated with verbs and adjectives are less suited to represent the meaning of a concept than images associated with nouns.

Kiela et al. (2015) suppose that lexicon induction via image similarity performs worse for

\(^7\)We also evaluate our visual representations on the set of 500 nouns used by Kiela et al. (2015), which results in $P@1=0.6$ and $MRR=0.63$ averaged over 5 language pairs for the $\text{AVGMAX}$ method.
datasets containing words that are more abstract. In order to approximate the degree of abstractness of a concept, they compute the image dispersion \(d\) for a word \(w\) as the average cosine distance between all image pairs in the image set \(\{i_j, \ldots, i_n\}\) associated with word \(w\) according to

\[
d(w) = \frac{2}{n(n-1)} \sum_{k<j \leq n} 1 - \frac{i_j \cdot i_k}{||i_j|| ||i_k||}
\]

In their analysis, Kiela et al. (2015) find that their model performs worse on datasets with a higher average image dispersion. Kiela et al. (2014) introduce a dispersion-based filtering approach for learning multi-modal representations of nouns. They show that the quality of their representations with respect to a monolingual word-similarity prediction task improves, if they include visual information only in cases where the dispersion of the visual data is low.

Computing the average image dispersion for our data across languages shows that image sets associated with verbs and adjectives have a higher average image dispersion than image sets associated with nouns (nouns: \(d = 0.60\), verbs: \(d = 0.68\), adjectives: \(d = 0.60\)).

Table 3 shows the image words associated with the image sets that have the highest and lowest dispersion values in the English image data. For nouns and adjectives, we observe that the words with lowest dispersion values express concrete concepts, whereas the words with highest dispersion values express more abstract concepts that can be displayed in many variants. Manually inspecting the dataset, we find e.g. that the images associated with the noun animal display many different animals, such as birds, dogs, etc, whereas the images for mug all show a prototypical mug.

Besides the dispersion values, we also analyze the number of word senses per POS using WordNet\(^8\). We find that the verbs in our dataset have a higher average number of word senses \((n = 9.18)\) than the adjectives \((n = 6.88)\) and the nouns \((n = 5.08)\). That we get worst results for the words with highest number of different word senses is in agreement with Gerz et al. (2016), who find that in a monolingual word similarity prediction task, models perform worse for verbs with more different senses than for less polysemous verbs.

<table>
<thead>
<tr>
<th>Lowest dispersion</th>
<th>Highest dispersion</th>
</tr>
</thead>
<tbody>
<tr>
<td>Word</td>
<td>(d)</td>
</tr>
<tr>
<td>mug</td>
<td>0.31</td>
</tr>
<tr>
<td>oscilloscope</td>
<td>0.32</td>
</tr>
<tr>
<td>padlock</td>
<td>0.33</td>
</tr>
<tr>
<td>vanish</td>
<td>0.43</td>
</tr>
<tr>
<td>shed</td>
<td>0.43</td>
</tr>
<tr>
<td>divide</td>
<td>0.47</td>
</tr>
<tr>
<td>yellow</td>
<td>0.39</td>
</tr>
<tr>
<td>white</td>
<td>0.40</td>
</tr>
<tr>
<td>fragile</td>
<td>0.43</td>
</tr>
</tbody>
</table>

Table 3: English image words associated with the image sets with highest and lowest dispersion scores \(d\).

5 Conclusion

We showed that existing work on learning cross-lingual word representations from images obtained via web image search does not scale to other POS than nouns. It is possible that training convolutional networks on different resources than ImageNet data will provide better features represent-

\(^8\)https://wordnet.princeton.edu/
ing verbs and adjectives. Finally, it would be interesting to extend the approach to multi-modal input, combining images and texts, e.g. from comparable corpora with images such as Wikipedia.

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Learning Semantic Textual Similarity from Conversations

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Abstract

We present a novel approach to learn representations for sentence-level semantic similarity using conversational data. Our method trains an unsupervised model to predict conversational responses. The resulting sentence embeddings perform well on the Semantic Textual Similarity (STS) Benchmark and SemEval 2017’s Community Question Answering (CQA) question similarity subtask. Performance is further improved by introducing multi-task training, combining conversational response prediction and natural language inference. Extensive experiments show the proposed model achieves the best performance among all neural models on the STS Benchmark and is competitive with the state-of-the-art feature engineered and mixed systems for both tasks.

1 Introduction

We propose a novel approach to sentence-level semantic similarity based on unsupervised learning from conversational data. We observe that semantically similar sentences have a similar distribution of potential conversational responses, and that a model trained to predict conversational responses should implicitly learn useful semantic representations. As illustrated in Figure 1, “How old are you?” and “What is your age?” are both questions about age, which can be answered by similar responses such as “I am 20 years old”. In contrast, “How are you?” and “How old are you?” use similar words but have different meanings and lead to different responses.

Deep learning models have been shown to predict conversational responses with increasingly good accuracy (Henderson et al., 2017; Kannan et al., 2016). The internal representations of such models resolve the semantics necessary to predict the correct response across a broad selection of input messages. Meaning similarity between sentences then can be obtained by comparing the sentence-level representations learned by such models. We follow this approach, and assess the quality of the resulting similarity scores on the Semantic Textual Similarity (STS) Benchmark (Cer et al., 2017) and a question similarity subtask from SemEval 2017’s Community Question Answering (CQA) evaluation. The STS benchmark scores sentence pairs based on their degree of meaning similarity. The Community Question Answering (CQA) subtask B (Nakov et al., 2017) ranks questions based on their similarity with a target question.

We first assess representations learned from unsupervised conversational input-response pairs. We then explore augmenting our model with multi-task training over a combination of unsupervised conversational response prediction and supervised training on Natural Language Inference (NLI) data, as training to NLI has been shown to independently yield useful general purpose representations (Conneau et al., 2017). Unsupervised training over conversational data yields represen-
Figure 2: The conversational response selection problem attempts to identify the correct response from a collection of candidate responses. We train using batch negatives with each candidate response serving as a positive example for one input and a negative sample for the remaining inputs.

2 Approach

This section describes the conversational learning task and our architecture for predicting conversational responses. We detail two encoding methods for converting sentences into sentence embeddings and describe multitask learning over conversational and NLI data.

2.1 Conversational Response Prediction

We formulate the conversational learning task as response prediction given an input (Kannan et al., 2016; Henderson et al., 2017). Following prior work, the prediction task is cast as a response selection problem. As shown in Figure 2, the model attempts to identify the correct response \( y \) from \( K - 1 \) randomly sampled alternatives.

2.2 Model Architecture

Our model architecture encodes input and response sentences into fixed-length vectors \( u \) and \( v \), respectively. The preference of an input described by \( u \) for a response described by \( v \) is scored by the dot product of the two vectors. The dot product scores are converted into probabilities using a softmax over the scores from all other candidate responses. Model parameters are trained to maximize the log-likelihood of the correct responses.

Figure 3 illustrates the input-response scoring model architecture. Tied parameters are used for the input and response encoders. In order to model the mapping between inputs and their expected responses, the response embeddings are passed through an additional feed-forward network to get the final response vector \( v' \) before computing the dot product with the input sentence embedding.\(^{1}\)

Training is performed using batches of \( K \) randomly shuffled input-response pairs. Within a batch, each response serves as the correct answer to its corresponding input and the incorrect response to the remaining \( K - 1 \) inputs in the batch. In the remaining sections, this architecture is referred to as the input-response model.

2.3 Encoders

Figure 4 illustrates the encoders we explore for obtaining sentence embeddings: DANs (Iyyer et al., 2015) and Transformer (Vaswani et al., 2017).\(^{2}\)

2.3.1 DAN

Deep averaging networks (DAN) compute sentence-level embeddings by first averaging word-level embeddings and then feeding the averaged representation to a deep neural network (DNN) (Iyyer et al., 2015). We provide our encoder with input embeddings for both words and bigrams in the sentence being encoded. This simple architecture has been found to outperform LSTMs on email response prediction (Henderson et al., 2017).

\(^{1}\)While feed-forward layers could have been added to the input encoder as well, early experiments suggested it was sufficient to add additional layers to only one of the encoders.

\(^{2}\)We tried other encoder architectures, notably LSTM (Hochreiter and Schmidhuber, 1997) and Bi-LSTM (Graves and Schmidhuber, 2005), but found they performed worse than transformer in preliminary experiments.
bigrams are learned during training of the input-response model. Our implementation sums the input embeddings and then divides by $\sqrt{n}$, where $n$ is the sentence length. The resulting vector is passed as input to the DNN.

The transformer encoder output is a variable-length sequence. We reduce it to fixed length by averaging across all sequence positions. Intuitively, this is similar to building a bag-of-words representation, except that the words have had a chance to interact with their contexts through the attention layers. In practice, we see that the learned attention masks focus largely on nearby words in the first layer, and attend to progressively more distant context in the higher layers.

2.4 Multitask Encoder

We anticipate that learning good semantic representations may benefit from the inclusion of multiple distinct tasks during training. Multiple tasks should improve the coverage of semantic phenomena that are critical to one task but less essential to another. We explore multitask models that use a shared encoder for learning conversational response prediction and natural language inference (NLI). The NLI data are from the Stanford Natural Language Inference (SNLI) (Bowman et al., 2015) corpus. The sentences are mostly non-conversational, providing a complementary learning signal.

Figure 5 illustrates the multitask model with SNLI. We keep the input-response model the same, and build another two encoders for SNLI pairs, sharing parameters with the input-response encoders. Following Conneau et al. (2017), we encode a sentence pair into vectors $u_1, u_2$ and construct a feature vector $(u_1, u_2, |u_1 - u_2|, u_1 \ast u_2)$. The feature vector is fed into a 3-way classifier consisting of a feedforward network culminating in a softmax layer. Following prior work, we use a single 512 unit hidden layer for our experiments.

3 Conversational Data

Our unsupervised model relies on structured conversational data. The data for our experiments are drawn from Reddit conversations spanning 2007 to 2016, extracted by Al-Rfou et al. (2016). This corpus contains 133 million posts and a total of 2.4 billion comments. The comments are mostly conversational and well structured, making it a good resource for training conversational models.

Figure 6 provides an example of a Reddit comment chain. Comment B is a child of comment A if comment B is a reply to comment A. We extract

3 $\sqrt{n}$ is one of TensorFlow’s built-in embedding combiners. The intuition behind dividing by $\sqrt{n}$ is as follows: We want our input embeddings to be sensitive to length. However, we also want to ensure that for short sequences the relative differences in the representations are not dominated by sentence length effects.

4 https://github.com/tensorflow/tensor2tensor
3.1 Model Configuration

Model configuration and hyperparameters are set based on prior experiments on Reddit response prediction and performance of the multi-task model on SNLI. All inputs are tokenized and normalized before being fed into model. For all experiments, we use SGD with a batch size of 128 and a learning rate of 0.01. The total training steps are 40 million steps for the Reddit model and 30 million steps for the Reddit+SNLI model. We adjust the batch size to 256 and learning rate to 0.001 after 30 million and 20 million steps for the Reddit and the Reddit+SNLI models, respectively. When training the multitask model, we initialize the shared parameters with a pretrained Reddit model. We employ a distributed training system with multiple workers, where 95% of workers are used to continue training the Reddit task and 5% of workers are used to train the SNLI task. We use a sentence embedding size of 500 in all experiments, and normalize sentence embeddings prior to use in subsequent network layers. The parameters were only lightly tuned to prevent overfitting on the SNLI task.

The encoder configurations are taken from the default parameters from previous work. For DAN, we employ a 3-layer DNN with layers containing 300, 300, and 500 hidden units. For the transformer encoder, our experiments make use of 6 attention layers (num_hidden_layers) and 8 attentions heads (num_heads). Within each attention layer, the feedforward network applied to each head has an input and output size of 512 (hidden_size) and makes use of a 2048 unit inner-layer (filter_size).

4 Experiments

We first evaluate the different encoders on the response prediction task. For the multitask models, we then examine their performance on SNLI. Finally, we evaluate the encoders on the STS Benchmark (Cer et al., 2017) and on SemEval 2017 Community Question Answering (CQA) subtask B (Nakov et al., 2017). We refer to the model trained over Reddit input-response pairs as Reddit and the multitask model as Reddit+SNLI.

4.1 Response Prediction

Following Henderson et al. (2017), we use precision at N (P@N) as an evaluation metric for the conversational response prediction task. Given an

<table>
<thead>
<tr>
<th>Encoder</th>
<th>P@1</th>
<th>P@3</th>
<th>P@10</th>
</tr>
</thead>
<tbody>
<tr>
<td>Transformer</td>
<td>65.7</td>
<td>78.7</td>
<td>89.8</td>
</tr>
<tr>
<td>DAN</td>
<td>56.1</td>
<td>70.2</td>
<td>83.6</td>
</tr>
</tbody>
</table>

Table 1: Precision at N (P@N) results on the Reddit response prediction test set for models built using the DAN and Transformer encoders. Models attempt to select the true response for an input against 99 randomly selected negatives.

/comments and their children to form the input-response pairs described above. Several rules are applied to filter out the noisy data. A comment is removed if any of the following conditions holds: number of characters \( \geq 350 \), percentage of alphabetic characters \( \leq 70\% \), starts with “https”, “/r/” or “@”, author’s name contains “bot”. The total number of extracted pairs is around 600 million.
Table 2: SNLI classification performance for the Reddit+SNLI model using the transformer encoder with reference evaluation numbers from prior work. We note that similar to InferSent, our goal is to use SNLI to obtain better sentence representations rather than achieving state-of-the-art performance on the SNLI task itself.

<table>
<thead>
<tr>
<th>Model</th>
<th>Accuracy</th>
</tr>
</thead>
<tbody>
<tr>
<td>Reddit+SNLI</td>
<td>84.1</td>
</tr>
<tr>
<td>InferSent</td>
<td>84.5</td>
</tr>
<tr>
<td>KIM Ensemble</td>
<td>89.0</td>
</tr>
<tr>
<td>Gumbel TreeLSTM</td>
<td>86.0</td>
</tr>
</tbody>
</table>

Table 2 shows the accuracy on the test set of the joint model and baselines. The multitask model achieves 84.1% accuracy and is close to the performance of InferSent. There are two significant differences between our model and prior work. First, the proposed model learns all model parameters from scratch, including the word embeddings. Due in part to the size of the SNLI training set, InferSent uses a large pre-trained word embedding model fit via GloVe (Pennington et al., 2014) on 840 billion tokens of web crawl data, which results in fewer out-of-vocabulary words. For our multitask model, the Reddit dataset is large enough that we do not necessarily require pre-trained word embeddings. However, it is possible the pre-trained GloVe embeddings provide slightly better performance on the SNLI task.\(^5\) Secondly, our multi-task model learns two tasks simultaneously, balancing performance between them, while InferSent only optimizes performance on SNLI. As will be presented below, our multi-task model performs better on STS. We suspect multi-task training both increases coverage of different language phenomena and acts as a regularizer across tasks that prevents the resulting sentence embeddings from overfitting any particular task, thus improving transfer performance to new tasks.\(^6\)

4.2 SNLI

SNLI (Bowman et al., 2015) annotates the inferential relationship between paired sentences as entailment, contradiction, or neutral. One sentence is entailed by another sentence if its meaning can be inferred from the other. Sentences contradict each other if the meaning of one implies that the other is not true. The sentence pairs in the dataset are partitioned into train (550,152), dev (10,000), and test (10,000). Model performance is evaluated based on classification accuracy.

Our multistask model learns a shared encoder for the conversational response prediction and SNLI tasks. We report evaluation results on the SNLI task in order to facilitate better comparison with InferSent (Conneau et al., 2017), which served as the inspiration for the inclusion of the SNLI task within a multistask model. For reference, we provide the results of Gumbel TreeLSTM (Williams et al., 2017), which is the best sentence encoder based model, and KIM Ensemble (Chen et al., 2017), which is the current state-of-the-art.

Sentence encoder based models first encode the two sentences in an SNLI input pair separately, and then feed the encodings into a classifier. By comparison, other models explicitly consider word-level interactions between the paired sentences (e.g., using cross-attention). We note that our model is sentence encoder based.

Preliminary experiments with pre-trained embeddings on a P@N Reddit response prediction evaluation revealed no performance advantage over embeddings learned directly from the data.\(^5\) We note that, if our model is reduced to just training on SNLI without multistask training on Reddit, it would be equivalent to InferSent but without the use of pretrained sentence embeddings. We do not provide results for this configuration as preliminary experiments suggested it performed poorly.\(^6\)
Table 3: Pearson’s $r$ on the STS Benchmark.

tual Similarity (STS) Benchmark. The benchmark includes English datasets from the SemEval/SEM STS shared tasks between 2012 and 2017 (Cer et al., 2017; Agirre et al., 2016, 2015, 2014, 2013, 2012). The data include 8,628 sentence pairs from three categories: captions, news and forums. Each pair is annotated with a human-labeled degree of meaning similarity, ranging from 0 to 5. The dataset is divided into train (5,749), dev (1,500) and test (1,379).

We report results using two configurations for the evaluation of the Reddit and Reddit+SNLI models. The first configuration is “out-of-the-box” with no adaptation for the STS task. Rather, we take the original sentence embeddings $u, v$ and directly score the sentence pair similarity based on the angular distance between the two vectors, $-\arccos\left(\frac{u \cdot v}{||u|| \cdot ||v||}\right)$.\footnote{\textit{arccos} is used to convert the cosine similarity scores into angular distances that obey the triangle inequality.} We suspect the original sentence embeddings from the Reddit and Reddit+SNLI models will not necessary weight all semantic distinctions in a way that is consistent with the annotations for STS. The second configuration for evaluating the two models uses a single transformation matrix to fine-tune the sentence embedding representations for the STS task. The matrix, which is parameterized using the STS training data, transforms the original sentence embedding vectors $u, v$ to $u^*, v^*$.\footnote{For both the STS shared task and the STS benchmark leaderboard, systems are allowed to use external datasets as long as they do not make use of supervised annotations on data that overlap with the evaluation sets. InferSent introduced the use of SNLI for STS. However, we discovered 4 out of the 1,379 pairs within the STS Benchmark dev set and 5 out of the 1,500 pairs in the STS Benchmark test set overlap with the SNLI training set. We do not believe this minimal overlap had a meaningful impact on the results presented here.}

Table 3 presents results on the dev and test sets of the STS Benchmark. For model comparisons, we include the state-of-the-art neural STS model CNN (HCTI) (Shao, 2017) and other systems in Cer et al. (2017).\footnote{As summarized by Cer et al. (2017), ENCU makes use of a large feature set that includes: n-gram overlap; edit distance; longest common prefix/suffix/substring; tree kernels; word alignment based similarity; summarization and MT evaluation metrics; kernel similarity of bags-of-words and bags-of-dependency triples; and pooled word embeddings. The manually engineered features are combined with scores from DANN and LSTM based deep learning models. BIT relies primarily on a measure of sentence information content (IC) with a non-trivial derivation that is optionally combined with either an alignment based similarity score or the cosine similarity of IDF weighed summed word embeddings.}

The untuned Reddit model is competitive with many of the other neural representation models, demonstrating that the sentence embeddings learned on Reddit conversations do keep text with similar semantics close in embedding space. The “out-of-the-box” multitask model, Reddit+SNLI, achieves an $r$ of 0.814 on the dev set and 0.782 on test. Using a transformation matrix to adapt the Reddit model trained without SNLI to STS, we achieve Pearson’s $r$ of 0.809 on dev and 0.781 on test. This surpasses InferSent and is close to the performance of the best neural representation approach, CNN (HCTI).\footnote{As summarized by Cer et al. (2017), InferSent uses a pre-trained model with a single transformation matrix to fine-tune the sentence embedding representations for the STS task. The matrix is parameterized using the STS training data, transforms the original sentence embedding vectors $u, v$ to $u^*, v^*$.}

The adapted multitask model achieves the best performance among all neural models, with an $r$ of 0.835 on the dev data and 0.808 on test. The results are competitive with state-of-the-art feature engineered and mixed systems, e.g. ENCU and BIT. However, our models are simpler and require no feature engineering.\footnote{As summarized by Cer et al. (2017), Inference using a large feature set that includes: n-gram overlap; edit distance; longest common prefix/suffix/substring; tree kernels; word alignment based similarity; summarization and MT evaluation metrics; kernel similarity of bags-of-words and bags-of-dependency triples; and pooled word embeddings. The manually engineered features are combined with scores from DANN and LSTM based deep learning models. BIT relies primarily on a measure of sentence information content (IC) with a non-trivial derivation that is optionally combined with either an alignment based similarity score or the cosine similarity of IDF weighed summed word embeddings.}

4.4 CQA Subtask B

To further validate the effectiveness of sentence representations learned from conversational data, we assess the proposed models on subtask B of SemEval Community Question Answering (CQA) (Nakov et al., 2017). In this task, given an “original” question $Q$, and the top ten related questions from a forum ($Q_1, \ldots, Q_{10}$) as retrieved by a search engine, the goal is to rank the related questions according to their similarity with respect to $Q$.\footnote{As summarized by Cer et al. (2017), ENCU makes use of a large feature set that includes: n-gram overlap; edit distance; longest common prefix/suffix/substring; tree kernels; word alignment based similarity; summarization and MT evaluation metrics; kernel similarity of bags-of-words and bags-of-dependency triples; and pooled word embeddings. The manually engineered features are combined with scores from DANN and LSTM based deep learning models. BIT relies primarily on a measure of sentence information content (IC) with a non-trivial derivation that is optionally combined with either an alignment based similarity score or the cosine similarity of IDF weighed summed word embeddings.}
<table>
<thead>
<tr>
<th>Score</th>
<th>Label</th>
<th>STS Input Sentences</th>
</tr>
</thead>
<tbody>
<tr>
<td>Good</td>
<td>-0.51</td>
<td>S1: a small bird sitting on a branch in winter.</td>
</tr>
<tr>
<td></td>
<td></td>
<td>S2: a small bird perched on an icy branch.</td>
</tr>
<tr>
<td>Good</td>
<td>-1.23</td>
<td>S1: microwave would be your best bet.</td>
</tr>
<tr>
<td></td>
<td></td>
<td>S2: your best bet is research.</td>
</tr>
<tr>
<td>Bad</td>
<td>-0.42</td>
<td>S1: a little boy is singing and playing a guitar.</td>
</tr>
<tr>
<td></td>
<td></td>
<td>S2: a man is singing and playing the guitar.</td>
</tr>
<tr>
<td>Bad</td>
<td>-0.45</td>
<td>S1: yes, you have to file a tax return in canada.</td>
</tr>
<tr>
<td></td>
<td></td>
<td>S2: you are not required to file a tax return in canada if you have no taxable income.</td>
</tr>
</tbody>
</table>

Table 4: Pearson’s $r$ of the proposed models on the STS Benchmark with a breakdown by category.

<table>
<thead>
<tr>
<th>Score</th>
<th>Label</th>
<th>STS Input Sentences</th>
</tr>
</thead>
<tbody>
<tr>
<td>Good</td>
<td></td>
<td>S1: a small bird sitting on a branch in winter.</td>
</tr>
<tr>
<td></td>
<td></td>
<td>S2: a small bird perched on an icy branch.</td>
</tr>
<tr>
<td>Good</td>
<td></td>
<td>S1: microwave would be your best bet.</td>
</tr>
<tr>
<td></td>
<td></td>
<td>S2: your best bet is research.</td>
</tr>
<tr>
<td>Bad</td>
<td></td>
<td>S1: a little boy is singing and playing a guitar.</td>
</tr>
<tr>
<td></td>
<td></td>
<td>S2: a man is singing and playing the guitar.</td>
</tr>
<tr>
<td>Bad</td>
<td></td>
<td>S1: yes, you have to file a tax return in canada.</td>
</tr>
<tr>
<td></td>
<td></td>
<td>S2: you are not required to file a tax return in canada if you have no taxable income.</td>
</tr>
</tbody>
</table>

Table 5: Example model and human similarity scores on pairs from the STS Benchmark. System scores are reported as the negative angular distance between the sentence embeddings. The scores can range from 0 to $-\pi$, but in practice are typically between 0 and $-\frac{1}{2}\pi$.

<table>
<thead>
<tr>
<th>Score</th>
<th>Label</th>
<th>STS Input Sentences</th>
</tr>
</thead>
<tbody>
<tr>
<td>Reddit+SNLI</td>
<td>47.42</td>
<td>MAP</td>
</tr>
<tr>
<td>Reddit</td>
<td>47.07</td>
<td>MAP</td>
</tr>
<tr>
<td>KeLP-contrastive1</td>
<td>49.00</td>
<td>MAP</td>
</tr>
<tr>
<td>SimBow-contrastive2</td>
<td>47.87</td>
<td>MAP</td>
</tr>
<tr>
<td>SimBow-primary</td>
<td>47.22</td>
<td>MAP</td>
</tr>
</tbody>
</table>

Table 6: Mean Average Precision (MAP) on Community Question Answering (CQA) subtask B.

to the original question. Mean average precision (MAP) is used to evaluate candidate models.

Each pairing of an original question and a related question $(Q, Q_i)$ is labeled “PerfectMatch”, “Relevant” or “Irrelevant”. Both “PerfectMatch” and “Relevant” are considered as good questions, which should rank above “Irrelevant” ones.

Similar to the STS experiments, we use cosine similarity between the original question and related questions, without considering any other interaction between the two questions.\(^{11}\) Given a related question $Q_i$ and its original question $Q$, we first encode them into vectors $u_i$ and $u$. Then the related questions are ranked based on the cosine similarity with respect to the original question, $\cos(u_i, u)$. Results are shown in table 6. SimBow (Charlet and Damnati, 2017) and KeLP (Filice et al., 2017), which are the best systems on the 2017 task, are used as baselines.\(^{12}\) Even without tuning on the training data provided by the task, our models show competitive performance. Reddit+SNLI outperforms SimBow-primary, which official ranked first during the 2017 shared task.

\(^{11}\)Our model also excludes the use of comments and user profiles provided by CQA as optional contextual features.

\(^{12}\)In the competition, each team can submit one primary run and two contrastive runs. Only the primary run is used for the official ranking.
5 Analysis

Model performance on the STS Benchmark can be partition by sentence pair source. The test set contains 625 sentence pairs drawn from captions, 500 pairs from news data, and 254 from online forums.

Table 4 provides results on each sub-group. For the captions category, adding the SNLI data improves the baseline Reddit model by about 8% absolute. Even with tuning to STS, mixing in SNLI data still helps dramatically on captions, as the STS tuned Reddit+SNLI model is 5% absolute higher than the STS tuned Reddit model on this category. The improvement is likely attributed to the fact that the SNLI sentences are from image captions, while Reddit doesn’t contain much caption-style data. Training with the SNLI data has a smaller impact on performance for the other categories, with even a slight decrease for the STS tuned models on news test.

We observe that the STS tuned models have only modest performance improvements on the forum data over the untuned models, with much larger improvements for captions and news. Moreover, for the Reddit+SNLI models, tuning produces a large performance increase for news with smaller increases for both captions and forums. This suggests tuning is impart compensating for domain limitations within the training data.\footnote{E.g., the Reddit+SNLI model is trained on image caption and discussion forum data but not news.}

5.1 Quantity of SNLI data and Performance

The experiments in the previous section show that supervised in-domain data, SNLI’s image captions, can be used to improve the semantic representations of in-domain (caption) sentences. However, supervised data is difficult to obtain, especially on the order of SNLI’s 570,000 sentence pairs. In order to learn how much supervised data is needed, we train multitask models with Reddit and varying amounts of SNLI data, ranging from 10% to 90% of the full dataset.

Figure 8 shows the STS Benchmark results for all data and for caption data only, on both dev and test sets. When first adding the SNLI data into the training task, Pearson’s $r$ increases rapidly across all measures. Even with only 10% of the SNLI data, $r$ reaches around 0.85 for captions data on both dev and test. The curves mostly flatten out after using 40% of the data, with performance only improving slightly past this point. This suggests encoders trained primarily on Reddit data can be efficiently adapted to perform well on other domains using a small sample of in-domain data.

6 Related Work

The STS task was first introduced by Agirre et al. (2012). Early methods focused on lexical semantics, surface form matching and basic syntactic similarity (Bär et al., 2012; Jimenez et al., 2012). More recently, deep learning based methods became competitive (Shao, 2017; Tai et al., 2015).
One approach to this task is to encode sentences into sentence-level embeddings and then calculate the cosine similarity between the encoded representations of the sentence pair. The encoding model can be directly trained on the STS task (Shao, 2017) or it can be trained on an alternative supervised (Conneau et al., 2017) or unsupervised (Pagliardini et al., 2017) task. The primary contribution of the work described in this paper falls into the latter category, introducing a new unsupervised task based on conversational data that achieves good performance on predicting semantic similarity scores. Training on input-response data has been previously shown to be effective at email response prediction (Kannan et al., 2016; Henderson et al., 2017). We extend prior work by exploring the effectiveness of representations learned from conversations in capturing general-purpose semantic information. The approach is similar to Skip-Thought (Kiros et al., 2015), which learns sentence-level representations through prior and next sentence prediction within a document. However, within our work, the adjacent sentences are pulled from turns in a conversation.

7 Conclusion

In this paper, we propose using conversational response prediction models to obtain sentence-level embeddings. We find that encodings learned for conversational response prediction perform well on sentence-level semantic similarity. Sentence embeddings extracted from a model trained on conversational data can be used to obtain results on the STS Benchmark that are competitive with well performing models based on sentence-level encoders. A multitask model trained on response prediction and SNLI achieves state-of-the-art performance for sentence encoding based models on the STS Benchmark, and surpasses prior work that trained on SNLI alone (InferSet). Finally, even without any task-specific training, the sentence embeddings obtained from both the conversational response prediction model and the multitask model that includes SNLI are competitive on CQA subtask B.

Acknowledgments

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Jeffrey Pennington, Richard Socher, and Christopher D. Manning. 2014. Glove: Global vectors for


Multilingual seq2seq training with similarity loss for cross-lingual document classification

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Abstract

In this paper, we continue the line of work where neural machine translation training is used to produce joint cross-lingual fixed-dimensional sentence embeddings. In this framework, we introduce a simple method of adding a loss to the learning objective which penalizes distance between representations of bilingually aligned sentences. We evaluate cross-lingual transfer using two approaches, cross-lingual similarity search on an aligned corpus (Europarl) and cross-lingual document classification on a recently published benchmark Reuters corpus, and find the similarity loss significantly improves performance on both. Our cross-lingual transfer performance is competitive with state-of-the-art, even while there is potential to further improve by investing in a better in-language baseline. Our results are based on a set of 6 European languages.

1 Introduction

Many real-world services collect data in many languages, and machine learning models on text need to support these languages. In practice, however, it is often only the top one or two dominant languages (usually English) which are supported because it is expensive to collect labeled training data for the task in every language. It is desirable, therefore, to obtain a representation of sequences of text that is joint across all languages, which allows for cross-lingual transfer on the languages without labeled data.

These representations typically take the form of a fixed-size embedding representing a complete sentence or document. Previous work has focused on several approaches in this setting, all of which rely on parallel corpora. In (AP et al., 2013), a predictive auto-encoder is used to learn a joint representation of a pair of sentences. (Hermann and Blunsom, 2014) constructs a bilingual sentence embedding by minimizing the squared distance between the embeddings of parallel sentences. (Pham et al., 2015) learns a common representation by simultaneously predicting n-grams in both languages from a common vector. In ( Mogadala and Rettinger, 2016), a similarity measure is used to minimize distance on both the sentence embeddings, and the average of the word embeddings of a pair of sentences. A method is also proposed to apply this approach to label-aligned corpora in the absence of sentence-aligned corpora by doing a pre-alignment.

Finally, multilingual representations can be learned using a sequence-to-sequence encoder-decoder neural machine translation (NMT) architecture, such as the one introduced in (Sutskever et al., 2014). Multilingual encoders have been successfully demonstrated in the NMT setting (Dong et al., 2015; Firat et al., 2017, 2016; Johnson et al., 2016). Recently (Schwenk et al., 2017) has proposed using this framework for generating multilingual sentence representations and apply it to cross-lingual document classification.

In this paper, we combine this NMT approach with the pairwise similarity approach to obtain better representations. In section 2 we describe our framework. Then in section 4 we present an evaluation of our method based on measuring similarity on the multiply aligned Europarl corpus ( Koehn, 2005). Section 5 contains our cross-lingual document classification experiments on the balanced version of the Reuters Corpus Volume 2 dataset (RCV2b), recently published by resampling from the Reuters Corpus Volume 2 to have a balanced distribution of languages and a similar label distribution for each language (Schwenk and Li, 2018).
Table 1: SentEval results: performance as a sentence encoder in English

<table>
<thead>
<tr>
<th>Method</th>
<th>SST</th>
<th>MR</th>
<th>CR</th>
<th>MPQA</th>
<th>SUBJ</th>
<th>TREC</th>
<th>Average</th>
</tr>
</thead>
<tbody>
<tr>
<td>(Conneau et al., 2017)</td>
<td>BLSTM, maxpool</td>
<td>81.1</td>
<td>86.3</td>
<td>92.4</td>
<td>90.2</td>
<td>84.6</td>
<td>88.2</td>
</tr>
<tr>
<td>ours, with similarity, meanpool</td>
<td>80.3</td>
<td>73.9</td>
<td>77.5</td>
<td>85.6</td>
<td>90.9</td>
<td>88.0</td>
<td>82.7</td>
</tr>
<tr>
<td>ours, with similarity, maxpool</td>
<td>80.4</td>
<td>75.0</td>
<td>79.6</td>
<td>87.3</td>
<td>91.1</td>
<td>88.0</td>
<td>83.56</td>
</tr>
<tr>
<td>ours, with similarity, self-attention</td>
<td>80.2</td>
<td>74.3</td>
<td>84.3</td>
<td>88.0</td>
<td>91.8</td>
<td>93.1</td>
<td>85.28</td>
</tr>
</tbody>
</table>

2 Multilingual encoder with similarity loss

We build mostly on the work of (Schwenk et al., 2017) of training an encoder to produce a fixed-dimension vector representation based on an aggregation over the encoder hidden states. Our setup involves a single shared encoder and decoder with six languages: English, German, French, Spanish, Italian, and Portuguese. We pair languages with English and Spanish, giving 10 unique pairings. The shared vocabulary is of size 85k.

The encoder consists of a two-layer LSTM with hidden sizes 512 and 1024, where the first layer is bidirectional. The decoder is an LSTM without attention, with hidden size 1024. Sentence representations will thus be 1024-dimensional.

We follow the method of prepending a token representing the target language as a first input for the decoder (Johnson et al., 2016). This avoids target-language specific encoder representations since the target language token is not an input to the encoder. We use gradient clipping with max norm 5. We use multi-cca trained word embeddings (Ammar et al., 2016) and allow trainable word embeddings.

2.1 Bilingual batch sampling

Our approach relies on bilingual aligned data. We do not assume multiply aligned (n-way parallel) data, even though we have it in training corpora such as Europarl. Inspired by the m:1 approach in (Schwenk et al., 2017), we train translation in both directions in each batch of bilingually aligned data.

2.2 Translation and similarity loss

We use the average over encoder hidden states to initialize the decoder, and also as a constant input to the decoder at each position, without using attention. The decoder then produces a probability distribution \( p_d(t|h) \) on the space of output sequences conditioned on the output of the encoder. Given a set of translation pairs \((s, t)\), let \( h(s) \) be the sentence embedding, an elementwise mean of the hidden states of the encoder. The translation loss penalizes the negative log likelihood of the target sequence, given the source:

\[
L_{NMT} = \frac{1}{n_t} \sum_{j=1}^{n_t} -\log p_d(t_j | t_1, \ldots, t_{j-1}; h(s))
\]

Meanwhile the similarity loss directly minimizes the distance between the embeddings of \( s \) and \( t \):

\[
L_{sim} = ||h(s) - h(t)||_2^2
\]

We combine these into our final loss term, adding weight regularization on the encoder:

\[
L = (L_{NMT}^{src\rightarrow tgt} + L_{NMT}^{tgt\rightarrow src}) + \alpha L_{sim},
\]

where \( \alpha \) needs to be chosen to balance the contributions from each term. Note that similarity loss by itself would have a degenerate solution, which is to map all inputs to a constant embedding vector. Introducing negative sampling or a contrastive loss would improve this situation. Note also that both the similarity loss has a regularization effect on the encoder weights. We also try replacing similarity loss term with an L2 norm on the encoder weights. We believe that regularizing encoder weights is important for cross-lingual transfer in that it helps prevent the encoder from “splitting” its output space by source language distribution.

The choice of \( \alpha \) depends on relative batch / weight normalization, the distribution of initial word embeddings, hidden size, and other factors. We find that starting with the two terms having comparable value is a good place to start tuning. We tune these parameters to one cross-lingual transfer task (Europarl similarity between De, En, Es).

Training takes about 1.5 days on 4 GPUs for 6 languages with 10 directions. All results are using a single trained encoder in with- and without-similarity loss settings.
Table 2: Europarl (5k) similarity retrieval accuracy from training { without encoder regularization / with encoder weight regularization / with similarity loss }. Some combinations are omitted for space.

<table>
<thead>
<tr>
<th>Retrieved language</th>
<th>De</th>
<th>En</th>
<th>Es</th>
<th>Fr</th>
<th>It</th>
<th>Pt</th>
<th>All</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>87.0</td>
<td>92.9 / 96.8</td>
<td>98.7 / 96.8</td>
<td>98.7 / 96.8</td>
<td>98.7 / 96.8</td>
<td>98.7 / 96.8</td>
<td>98.7 / 96.8</td>
</tr>
<tr>
<td>De</td>
<td>85.9</td>
<td>89.1 / 98.9</td>
<td>97.1 / 97.1</td>
<td>97.1 / 97.1</td>
<td>97.1 / 97.1</td>
<td>97.1 / 97.1</td>
<td>97.1 / 97.1</td>
</tr>
<tr>
<td>En</td>
<td>85.4</td>
<td>87.8 / 87.8</td>
<td>90.2 / 92.0</td>
<td>94.2 / 97.1</td>
<td>94.2 / 97.1</td>
<td>94.2 / 97.1</td>
<td>94.2 / 97.1</td>
</tr>
<tr>
<td>Fr</td>
<td>83.8</td>
<td>87.4 / 87.8</td>
<td>88.9 / 91.1</td>
<td>91.9 / 92.1</td>
<td>91.9 / 92.1</td>
<td>91.9 / 92.1</td>
<td>91.9 / 92.1</td>
</tr>
<tr>
<td>It</td>
<td>82.2</td>
<td>85.3 / 85.9</td>
<td>86.7 / 89.4</td>
<td>90.3 / 97.0</td>
<td>90.3 / 97.0</td>
<td>90.3 / 97.0</td>
<td>90.3 / 97.0</td>
</tr>
<tr>
<td>Pt</td>
<td>84.5</td>
<td>87.6 / 87.9</td>
<td>90.0 / 91.2</td>
<td>92.2 / 93.0</td>
<td>92.2 / 93.0</td>
<td>92.2 / 93.0</td>
<td>92.2 / 93.0</td>
</tr>
<tr>
<td>All</td>
<td>86.4</td>
<td>89.0 / 89.2</td>
<td>90.0 / 91.7</td>
<td>92.3 / 92.6</td>
<td>92.6 / 93.0</td>
<td>92.6 / 93.0</td>
<td>92.6 / 93.0</td>
</tr>
</tbody>
</table>

Table 3: Example top 3 retrieved sentences in Europarl 5k: the correctly retrieved sentence is omitted.

<table>
<thead>
<tr>
<th>Retrieving sentence</th>
<th>Retrieved (It)</th>
<th>Retrieved (Fr)</th>
</tr>
</thead>
<tbody>
<tr>
<td>Mr President, as it is now Christmas, I would be grateful if you would allow me to speak for a moment.</td>
<td>Signor Presidente, resto in Aula perché mi è stato fatto sapere che, per poter presentare una dichiarazione di voto, occorre essere presenti.</td>
<td>Monsieur le Président, je reste ici parce que l’on m’a expliqué qu’il fallait être présent dans l’hémicycle pour être autorisé à déposer des explications de vote. Puisque M. Prodi est présent, je vais lui donner la parole en premier, s’il le accepte.</td>
</tr>
</tbody>
</table>

3 English performance

We first evaluate our sentence embeddings on a set of English transfer tasks (SentEval). We compare mean pooling, max pooling, and self-attention (Lin et al., 2017) as aggregation methods, with an MLP with one hidden layer of size 128. Our results are several points lower than current best SentEval results.

4 Cross-lingual similarity search

As one of our evaluation methods, we follow (Schwenk et al., 2017) in validating that the closest sentence in an aligned corpus based on our sentence embeddings is the aligned sentence. We use cosine similarity. We use a Europarl development set of 5k sentences across 6 languages and report the accuracy of retrieval in each direction. Note that the corpus has duplicates, thus retrieval cannot be perfect, as reflected in the in-language results. We notice that Portuguese is best for retrieving Spanish sentences and Spanish is best for retrieving Italian and Portuguese sentences.

The results are shown in table 2. As a baseline, we take our setup with NMT loss only, and compare the results with similarity loss added. We see that both encoder weight regularization and similarity loss significantly improve retrieval performance, with similarity loss possibly slightly better.

5 Cross-lingual document classification

One of the main motivations for pursuing multilingual sentence embeddings is to achieve cross-lingual transfer on NLP tasks such as document classification. The multilingual Reuters News Corpus has been adopted as a standard dataset for this task. We will be using a version of this dataset that has been subsampled to obtain even label distribution prior across languages (Schwenk and Li, 2018), to make the interpretation of transfer results easier.

For these tests, we use a linear classifier (logistic regression) and tune the regularization parameter to the development set defined in RCV2Balanced.

5.1 Document segmentation

<table>
<thead>
<tr>
<th>Method</th>
<th>Mean accuracy</th>
</tr>
</thead>
<tbody>
<tr>
<td>Punctuation, meanpool</td>
<td>74.6</td>
</tr>
<tr>
<td>Punctuation, maxpool</td>
<td>67.0</td>
</tr>
<tr>
<td>Fixed window, meanpool</td>
<td>73.5</td>
</tr>
<tr>
<td>Fixed window, maxpool</td>
<td>68.2</td>
</tr>
</tbody>
</table>

Table 5: Comparison of aggregation methods for document embedding (RCV2Balanced)

Documents in the Reuters corpus are composed of many sentences. In principle, it is possible consider each document as a long sequence and use the resulting embedding from our encoder as-is;
Table 4: Cross-lingual document classification results (RCV2Balanced): from training { without encoder regularization / with similarity loss }. Zero-shot paradigm. Bold indicates best result for target language.

<table>
<thead>
<tr>
<th></th>
<th>De</th>
<th>En</th>
<th>Es</th>
<th>Fr</th>
<th>It</th>
<th>Ru</th>
<th>All</th>
</tr>
</thead>
<tbody>
<tr>
<td>De</td>
<td>(91.1 / 90.5)</td>
<td>76.8 / 77.1</td>
<td>67.2 / 76.4</td>
<td>75.3 / 81.7</td>
<td>63.5 / 71.8</td>
<td>49.5 / 60.5</td>
<td>70.5 / 73.3</td>
</tr>
<tr>
<td>En</td>
<td>72.9 / 80.2</td>
<td>(89.0 / 89.4)</td>
<td>72.2 / 74.1</td>
<td>73.0 / 71.0</td>
<td>63.4 / 70.8</td>
<td>60.9 / 65.7</td>
<td>71.9 / 76.8</td>
</tr>
<tr>
<td>Es</td>
<td>76.4 / 79.5</td>
<td>74.1 / 73.4</td>
<td>(92.0 / 92.4)</td>
<td>78.1 / 78.9</td>
<td>68.2 / 72.0</td>
<td>58.7 / 58.0</td>
<td>74.6 / 75.7</td>
</tr>
<tr>
<td>Fr</td>
<td>79.5 / <strong>82.5</strong></td>
<td>79.0 / <strong>80.8</strong></td>
<td>77.1 / 76.5</td>
<td>(87.5 / 89.9)</td>
<td>68.1 / <strong>72.7</strong></td>
<td>63.4 / 59.4</td>
<td>75.7 / 77.0</td>
</tr>
<tr>
<td>It</td>
<td>76.6 / 78.3</td>
<td>74.8 / 71.2</td>
<td><strong>76.8</strong> / 75.5</td>
<td>66.8 / 74.0</td>
<td>(81.3 / 81.8)</td>
<td>63.8 / 55.9</td>
<td>73.3 / 72.8</td>
</tr>
<tr>
<td>Ru</td>
<td>74.2 / 70.0</td>
<td>72.3 / 71.3</td>
<td>56.3 / 61.8</td>
<td>69.5 / 66.5</td>
<td>64.9 / 60.9</td>
<td>(82.2 / 84.0)</td>
<td>69.9 / 69.1</td>
</tr>
<tr>
<td>All</td>
<td>78.4 / 80.2</td>
<td>77.6 / 77.2</td>
<td>73.6 / 76.1</td>
<td>75.0 / 78.7</td>
<td>68.2 / 71.7</td>
<td>63.1 / 63.9</td>
<td><strong>72.7</strong> / 74.6</td>
</tr>
<tr>
<td>LOO</td>
<td>- / 78.5</td>
<td>- / (89.4)</td>
<td>- / 73.0</td>
<td>- / 80.5</td>
<td>- / 70.0</td>
<td>- / 65.6</td>
<td>- / (76.2)</td>
</tr>
</tbody>
</table>

Figure 1: t-SNE projection of document embeddings in RCV2Balanced, De test set

however, our encoder would have problems representing such long input sequences with fixed dimensional embeddings, especially because no attention mechanism is present. As a result, we need a method to split a document into smaller sequences, and an aggregation method to go from short sequence embeddings to a document embedding. For splitting we consider simply using the sentences, delimited by punctuation (the characters [,!?]). We also try splitting by a fixed window size (128 words) and fixed stride (64 words). For aggregation, we try elementwise mean- and max-pooling. We find that splitting on punctuation and using mean pooling works best (Table 5).

5.2 Evaluation paradigms

<table>
<thead>
<tr>
<th>Evaluation paradigm</th>
<th>Mean accuracy</th>
</tr>
</thead>
<tbody>
<tr>
<td>Zero-shot transfer</td>
<td>74.6</td>
</tr>
<tr>
<td>Targeted transfer</td>
<td>75.6</td>
</tr>
</tbody>
</table>

Table 6: Comparison of tuning to source- versus target-language development data

Following (Schwenk and Li, 2018), we use two transfer learning paradigms: zero-shot learning and targeted transfer. In zero-shot learning, we tune regularization hyperparameters to the development set in the training/source language and test on the transfer/target language, and the trained model is the same for all directions with the same source; in targeted transfer, we tune these parameters to the target development set and each model is unique for each dialect direction.

Results are compiled in table 4. It can be seen that adding similarity loss significantly improves over our baseline on average by nearly 2 points. Our best results per target language are better than best results per target language in the zero-shot paradigm in (Schwenk and Li, 2018) using word embeddings and sentence embeddings; however, these are not directly comparable given we are using significantly more training data. Finally, Figure 1 shows a t-SNE representation of the document embeddings over the four classes on a sample of RCVBalanced dataset.

We also try “leaving one out” (LOO) where we pool training data over all languages except the target to augment training data, while tuning to the English development set. However results do not improve over the best single-language transfer numbers (last row in table 4).

6 Conclusion

We presented an improved method for training multi-lingual sentence embeddings, including higher benchmark results for the RCV2 balanced dataset. We showed that including an explicit
similarity loss combined with the encoder-decoder framework improves the quality of multilingual representations. We demonstrated that our representations allow better transfer from one language to another of document classification performance.

We note that although we have shown improvements in RCV2Balanced, our English-only SentEval results are lagging state-of-the-art by at least 2 points. For future work, it is conceivable that starting from a fixed state-of-the-art English encoder (possibly with multitask training with a fixed decoder joint with the English encoder), the similarity loss method could be used to produce the same relative cross-lingual quality while preserving strong in-language performance.

Acknowledgments

We wish to thank Veselin Stoyanov for his mentorship, and our anonymous reviewers for their insightful comments. We look forward to improving this work.

References


Abstract

While recurrent neural networks have found success in a variety of natural language processing applications, they are general models of sequential data. We investigate how the properties of natural language data affect an LSTM’s ability to learn a non-linguistic task: recalling elements from its input. We find that models trained on natural language data are able to recall tokens from much longer sequences than models trained on non-language sequential data. Furthermore, we show that the LSTM learns to solve the memorization task by explicitly using a subset of its neurons to count timesteps in the input. We hypothesize that the patterns and structure in natural language data enable LSTMs to learn by providing approximate ways of reducing loss, but understanding the effect of different training data on the learnability of LSTMs remains an open question.

1 Introduction

Recurrent neural networks (RNNs; Elman, 1990), especially variants with gating mechanisms such as long short-term memory units (LSTM; Hochreiter and Schmidhuber, 1997) and gated recurrent units (GRU; Cho et al., 2014), have significantly advanced the state of the art in many NLP tasks (Mikolov et al., 2010; Vinyals et al., 2015; Bahdanau et al., 2015, among others). However, RNNs are general models of sequential data; they are not explicitly designed to capture the unique properties of language that distinguish it from generic time series data.

In this work, we probe how linguistic properties such as the hierarchical structure of language (Everaert et al., 2015), the dependencies between tokens, and the Zipfian distribution of token frequencies (Zipf, 1935) affect the ability of LSTMs to learn. To do this, we define a simple memorization task where the objective is to recall the identity of the token that occurred a fixed number of timesteps in the past, within a fixed-length input. Although the task itself is not linguistic, we use it because (1) it is a generic operation that might form part of a more complex function on arbitrary sequential data, and (2) its simplicity allows us to unfold the mechanism in the trained RNNs.

To study how linguistic properties of the training data affect an LSTM’s ability to solve the memorization task, we consider several training regimens. In the first, we train on data sampled from a uniform distribution over a fixed vocabulary. In the second, the token frequencies have a Zipfian distribution, but are otherwise independent of each other. In another, the token frequencies have a Zipfian distribution but we add Markovian dependencies to the data. Lastly, we train the model on natural language sequences. To ensure that the models truly memorize, we evaluate on uniform samples containing only rare words.¹

We observe that LSTMs trained to perform the memorization task on natural language data or data with a Zipfian distribution are able to memorize from sequences of greater length than LSTMs trained on uniformly-sampled data. Interestingly, increasing the length of Markovian dependencies in the data does not significantly help LSTMs to learn the task. We conclude that linguistic properties can help or even enable LSTMs to learn the memorization task. Why this is the case remains an open question, but we propose that the additional structure and patterns within natural language data provide additional noisy, approximate

¹This distribution is adversarial with respect to the Zipfian and natural language training sets.
paths for the model to minimize its loss, thus offering more training signal than the uniform case, in which the only way to reduce the loss is to learn the memorization function.

We further inspect how the LSTM solves the memorization task, and find that some hidden units count the number of inputs. Shi et al. (2016a) analyzed LSTM encoder-decoder translation models and found that similar counting neurons regulate the length of generated translations. Since LSTMs better memorize (and thus better count) on language data than on non-language data, and counting plays a role in encoder-decoder models, our work could also lead to improved training for sequence-to-sequence models in non-language applications (e.g., Schwaller et al., 2017).

2 The Memorization Task

To assess the ability of LSTMs to retain and use information, we propose a simple memorization task. The model is presented with a sequence of tokens and is trained to recall the identity of the middle token.2 We predict the middle token since predicting items near the beginning or the end might enable the model to avoid processing long sequences (e.g., to perfectly memorize the last token, simply set the forget gate to 0 and the input gate to 1).3 All input sequences at train and test time are of equal length. To explore the effect of sequence length on LSTM task performance, we experiment with different input sequence lengths (10, 20, 40, 60, . . . , 300).

3 Experimental Setup

We modify the linguistic properties of the training data and observe the effects on model performance. Further details are found in Appendix A, and we release code for reproducing our results.4

Model. We train an LSTM-based sequence prediction model to perform the memorization task. The model embeds input tokens with a randomly initialized embedding matrix. The embedded inputs are encoded by a single-layer LSTM and the final hidden state is passed through a linear projection to produce a probability distribution over the vocabulary. Our goal is to evaluate the memorization ability of the LSTM, so we freeze the weights of the embedding matrix and the linear output projection during training. This forces the model to rely on the LSTM parameters (the only trainable weights), since it cannot gain an advantage in the task by shifting words favorably in either the (random) input or output embedding vector spaces. We also tie the weights of the embeddings and output projection so the LSTM can focus on memorizing the timestep of interest rather than also transforming input vectors to the output embedding space.5 Finally, to examine the effect of model capacity on memorization ability, we experiment with different hidden state size values.

Datasets. We experiment with several distributions of training data for the memorization task. In all cases, a 10K vocabulary is used.

- In the uniform setup, each token in the training dataset is randomly sampled from a uniform distribution over the vocabulary.
- In the unigram setup, we modify the uniform data by integrating the Zipfian token frequencies found in natural language data. The input sequences are taken from a modified version of the Penn Treebank (Marcus et al., 1993) with randomly permuted tokens.
- In the 5gram, 10gram, and 50gram settings, we seek to augment the unigram setting with Markovian dependencies. We generate the dataset by grouping the tokens of the Penn Treebank into 5, 10, or 50-length chunks and randomly permuting these chunks.
- In the language setup, we assess the effect of using real language. The input sequences here are taken from the Penn Treebank, and thus this setup further extends the 5gram, 10gram, and 50gram datasets by adding the remaining structural properties of natural language.

We evaluate each model on a test set of uniformly sampled tokens from the 100 rarest words in the vocabulary. This evaluation setup ensures that, regardless of the data distribution the models were trained on, they are forced to generalize in

\(^{2}\) Or the \((\frac{n}{2} + 1)\)th token if the sequence length \(n\) is even.

\(^{3}\) We experimented with predicting tokens at a range of positions, and our results are not sensitive to the choice of predicting exactly the middle token.

\(^{4}\) http://nelsonliu.me/papers/lstms-exploit-linguistic-attributes/

\(^{5}\) Tying these weights constrains the embedding size to always equal the LSTM hidden state size.
order to perform well on the test set. For instance, in a test on data with a Zipfian token distribution, the model may do well by simply exploiting the training distribution (e.g., by ignoring the long tail of rare words).

4 Results

We first observe that, in every case, the LSTM is able to perform the task perfectly (or nearly so), up to some input sequence length threshold. Once the input sequence length exceeds this threshold, performance drops rapidly.

How does the training data distribution affect performance on the memorization task? Figure 1 compares memorization performance of an LSTM with 50 hidden units on various input sequence lengths when training on each of the datasets. Recall that the test set of only rare words is fixed for each length, regardless of the training data. When trained on the uniform dataset, the model is perfect up to length 10, but does no better than the random baseline with lengths above 10. Training on the unigram setting enables the model to memorize from longer sequences (up to 20), but it begins to fail with input sequences of length 40; evaluation accuracy quickly falls to 0. Adding Markovian dependencies to the unigram dataset leads to small improvements, enabling the LSTM to successfully learn on inputs of up to length 40 (in the case of 5gram and 10gram) and inputs of up to length 60 (in the case of 50gram). Lastly, training on language significantly improves model performance, and it is able to perfectly memorize with input sequences of up to 160 tokens before any significant degradation. These results clearly indicate that training on data with linguistic properties helps the LSTM learn the non-linguistic task of memorization, even though the test set has an adversarial non-linguistic distribution.

How does adding hidden units affect memorization performance? Figure 2 compares memorization performance on each dataset for LSTMs with 50, 100, and 200 hidden units. When training on the uniform dataset, increasing the number of LSTM hidden units (and thus also the embedding size) to 100 or 200 does not help it memorize longer sequences. Indeed, even at 400

Figure 1: Test set accuracy of LSTMs with 50 hidden units trained on the uniform, 5gram, and language datasets with various input sequence lengths. 5gram and 10gram perform nearly identically, so the differences may not be apparent in the figure. unigram accuracy plateaus to 0, and uniform accuracy plateaus to \( \approx 0.01\% \) (random baseline). Best viewed in color.

Figure 2: Test set accuracy of LSTMs with 50, 100 or 200 hidden units trained on each dataset with various input sequence lengths.
5 Analysis

Throughout this section, we analyze an LSTM with 100 hidden units trained with the language setting with an input sequence length of 300. This setting is a somewhat closer simulation of current NLP models, since it is trained on real language and recalls perfectly with input sequence lengths of 300 (the most difficult setting tested).

How do LSTMs solve the memorization task?
A simple way to solve the memorization task is by counting. Since all of the input sequences are of equal length and the timestep to predict is constant throughout training and testing, a successful learner could maintain a counter from the start of the input to the position of the token to be predicted (the middle item). Then, it discards its previous cell state, consumes the label’s vector, and maintains this new cell state until the end of the sequence (i.e., by setting its forget gate near 1 and its input gate near 0).

While LSTMs clearly have the expressive power needed to count and memorize, whether they can learn to do so from data is another matter. Past work has demonstrated that the LSTMs in an encoder-decoder machine translation model learn to increment and decrement a counter to generate translations of proper length (Shi et al., 2016a) and that representations produced by autoencoding LSTMs contain information about the input sequence length (Adi et al., 2017). Our experiments isolate the counting aspect from other linguistic properties of translation and autoencoding (which may indeed be correlated with counting), and also test this ability with an adversarial test distribution and much longer input sequences.

We adopt the method of Shi et al. (2016a) to investigate whether LSTMs solve the memorization task by learning to count. We identify the neurons that best predict timestep information by fitting a linear regression model to predict the number of inputs seen from the hidden unit activation. When evaluating on the test set, we observe that the LSTM cell state as a whole is very predictive of the timestep, with \( R^2 = 0.998 \).

While no single neuron perfectly records the timestep, several of them are strongly correlated. In our model instance, neuron 77 has the highest correlation \( R^2 = 0.919 \), and neuron 61 is next \( R^2 = 0.901 \). The activations of these neurons over time for a random correctly classified test input linearly increase up to the target token, after which the activations fall to nearly 0 (Figure 3).

One hypothesis for why linguistic data helps.
During training, the LSTM must: (1) determine what the objective is (here, “remember the middle token”) and (2) adjust its weights to minimize loss. We observed that adding hidden units to LSTMs trained on unigram or language sets improves their ability to learn from long input sequences, but does not affect LSTMs trained on the uniform dataset. One explanation for this disparity is that LSTMs trained on uniform data are simply not learning what the task is—they do not realize that the label always matches the token in the middle of the input sequence, and thus they cannot properly optimize for the task, even with more hidden units. On the other hand, models trained on unigram or language can determine that the label is always the middle token, and can thus learn the task. Minimizing training loss ought to be easier with more parameters, so adding hidden units to LSTMs trained on data with linguistic attributes increases the length of input sequences that they can learn from.

But why might LSTMs trained on data with linguistic attributes be able to effectively learn the task for long input sequences, whereas LSTMs trained on the uniform dataset cannot? We conjecture that linguistic data offers more reasonable, if approximate, pathways to loss minimization, such as counting frequent words or phrases. In the uniform setting, the model has only one path to success: true memorization, and it cannot find an effective way to reduce the loss. In other words, linguistic structure and the patterns of language may provide additional signals that correlate with the label and facilitate learning the memorization task.
Figure 4: Model validation and test accuracy over time during training. Validation improves faster than test, indicating that the model exploits linguistic properties of the data during training.

Figure 4 shows that models trained on the unigram and language datasets converge to high validation accuracy faster than high test accuracy. This suggests that models trained on data with linguistic attributes first learn to do well on the training data by exploiting the properties of language and not truly memorizing. Perhaps the model generalizes to actually recalling the target token later, as it refines itself with examples from the long tail of infrequent tokens.

Figure 4 may show this shift from exploiting linguistic properties to true memorization. The validation and test accuracy curves are quite synchronized from epoch 37 onward, indicating that the model’s updates affect both sets identically. The model clearly learns a strategy that works well on both datasets, which strongly suggests that it has learned to memorize. In addition, when the model begins to move toward true memorization, we’d expect validation accuracy to momentarily falter as it moves away from the crutches of linguistic features—this may be the dip at around epoch 35 from perfect validation accuracy to around 95% accuracy.

6 Related Work

To our knowledge, this work is the first to study how linguistic attributes in training data affect the ability of LSTMs to learn a simple memorization task. We find that LSTMs trained on uniformly sampled data are only able to learn the task with the sequence length of 10, whereas LSTMs trained with language data are able to learn on sequences of up to 300 tokens.

We further investigate how the LSTM learns to solve the task, and find that it uses a subset of its hidden units to track timestep information. It is still an open question why LSTMs trained on linguistic data are able to learn the task whereas LSTMs trained on uniformly sampled data cannot; based on our observations, we hypothesize that the additional patterns and structure in language-based data may provide the model with approximate paths of loss minimization, and improve LSTM trainability as a result.

7 Conclusion

In this work, we examine how linguistic attributes in training data can affect an LSTM’s ability to learn a simple memorization task. We find that LSTMs trained on uniformly sampled data are only able to learn the task with the sequence length of 10, whereas LSTMs trained with language data are able to learn on sequences of up to 300 tokens.

We further investigate how the LSTM learns to solve the task, and find that it uses a subset of its hidden units to track timestep information. It is still an open question why LSTMs trained on linguistic data are able to learn the task whereas LSTMs trained on uniformly sampled data cannot; based on our observations, we hypothesize that the additional patterns and structure in language-based data may provide the model with approximate paths of loss minimization, and improve LSTM trainability as a result.

Acknowledgments

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Appendices

A Experimental Setup Details

**Penn Treebank Processing**  Our experiments use a preprocessed version of the Penn Treebank commonly used in the language modeling community and first introduced by Mikolov et al. (2011). This dataset has 10K types, hence why we use this vocabulary size for all experiments. We generate examples by concatenating the sentences together and taking subsequences of the desired input sequence length.

**Training**  The model is trained end-to-end to directly predict the tokens at a particular timestep in the past; it is optimized with Adam (Kingma and Ba, 2015) with an initial learning rate of 0.001, which is halved whenever the validation dataset (a held-out portion of the training dataset) loss fails to improve for three consecutive epochs. The model is trained for a maximum of 240 epochs or until it converges to perfect validation performance. We do not use dropout; we included it in initial experiments, but it severely hampered model performance and does not make much sense for a task where the goal is to explicitly memorize. We ran each experiment three times with different random seeds and evaluate the model with the highest validation accuracy on the test set. We take the best since we are interested in whether the LSTMs can be trained for the task.
Learning Distributional Token Representations from Visual Features

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Abstract

In this study, we compare token representations constructed from visual features (i.e., pixels) with standard lookup-based embeddings. Our goal is to gain insight about the challenges of encoding a text representation from low-level features, e.g., from characters or pixels. We focus on Chinese, which—as a logographic language—has properties that make a representation via visual features challenging and interesting. To train and evaluate different models for the token representation, we chose the task of character-based neural machine translation (NMT) from Chinese to English. We found that a token representation computed only from visual features can achieve competitive results to lookup embeddings. However, we also show different strengths and weaknesses in the models’ performance in a part-of-speech tagging task and also a semantic similarity task. In summary, we show that it is possible to achieve a text representation only from pixels. We hope that this is a useful stepping stone for future studies that exclusively rely on visual input, or aim at exploiting visual features of written language.

1 Introduction

Language representation beyond the word level can be advantageous for words from the tail of the distribution, as has been shown in recent neural approaches for various tasks (Schütze, 2017; Kim et al., 2016; Lee et al., 2017; Wu et al., 2016; Sennrich et al., 2016). In these approaches, a neural model represents an input text based on its sequence of characters or character n-grams (instead of its words). This helps the model to handle out-of-vocabulary tokens and avoids the need of text segmentation or tokenization, which remains an unsolved problem for many languages. For some applications, e.g., language processing for social media, models should be able to capture the creative use of language. However, a disadvantage is that we loose the certainty of a known vocabulary. Also, regarding computational complexity, there is a trade-off between the memory that is required for large embedding lookup tables and the additional computational cost that is required to compose representations from low-level features.

In contrast to the characters of languages based on the Latin alphabet, the Chinese written language is defined over a set of \(\approx 8000\) characters that already carry meaning. The characters can either appear in a traditional or simplified form, and many share visual components that can indicate a related meaning. Thus, it is reasonable to hypothesize that encoding Chinese characters directly from their visual components might improve their token representation in a neural network model.

In recent studies, Liu et al. (2017) evaluated character encodings from visual features by classifying Wikipedia titles into 12 categories and Su and Lee (2017) evaluated such encodings by measuring correlation with human similarity judgments. Both studies found that using character encodings from visual input did not outperform or were equal to lookup-based embeddings. No support for the hypothesis above is thus provided.

In this study, we aim to explore the question of whether and when visual-feature representations are beneficial or detrimental. We make the following contributions: (i) Since the evaluation tasks from prior studies did not test the capabilities of the visual features for text representation, we propose to employ the task of neural machine trans-
We argue that NMT requires the token representation to serve as a reliable syntactic and semantic signal.

(ii) Prior work reported evaluations for one architecture to encode the visual features. In this paper, we evaluate different settings and argue for architecture choices that performed well in our experiments. (iii) We provide evidence for the possibility to compute token representations from visual features that perform on-par with lookup-based embeddings in NMT. (iv) Finally, we revisit two of the tasks from prior work that use visual features: measuring correlation with semantic similarity judgments by humans as well as joint segmentation and part-of-speech tagging. We use the best models from the NMT evaluation in these two tasks. For semantic similarity, we find that token representation from pixels are clearly beneficial for unseen characters, while a lookup embedding performs better for seen characters. For joint segmentation and part-of-speech tagging we find no clear difference between lookup embeddings and character representation from pixels.

2 Related Work

We start by summarizing prior work, in which token representations were obtained from bitmaps of Chinese characters.

Su and Lee (2017) propose several embedding models for Chinese, including one from visual features. For the evaluation they create a dataset with Chinese word pairs in traditional Chinese. In their evaluation, they measure the correlation of the cosine similarity of embedded word pairs against human similarity judgments for those word pairs. In their experiments, they found that the representation from visual features was not competitive to an embedding lookup model, but a combination of lookup embeddings with visual features was. For their model, they train a 5-layer CNN auto-encoder on bitmaps of Chinese characters. The auto-encoder character representation is fed into two GRU layers and two fully connected layers with ELU activation to encode the characters into a word. The model was trained to predict the Skip-Gram objective (Mikolov et al., 2013) or the Glove objective (Pennington et al., 2014).

Liu et al. (2017) evaluate the classification of Chinese Wikipedia titles into 12 categories. They found that the representation from visual features representations was not competitive to embedding lookup models. They did find, however, that a combination of lookup-based embeddings and visual features can be beneficial. Their character encoder is a convolutional neural network (CNN). The CNN consists of 3 convolutional layers with max-pooling followed by a fully connected transformation layer with ReLU activation. The character encodings are fed as input to a recurrent neural network (RNN) that encodes the Wikipedia title. Both, character encoder and classifier were trained jointly.

Costa-jussà et al. (2017) investigate the neural machine translation from Chinese to Spanish. To investigate the helpfulness of visual features they augmented lookup embeddings by concatenating the corresponding bitmap features of the characters. They showed that this approach improved the performance of both, their character-based and their word-based NMT system. They did not study the exclusive use of visual features.

Shao et al. (2017) investigate joint part-of-speech tagging and segmentation for Chinese. In contrast to the studies mentioned above, they found that the inclusion of visual features did not help. However, in their study, the CNN encoder was not pre-trained and only a small amount of training data was available, which may explain this finding. In their approach, they augmented character and n-gram embeddings with visual features. The model is a 2-layer CNN with max-pooling and a consecutive fully-connected transformation layer. They concatenated the CNN-based character embeddings to the lookup embeddings and fed them to a BI-LSTM-CRF.

Other related approaches with regards to encoding symbols from visual features are: Deng et al. (2017) proposed to translate from images to a markup language; e.g.; images of mathematical expressions to \LaTeX code. While their system performed remarkably well on those regular languages, they did not discuss their approach in the context of natural language. Remotely related is optical character recognition (OCR) and multi-modal machine translation. While OCR could benefit from this study, its main goal is character recognition under noisy conditions. Multi-modal machine translation augments a main task with additional features or when the translation is from a non-textual source, e.g. images, to text (Elliott et al., 2017). In contrast, our goal is to study text representation from low-level features.
3 Character Encoding

In this section, we first describe how a tokenized text is mapped to vector representations of its tokens. Then, we describe the character encoders that we have evaluated.

3.1 Preliminaries

Neural approaches that compute a function from text input commonly map each token \( t_i \) (e.g., a word) from the input text \( T \) to a dense vector representation \( t_i \in \mathbb{R}^d \) via some function \( D(t_i) \rightarrow t_i \). Lookup-based models associate each token of a fixed vocabulary \( V \) with its own dense representation (embedding). In particular, lookup-based models use (or learn) an embedding matrix \( E \in \mathbb{R}^{n \times d} \), where \( n = |V| \). The \( j \)-th row of \( E \) holds the embedding of the \( j \)-th token in the vocabulary. Thus \( D(t_i) = e_j^T E \), where \( j \) is the token number of token \( t_i \) in \( V \) and \( e_j \) denotes the \( j \)-th standard basis vector.

If we represent each character from its visual features—i.e., the pixels of its bitmap image—, we compute the embedding instead of performing a lookup in an embedding matrix. In particular, \( D(t_i) \) computes a dense representation of token \( t_i \) from its pixels \( p(t_i) \) via a character encoder \( C \). Therefore, there is no fixed vocabulary of tokens any more. We have \( D(t_i) = C(p(t_i)) \).

3.2 Character Encoders

As discussed in Section 2, most of the prior related work—except Costa-jussà et al. (2017)—used CNNs to learn position-invariant features of the character image. In Table 2, we report the configuration of the convolutional layers that was used in our models. All studies report different numbers of layers and different numbers of CNN features, therefore, we vary the CNN feature size \( F \) as a hyper-parameter. In Table 1 we report the character encoder architectures, that roughly cover the architectures from previous work. After the CNN, Liu et al. (2017) and Shao et al. (2017) use fully connected layers (CNN+FC+ReLU), while Su and Lee (2017) directly feed the CNN features into a recurrent neural network encoder (CNN+ReLU). We also consider a similar setup as Costa-jussà et al. (2017) by including two settings with one or two fully connected layers, (FC 1L and FC 2L). In addition to previous work, we also evaluated a model architecture that computes a softmax activation over the image features (CNN+SM+FC). Our hypothesis is that the sparse activation from the softmax may act like a soft lookup (in an \( \times 512 \) embedding matrix).

4 Experimental Study

In this section, we describe the results of our experimental study. First, we report how the character images were created, followed by the experiments for the neural machine translation task. To expand on these results, we describe experiments and results for joint segmentation and part-of-speech tagging, as well as for measuring correlation with semantic similarity judgments.
4.1 Character Images

We convert each character from the source vocabulary into a $22 \times 22$ binary representation of its glyph.\(^1\) This was the lowest resolution that did not collapse nearby strokes into indistinguishable clusters.

4.2 Experiments for Machine Translation

For the machine translation experiments, we trained a NMT model to translate from Chinese to English. In the following sections, we describe the model, the data and the training settings, followed by the results.

NMT Model We used a standard sequence-to-sequence model with a recurrent encoder and attention-based decoder (Luong et al., 2015). This architecture does not represent the state of the art in neural machine translation. However, due to its wide adoption in empirical research, there is broad knowledge about suitable hyper-parameters, which makes it a preferred choice for our study.

The coarse architecture of this model can be described by an encoder $\text{enc}$ and a decoder $\text{dec}$. The encoder $\text{enc}$ is a function that takes a tokenized text $T$ in a source vocabulary as input and computes a representation that is the input for the subsequent decoder. The decoder $\text{dec}$ then creates a sequence of tokens in the target vocabulary. The final output is $\text{dec}(\text{enc}(T))$.

The input of the encoder $\text{enc}$ is first transformed to a dense vector representation, i.e., each input token is transformed by function $D$ of Section 3.1. In the following experiments, we evaluate different choices for $D$.

Data For training the translation model, we used a subset of the available data from the WMT 2017 Workshop on Machine Translation (Bojar et al., 2017),\(^2\) namely the News Commentary v12 corpus as well as Casia2015 and Neu2017 from the CWMT Corpus. Overall, these datasets yielded 3,277,330 sentences. For development and evaluation, we used WMT 2017 dev and test data, respectively.

Training For Chinese, we use characters as input. For English, we use byte pair encoding (Sennrich et al., 2016) with $\approx 32,000$ symbols. We implemented the model in PyTorch (Version 0.3.1) (Paszke et al., 2017). The recurrent neural networks in the encoder and decoder are LSTMs (Hochreiter and Schmidhuber, 1997). Table 3 summarizes the hyper-parameters of the NMT model.

For training, we used Adam (Kingma and Ba, 2014) with the standard parameters of PyTorch. We used a learning rate of $10^{-3}$ for 4 epochs, then we halved the learning rate every epoch until we reached $10^{-5}$. For regularization, we used dropout with probability 0.2 in both encoder and decoder RNNs. We pretrained the decoder for 20 epochs to a perplexity of 108.21 on the validation data. This did result in: faster convergence, improved fluency of translations, and less variance of evaluation results. We trained all the models for 10 epochs. We selected the models by the best batch-based approximate BLEU score on the development set. This worked slightly better than selecting them by the best perplexity. We perform translation via beam search with a beam-size of 10 and length normalization of 0.9.

We trained the NMT models with each of the character encoders of Table 1 as well as with lookup-based embeddings (EMB) as a baseline.

Results Table 4 summarizes the results of the machine translation experiments. We include the result from a WMT 2017 system to give context for the expected BLEU score in this task. This result is from a baseline NMT system for Chinese to English (Wang et al., 2017), which is most comparable to our approach (i.e., no reranking, no ensembles, no special treatment of names and numbers). However, a crucial difference is that the WMT 2017 system uses pre-segmented text, which yields a vocabulary size of 300k on the

Table 3: Hyperparameters of the sequence-to-sequence model for the NMT task.

<table>
<thead>
<tr>
<th>Encoder/Decoder</th>
<th>Emb. size</th>
<th>Layers</th>
<th>Hid. size</th>
<th>Voc. size</th>
<th>Voc. type</th>
</tr>
</thead>
<tbody>
<tr>
<td>Encoder</td>
<td>512</td>
<td>bi-dir + uni-dir</td>
<td>512 + 1024</td>
<td>8457</td>
<td>character</td>
</tr>
<tr>
<td>Decoder</td>
<td>512</td>
<td>uni-dir + uni-dir</td>
<td>512 + 1024</td>
<td>32413</td>
<td>byte pair enc.</td>
</tr>
</tbody>
</table>

\(^1\)We use ImageMagick's convert command together with the open source font NotoSansCJK-Regular.

\(^2\)See http://www.statmt.org/wmt17/translation-task.html
Table 4: BLEU scores for translation zh-en on the WMT17 test data. Result for a word-segmented baseline model reported by Wang et al. (2017).

<table>
<thead>
<tr>
<th>Model</th>
<th>F</th>
<th>BLEU</th>
</tr>
</thead>
<tbody>
<tr>
<td>EMB</td>
<td>-</td>
<td>16.63</td>
</tr>
<tr>
<td>CNN+FC+ReLU 256</td>
<td>16.34</td>
<td></td>
</tr>
<tr>
<td>512</td>
<td>16.33</td>
<td></td>
</tr>
<tr>
<td>1024</td>
<td>16.60</td>
<td></td>
</tr>
<tr>
<td>2048</td>
<td>16.40</td>
<td></td>
</tr>
<tr>
<td>4096</td>
<td>16.20</td>
<td></td>
</tr>
<tr>
<td>CNN+SM+FC 512</td>
<td>15.85</td>
<td></td>
</tr>
<tr>
<td>1024</td>
<td>15.49</td>
<td></td>
</tr>
<tr>
<td>2048</td>
<td>15.36</td>
<td></td>
</tr>
<tr>
<td>4096</td>
<td>16.03</td>
<td></td>
</tr>
<tr>
<td>CNN+ReLU</td>
<td>16.22</td>
<td></td>
</tr>
<tr>
<td>FC 1L</td>
<td>15.75</td>
<td></td>
</tr>
<tr>
<td>FC 2L</td>
<td>16.32</td>
<td></td>
</tr>
</tbody>
</table>

Word-segmented - 19.4

Table 5: BLEU scores on the WMT17 test data for sentence buckets with low, medium and high frequency characters.

<table>
<thead>
<tr>
<th>Model</th>
<th>full</th>
<th>low</th>
<th>mid</th>
<th>high</th>
</tr>
</thead>
<tbody>
<tr>
<td>EMB</td>
<td>16.63</td>
<td>15.60</td>
<td>17.17</td>
<td>16.03</td>
</tr>
<tr>
<td>CNN+SM+FC 4096</td>
<td>16.03</td>
<td>13.91</td>
<td>16.86</td>
<td>15.41</td>
</tr>
<tr>
<td>CNN+FC+ReLU 1024</td>
<td>16.60</td>
<td>14.47</td>
<td>17.11</td>
<td>16.91</td>
</tr>
<tr>
<td>FC 2L</td>
<td>16.32</td>
<td>15.10</td>
<td>17.20</td>
<td>16.37</td>
</tr>
</tbody>
</table>

source side, while our system is character-based.

Our results indicate that using only one fully connected transformation layer FC 1L for character encoding is possible but does not yield comparable results to the best-performing convolutional architectures. However, the FC 2L comes close to the CNN models. Note that CNN+SM+FC 4096 needs a larger number of CNN features to perform comparable to CNN+FC+ReLU 1024. In terms of BLEU score, the best character encoders perform equal to a standard lookup embedding.

To gain insight into the differences between the models, we split the test data into three buckets. Each sentence is scored with $1/n \sum_{i=1}^{n} \#t_i$, where $\#t_i$ is the training data frequency of character $t_i$. We partition the test data into a low and high bucket with the sentences scored in the lower and upper quartile, respectively, and a mid bucket with the remaining 2nd and 3rd quartile. Table 5 shows the results per bucket for CNN+FC+ReLU 1024, CNN+SM+FC 4096, FC 2L and EMB. Interestingly, the CNN+FC+ReLU 1024 and EMB perform differently well in the frequency buckets. The CNN+FC+ReLU 1024 model, surprisingly, performs better than EMB for high-frequency characters, while this is inverted for the low-frequency characters.

For the CNN+SM+FC encoders, our hypothesis was that the sparse activation of the softmax could act like a soft lookup function; i.e., it selects only few rows for the subsequent fully connected layer. We measured the top-3 softmax activation magnitudes per character from a random sample of 350 sentences from the training data (a total of 11234 words). As shown in Figure 1, the activations are indeed spiked, which supports our hypothesis. However, in our evaluation, we found no advantage over CNN+FC+ReLU encoders.

### 4.3 Experiments for Joint Part-of-Speech Tagging and Segmentation

In the translation experiment we evaluated the token representations for a syntactic and semantic signal. In this experiment, we want to investigate the morpho-syntactic information in the representations. To evaluate this, we employ the task of joint part-of-speech tagging and segmentation.

#### Model

We used a Linear-Chain-CRF with a CNN encoder (Strubell et al., 2017) to compute the emissions. The encoder is a three-layer CNN with kernel size 3, iteratively growing dilations, residual connections, ReLU activations, and a hidden size of 512. We do not report numbers for a BI-RNN-CRF similar to Shao et al. (2017), because our implementation did not yield their results for unigrams which is most likely caused by the different embeddings.

![Figure 1: Distributions of the top-3 softmax activation magnitudes per character.](image-url)

Figure 1 shows the distributions of the top-3 softmax activation magnitudes per character.
<table>
<thead>
<tr>
<th>Model</th>
<th>F1 adapt</th>
<th>F1 fix</th>
</tr>
</thead>
<tbody>
<tr>
<td>EMB</td>
<td>91.42</td>
<td>92.18</td>
</tr>
<tr>
<td>CNN+FC+ReLU 1024</td>
<td>91.99</td>
<td>90.75</td>
</tr>
<tr>
<td>FC2L 512</td>
<td>88.45</td>
<td>91.32</td>
</tr>
<tr>
<td>CNN+SM+FC 4096</td>
<td>71.54</td>
<td>87.63</td>
</tr>
<tr>
<td>1-gram BI-RNN-CRF*/**</td>
<td>92.45</td>
<td></td>
</tr>
<tr>
<td>3-gram BI-RNN-CRF*</td>
<td>94.07</td>
<td></td>
</tr>
</tbody>
</table>

Table 6: Results for joint segmentation and POS tagging on the test set of CTB-5.0. (*) 1-gram and 3-gram BI-RNN-CRF reported by Shao et al. (2017), (**) the 1-gram BI-RNN-CRF result was reported on the development set.

**Data**  The Chinese Treebank 5.0 (Xue et al., 2005) has joint annotations for word segmentation and part-of-speech tags. The commonly used cross validation split was reported by Jiang et al. (2008). Instead of the BIES tagging scheme (begin, inside, end, single), we used the BI scheme, which worked better for our models.

**Training** We use the parameters from the EMB, CNN+FC+ReLU 1024, FC2L 512 and CNN+SM+FC 4096 of the NMT task. We either adjust these parameters during training (adapt) or keep them fixed (fix). For training, we used Adagrad (Duchi et al., 2010) with an initial learning rate of 0.1, and dropout for regularization with probability 0.1 for the encoder and 0.5 for the output classifier. We trained the models for 50 epochs and selected the model with the best accuracy on the development set.

**Results** Our results reported in Table 6 are averaged over two distinct runs each. For comparison, we show results reported for the 1-gram and 3-gram lookup model of Shao et al. (2017). Our models correspond to the 1-gram model, as they are character-based. We find no clear difference between lookup-based and the best character encoder models. Interestingly, CNN+SM+FC 4096 (softmax) is the weakest model. The sparseness in the in the signal apparently removes syntactic information. Comparing the adapt and the fix setting, we see that adapting the character encoder during training is—in most cases—not helpful. This is most likely due to the small amount of training data.

### 4.4 Experiments for Word Similarity

In our final experiment, we evaluated the representation of semantic information. The task is to compute a similarity score for pairs of words. The evaluation is the Spearman’s correlation between the scores and numerical similarity judgments by humans. The words in this data set are translated into traditional Chinese. The training data from the NMT task is mostly from news and web sources, which is why our vocabulary contains many but not all Chinese characters. The characters in the modern simplified Chinese sometimes can be visually similar to their predecessors in traditional Chinese, i.e. they share visual components with a related meaning. Therefore, we can also evaluate the capability of the models to generalize to unseen characters.

**Data and Experimental Setup** For the experiments we use the WordSim-240, WordSim-296 and SimLex-999 datasets provided and described by Su and Lee (2017). Due to the use of traditional Chinese we are missing at least one character in 430 out of 1536 test examples. On average, we are missing 1.1 characters per word, where each word has in average 2.13 characters. We split the data into word pairs in which all characters are completely covered (seen words), and into word pairs where at least one character is not seen (unseen words). We evaluate unseen words in two settings: either we remove the unseen characters (seen chars) or not (all chars).

We did not train any model for this experiments. Most of the words in the dataset are composed of multiple characters, therefore, we average the output of the character encoder into a single word vector. Subsequently, we compute the cosine similarity between these vectors of word pairs.

**Results** The results in Table 7 show that, for words with seen characters, the lookup embeddings correlate better with human judgments of similarity than the embeddings based on character encoders. However, especially the results for unseen words in WordSim-240 show that the character encoders can generalize to all characters. Surprisingly, the FC2L model, the model without CNNs, yields the best results. In the results reported by Su and Lee (2017), the lookup embedding model SG-EMB performs much better than EMB. However, they learn a RNN-based character composition, while we average the embeddings.
Table 7: Results for semantic similarity on the WordSim-240, WordSim-296 and SimLex-999 data. Lookup SkipGram (SG-EMB) and SkipGram char. encoder (SG-CNN) reported by Su and Lee (2017).

5 Conclusion

We have shown that it is possible to compute useful text representations from visual features, i.e., pixels of Chinese characters. We used neural machine translation as a framework for training and evaluating token representations. In the NMT experiment, we found that representations from visual features are competitive to lookup embeddings in a standard NMT model. In contrast to our expectations, the visual features outperform lookup embeddings on high-frequency characters, but are weaker in low-frequency characters. We conjecture that one of the reasons is that shared visual features between characters can have related meanings or related syntactic functions, but also the opposite can be true. For example,沐 is the reflexive verb of "washing" and it contains the phonetic component 木, which can also mean "tree". Therefore, the visual information for rare characters introduces probably as many difficulties as it can be helpful.

We performed additional experiments akin to prior studies and could show that the advantage of a representation by visual features is the ability to generalize to unseen Chinese characters. Also, we could show that joint part-of-speech tagging and segmentation achieved similar results with representation from visual features.

With regard to the character encoder architecture, we find only a slight advantage of using CNNs to simply using two fully connected layers in the NMT experiment, while 2 fully connected layers outperform CNNs in the similarity experiment. Whether or not there is an advantage could only be answered by an exhaustive hyper-parameter search.

Future work is to explore how we can create a model that can distinguish helpful from unhelpful visual information for rare characters.

References


Jointly Embedding Entities and Text with Distant Supervision

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Abstract

Learning representations for knowledge base entities and concepts is becoming increasingly important for NLP applications. However, recent entity embedding methods have relied on structured resources that are expensive to create for new domains and corpora. We present a distantly-supervised method for jointly learning embeddings of entities and text from an unannotated corpus, using only a list of mappings between entities and surface forms. We learn embeddings from open-domain and biomedical corpora, and compare against prior methods that rely on human-annotated text or large knowledge graph structure. Our embeddings capture entity similarity and relatedness better than prior work, both in existing biomedical datasets and a new Wikipedia-based dataset that we release to the community. Results on analogy completion and entity sense disambiguation indicate that entities and words capture complementary information that can be effectively combined for downstream use.

1 Introduction

Distributed representations of knowledge base entities and concepts have become key elements of many recent NLP systems, for applications from document ranking (Jimeno-Yepes and Berlanga, 2015) and knowledge base completion (Toutanova et al., 2015) to clinical diagnosis code prediction (Choi et al., 2016a,b). These works have taken two broad tacks for the challenge of learning to represent entities, each of which may have multiple unique surface forms in text. Knowledge-based approaches learn entity representations based on the structure of a large knowledge base, often augmented by annotated text resources (Yamada et al., 2016; Cao et al., 2017). Other methods utilize explicitly annotated data, and have been more popular in the biomedical domain (Choi et al., 2016a; Mencia et al., 2016). Both approaches, however, are often limited by ignoring some or most of the available textual information. Furthermore, such rich structures and annotations are lacking for many specialized domains, and can be prohibitively expensive to obtain.

We propose a fully text-based method for jointly learning representations of words, the surface forms of entities, and the entities themselves, from an unannotated text corpus. We use distant supervision from a terminology, which maps entities to known surface forms. We augment the well-known log-linear skip-gram model (Mikolov et al., 2013) with additional term- and entity-based objectives, and evaluate our learned embeddings in both intrinsic and extrinsic settings.

Our joint embeddings clearly outperform prior entity embedding methods on similarity and relatedness evaluations. Entity and word embeddings capture complementary information, yielding improved performance when they are combined. Analogy completion results further illustrate these differences, demonstrating that entities capture domain knowledge, while word embeddings capture morphological and lexical information. Finally, we see that an oracle combination of entity and text embeddings nearly matches a state of the art unsupervised method for biomedical word sense disambiguation that uses complex knowledge-based approaches. However, our embeddings show a significant drop in performance compared to prior work in a newswire disambiguation dataset, indicating that knowledge graph structure contains entity information that a purely text-based approach does not capture.
2 Related Work

Knowledge-based approaches to entity representation are well-studied in recent literature. Several approaches have learned representations from knowledge graph structure alone (Grover and Leskovec, 2016; Yang et al., 2016; Wang et al., 2017). Wang et al. (2014), Yamada et al. (2016), and Cao et al. (2017) all use a joint embedding method, learning representations of text from a large corpus and entities from a knowledge graph; however, they rely on the disambiguated entity annotations in Wikipedia to align their models. Fang et al. (2016) investigate heuristic methods for joint embedding without annotated entity mentions, but still rely on graph structure for entity training.

The robust terminologies available in the biomedical domain have been instrumental to several recent annotation–based approaches. De Vine et al. (2014) use string matching heuristics to find possible occurrences of known biomedical concepts in literature abstracts, and use the sequence of these noisy concepts (without the document text) as input for skip-gram training. Choi et al. (2016c) and Choi et al. (2016a) use sequences of structured medical observations from patients’ hospital stays for context-based learning. Finally, Mencia et al. (2016) take documents tagged with Medical Subject Heading (MeSH) topics, and use their texts to learn representations of the MeSH headers. These methods are able to draw on rich structured and semi-structured data from medical databases, but discard important textual information, and empirically are limited in the scope of the vocabularies they can embed.

3 Methods

In order to jointly learn entity and text representations from an unannotated corpus, we use distant supervision (Mintz et al., 2009) based on known terms, strings which can represent one or more entities. The mapping between terms and entities is many-to-many; for example, the same infection can be expressed as “cold” or “acute rhinitis”, but “cold” can also describe the temperature or refer to chronic obstructive lung disease.

Mappings between terms and entities are defined by a terminology.1 We extracted terminologies from two well-known knowledge bases:

Table 1: Statistics of the many-to-many mapping between terms and entities in our terminologies, including the maximum # of terms per entity.

The Unified Medical Language System (UMLS; Bodenreider, 2004): we use the mappings between concepts and strings in the MRCONSO table as our terminology. This yields 3.5 million entities, represented by 7.6 million strings in total.

Wikipedia: we use page titles and redirects as our terminology. This yields 9.7 million potential entities (pages), represented by 17.1 million total strings. Table 1 gives further statistics about the mapping between entities and surface forms in each of these terminologies.

While iterating through the training corpus, we identify any exact matches of the terms in our terminologies.2 We allow for overlapping terms: thus, “in New York City” will include an occurrence of both the terms “New York” and “New York.” Each matched term may refer to one or more entities; we do not use a disambiguation model in preprocessing, but rather assign a probability distribution over the possible entities.

3.1 Model

We extend the skip-gram model of Mikolov et al. (2013), to jointly learn vector representations of words, terms, and entities from shared textual contexts. For a given target word, term, or entity \( v \), let \( C_v = c_{-k} \ldots c_k \) be the observed contexts in a window of \( k \) words to the left and right of \( v \), and let \( N_v = n_{-k,1} \ldots n_{k,d} \) be the \( d \) random negative samples for each context word. Then, the context-based objective for training \( v \) is

\[
O(v, C_v, N_v) = \sum_{c \in C_v} \log(p(c \mid v)) + \sum_{n \in N_v} \log(p(n \mid v)) \tag{1}
\]

1Terminology is overloaded with both biomedical and lexical senses; we use it here strictly to mean a mapping between terms and entities.

2We lowercase and strip special characters and punctuation from both terms and corpus text, and then find all exact matches for the terms.
Table 2: Statistics of our embedding training corpora. # mentions is the number of exact matches found for terms in the relevant terminology. CP = corpus polysemy of a given entity. B = billion.

<table>
<thead>
<tr>
<th></th>
<th>Pubmed</th>
<th>Wikipedia</th>
<th>Gigaword</th>
</tr>
</thead>
<tbody>
<tr>
<td># tokens</td>
<td>2.6B</td>
<td>1.9B</td>
<td>4.3B</td>
</tr>
<tr>
<td># mentions</td>
<td>1.5B</td>
<td>1.4B</td>
<td>3.2B</td>
</tr>
<tr>
<td>Avg CP</td>
<td>2.54</td>
<td>1.01</td>
<td>1.01</td>
</tr>
<tr>
<td>% of entities by polysemy impact</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>CP ≥ 1</td>
<td>99.1%</td>
<td>98.6%</td>
<td>98.8%</td>
</tr>
<tr>
<td>CP ≥ 2</td>
<td>9.3%</td>
<td>3.5%</td>
<td>2.2%</td>
</tr>
<tr>
<td>CP ≥ 10</td>
<td>0.3%</td>
<td>0%</td>
<td>&lt;0.1%</td>
</tr>
</tbody>
</table>

where \(\sigma\) is the logistic function.

We use a sliding context window to iterate through our corpus. At each step, the word \(w\) at the center of the window \(C_w\) is updated using \(O(w, C_w, N_w)\), where \(N_w\) are the randomly-selected negative samples.

As terms are of variable token length, we treat each term \(t\) as an atomic unit for training, and set \(C_t\) to be the context words prior to the first token of the term and following the final token. Negative samples \(N_t\) are sampled independently of \(N_w\).

Finally, each term \(t\) can represent a set of entities \(E_t\). Vectors for these entities are updated using the same \(C_t\) and \(N_t\) from \(t\). Since the entities are latent, we weight updates with uniform probability \(|E_t|^{-1}\); attempts to learn this probability did not produce qualitatively different results from the uniform distribution. Thus, letting \(T\) be the set of terms completed at \(w\), the full objective function to maximize is:

\[
\hat{O} = O(w, C_w, N_w) + \sum_{t \in T} \left[ O(t, C_t, N_t) + \sum_{e \in E_t} \frac{1}{|E_t|} O(e, C_t, N_t) \right]
\]

(2)

Term and entity updates are only calculated when the final token of one or more terms is reached; word updates are applied at each step. To assign more weight to near contexts, we sub sample the window size at each step from \([1, k]\).

3.2 Training corpora

We train embeddings on three corpora. For our biomedical embeddings, we use 2.6 billion tokens of biomedical abstract texts from the 2016 PubMed baseline (1.5 billion noisy annotations). For comparison to previous open-domain work, we use English Wikipedia (5.5 million articles from the 2018-01-20 dump); we also use the Gigaword 5 newswire corpus (Parker et al., 2011), which does not have gold entity annotations.

As our model does not include a disambiguation module for handling ambiguous term mentions, we also calculate the expected effect of polysemous terms on each entity that we embed using a given corpus. We call this the entity’s corpus polysemy, and denote it with \(CP(e)\). For entity \(e\) with corresponding terms \(T_e\), \(CP(e)\) is given as

\[
CP(e) = \sum_{t \in T_e} \frac{f(t)}{Z_{\text{polysemy}(t)}}
\]

(3)

where \(f(t)\) is the corpus frequency of term \(t\), \(Z\) is the frequency of all terms in \(T_e\), and \(polysemy(t)\) is the number of entities that \(t\) can refer to.

Table 2 breaks down expected polysemy impact for each corpus. The vast majority of entities experience some polysemous effect in training, but very few have an average ambiguity per mention of 50% or greater. Most entities with high corpus polysemy are due to a few highly ambiguous generic strings, such as combinations and unknown. However, some specific terms are also high ambiguity: for example, Washington County refers to 30 different US counties.

3.3 Hyperparameters

For all of our embeddings, we used the following hyperparameter settings: a context window size of 2, with 5 negative samples per word; initial learning rate of 0.05 with a linear decay over 10 iterations through the corpus; minimum frequency for both words and terms of 10, and a subsampling coefficient for frequent words of 1e-5.

3.4 Baselines

We compare the words, terms, and entities learned in our model against two prior biomedical embedding methods, using pretrained embeddings from each. De Vinc et al. (2014) use sequences of automatically identified ambiguous entities for skip-gram training, and Mencia et al. (2016) use texts of documents tagged with MeSH headers to represent the header codes. The most recent comparison method for Wikipedia entities is MPME (Cao et al., 2017), which uses link anchors and graph structure to augment textual contexts. We also include skip-gram vectors as a final baseline; for Pubmed, we use pretrained embeddings with optimized hyperparameters from Chiu et al. (2016a), and we train our own embeddings with word2vec for both Wikipedia and Gigaword.

\(^3\)Unknown terms were handled by backing off to words.
4 Evaluations

Following Chiu et al. (2016b), Cao et al. (2017), and others, we evaluate our embeddings on both intrinsic and extrinsic tasks. To evaluate the semantic organization of the space, we use the standard intrinsic evaluations of similarity and relatedness and analogy completion. To explore the applicability of our embeddings to downstream applications, we apply them to named entity disambiguation. Results and analyses for each experiment are discussed in the following subsections.

4.1 Similarity and relatedness

We evaluate our biomedical embeddings on the UMNSRS datasets (Pakhomov et al., 2010), consisting of pairs of UMLS concepts with judgments of similarity (566 pairs) and relatedness (587 pairs), as assigned by medical experts. For evaluating our Wikipedia entity embeddings, we created WikiSRS, a novel dataset of similarity and relatedness judgments of paired Wikipedia entities (people, places, and organizations), as assigned by Amazon Mechanical Turk workers. We followed the design procedure of Pakhomov et al. (2010) and produced 688 pairs each of similarity and relatedness judgments; for further details on our released dataset, please see the Appendix.

For each labeled entity pair, we calculated the cosine similarity of their embeddings, and ranked the pairs in order of descending similarity. We report Spearman’s ρ on these rankings as compared to the ranked human judgments: Table 3 shows results for UMNSRS, and Table 4 for WikiSRS.

As the dataset includes both string and disambiguated entity forms for each pair, we evaluate each type of embeddings learned in our model. Additionally, as words and entities are embedded in the same space (and thus directly comparable), we experiment with two methods of combining their information. Entity+Word sums the cosine similarities calculated between the entity embeddings and word embeddings for each pair; the Cross setting further adds comparisons of each entity in the pair to the string form of the other.

4.1.1 Results

Our proposed method clearly outperforms prior work and text-based baselines on both datasets. Further, we see that the words and entities learned by our model include complementary information, as combining them further increases our ranking performance by a large margin. As the results on UMNSRS could have been due to our model’s ability to embed many more entities than prior methods, we also filtered the dataset to the 255 similarity pairs and 260 relatedness pairs that all evaluated entity-level methods could represent;\(^4\) Table 3 shows similar gains on this even footing. We follow Rastogi et al. (2015) in calculating significance, and use their statistics to estimate the minimum required difference for significant improvements on our datasets.

In UMNSRS, we found that cosine similarity of entities consistently reflected human judgments of similarity better than of relatedness; this reflects previous observations by Agirre et al. (2009) and Muneeb et al. (2015). Interestingly, we see the opposite behavior in WikiSRS, where relatedness is captured better than similarity in all settings. In fact, we see a number of errors of relatedness

\(^4\)For WikiSRS, all methods covered all pairs.

<table>
<thead>
<tr>
<th>Method</th>
<th>Full</th>
<th>Filtered</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>Sim</td>
<td>Rel</td>
</tr>
<tr>
<td>Prior work</td>
<td></td>
<td></td>
</tr>
<tr>
<td>word2vec</td>
<td>0.559</td>
<td>0.496</td>
</tr>
<tr>
<td>DeVine’14</td>
<td>0.455</td>
<td>0.422</td>
</tr>
<tr>
<td>Mencia’16</td>
<td>0.565</td>
<td>0.534</td>
</tr>
</tbody>
</table>

Proposed

<table>
<thead>
<tr>
<th>Method</th>
<th>Full</th>
<th>Filtered</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>Sim</td>
<td>Rel</td>
</tr>
<tr>
<td>Word</td>
<td>0.561</td>
<td>0.490</td>
</tr>
<tr>
<td>Term</td>
<td>0.619</td>
<td>0.557*</td>
</tr>
<tr>
<td>Entity</td>
<td>0.663*</td>
<td>0.563*</td>
</tr>
<tr>
<td>Entity+Word</td>
<td>0.662*</td>
<td>0.588*</td>
</tr>
<tr>
<td>+Cross</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

Table 3: Spearman’s ρ for similarity/relatedness predictions in UMNSRS. Filtered results indicate performance on the shared-vocabulary subset. *=significantly better (p < 0.05) than word baseline (full). DeVine et al (filtered).

<table>
<thead>
<tr>
<th>Method</th>
<th>Wikipedia</th>
<th>Gigaword</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>Sim</td>
<td>Rel</td>
</tr>
<tr>
<td>Prior work</td>
<td></td>
<td></td>
</tr>
<tr>
<td>word2vec</td>
<td>0.630</td>
<td>0.630</td>
</tr>
<tr>
<td>MPME</td>
<td>0.506</td>
<td>0.567</td>
</tr>
</tbody>
</table>

Proposed

<table>
<thead>
<tr>
<th>Method</th>
<th>Wikipedia</th>
<th>Gigaword</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>Sim</td>
<td>Rel</td>
</tr>
<tr>
<td>Word</td>
<td>0.646</td>
<td>0.655</td>
</tr>
<tr>
<td>Term</td>
<td>0.607</td>
<td>0.667</td>
</tr>
<tr>
<td>Entity</td>
<td>0.594</td>
<td>0.648</td>
</tr>
<tr>
<td>Entity+Word</td>
<td>0.718*</td>
<td>0.754*</td>
</tr>
<tr>
<td>+Cross</td>
<td>0.697*</td>
<td>0.753*</td>
</tr>
</tbody>
</table>

Table 4: Spearman’s ρ for similarity/relatedness predictions in WikiSRS, training on two corpora. All Proposed results are significantly better than MPME; *=significantly better than strongest word-level baseline (p < 0.05).
4.1.2 Comparing entities and words

We observe clear differences in the rankings made by entity vs word embeddings. As shown in Table 5, highly related entities tend to have high cosine similarity, while word embeddings are more sensitive to lexical overlap and direct cooccurrence. Combining both sources often gives the most intuitive results, balancing lexical effects with relatedness. For example, while the top three pairs in UMNSRS are likely to co-occur, the top three in WikiSRS are pairs of drug choices (antibiotics, ACE inhibitors, and chemotherapy drugs, respectively), only one of which is likely to be prescribed to any given patient at once.

These differences also play out in erroneous predictions. Entity embeddings often fix the worst misrankings by words: for example, “Tony Blair” and “United Kingdom” (gold rank: 28) are ranked highly unrelated (position 633) by words, but entities move this pair back up the list (position 86). However, errors made by entity embeddings are often also made by words: e.g., “C0011175 (dehydration)” and “C0017160 (gastroenteritis)” are erroneously ranked as highly unrelated by both methods. Interestingly, we find no correlation between the corpus polysemy of entity pairs and ranking performance, indicating that ambiguity of term mentions is not a significant confound for this task.

<table>
<thead>
<tr>
<th>Dataset</th>
<th>Words</th>
<th>Entities</th>
<th>Entity+Word+Cross</th>
</tr>
</thead>
<tbody>
<tr>
<td>UMNSRS</td>
<td>Iron/Iron</td>
<td>Iron/Iron</td>
<td>Levaquin/Avelox</td>
</tr>
<tr>
<td></td>
<td>Nausea/Vomiting</td>
<td>Sinemet/Sinemet</td>
<td>Enalapril/Lisinopril</td>
</tr>
<tr>
<td></td>
<td>Lipitor/Zocor</td>
<td>Enalapril/Lisinopril</td>
<td>Carboplatin/Cisplatin</td>
</tr>
<tr>
<td>WikiSRS</td>
<td>Minas Tirith/Minas Morgul</td>
<td>Real Madrid/FC Barcelona</td>
<td>Ferrari/Lamborghini</td>
</tr>
<tr>
<td></td>
<td>Moscow/Moscow Kremlin</td>
<td>Minas Tirith/Minas Morgul</td>
<td>Moscow/Moscow Kremlin</td>
</tr>
<tr>
<td></td>
<td>Norway/Denmark</td>
<td>Charlize Theron/Screen Actor’s Guild</td>
<td>Toshiro Mifune/Akira Kurosawa</td>
</tr>
</tbody>
</table>

Table 5: Top 3 pairs in the Relatedness datasets, as ranked by different embedding methods.

in WikiSRS predictions, e.g., “Hammurabi I” and “Syria” are marked highly similar, while the composers “A.R. Rahman” and “John Phillip Sousa” are marked dis-similar. MPME embeddings tend towards over-relatedness as well (e.g., ranking “Richard Feynman” and “Paris-Sorbonne University” much more highly than gold labels). Despite better similarity performance, this trend of over-relatedness also holds in biomedical embeddings: for example, “C0027358 (Narcan)” and “C0026549 (morphine)” are consistently marked highly similar across embedding methods, even though Narcan blocks the effects of opioids like morphine.

4.2 Analogy completion

We use analogy completion to further explore the properties of our joint embeddings. Given analogy \( a : b :: c : d \), the task is to guess \( d \) given \( (a, b, c) \), typically by choosing the word or entity with highest cosine similarity to \( b - a + c \) (Levy and Goldberg, 2014). We report accuracy using the top guess (ignoring \( a, b, \) and \( c \) as candidates, per Linzen, 2016).

<table>
<thead>
<tr>
<th>Method</th>
<th>B3</th>
<th>H1</th>
<th>C6</th>
<th>L1</th>
<th>L6</th>
</tr>
</thead>
<tbody>
<tr>
<td>Oracle</td>
<td>20.7</td>
<td>22.9</td>
<td>12.1</td>
<td>55.0</td>
<td>70.9</td>
</tr>
<tr>
<td>Words</td>
<td>18.3</td>
<td>22.4</td>
<td>4.5</td>
<td>10.6</td>
<td>10.0</td>
</tr>
</tbody>
</table>

Table 6: Accuracy % on 5 of the relations in BMASS with greatest absolute difference in word performance vs entity performance: B3 (gene-encodes-product), H1 (refers-to), C6 (associated-with), L1 (form-of), and L6 (has-free-acid-or-base-form). The better of word and entity performance is highlighted; all entity vs word differences are significant (McNemar’s test; \( p < 0.01 \)).

4.2.1 Biomedical analogies

To compare between word and entity representations, we use the entity-level biomedical dataset BMASS (Newman-Griffis et al., 2017), which includes both entity and string forms for each analogy. In order to test if words and entities are capturing complementary information, we also include an oracle evaluation, in which an analogy is counted as correct if either words or entities produce a correct response.\(^5\) We do not compare against prior biomedical entity embedding methods on this dataset, due to their limited vocabulary.

Table 6 contrasts the performance of different jointly-trained representations for five relations with the largest performance differences from this dataset. For gene-encodes-product and refers-to, both of which require structured domain knowledge, entity embeddings significantly

\(^5\)We use the Multi-Answer setting for our evaluation (a single \( (a, b, c) \) triple, but a set of correct values for \( d \)).
outperform word-level representations. Many of the errors made by word embeddings in these relations are due to lexical over-sensitivity: for example, in the renaming analogy spinal epidural hematoma:epidural hemorrhage::canis familiaris:, words suggest latinate completions such as latrans and caballus, while entities capture the correct C1280551 (dog). However, on more morphological relations such as has-free-acid-or-base-form, words are by far the better option.

The success of the oracle combination method for entity and word predictions clearly indicates that not only are words and entities capturing different knowledge, but that it is complementary. In the majority of the 25 relations in BMASS, oracle results improved on words and entities alone by at least 10% relative. In some cases, as with has-free-acid-or-base-form, one method does most of the heavy lifting. In several others, including the challenging (and open-ended) associated-with, entities and words capture nearly orthogonal cases, leading to large jumps in oracle performance.

### 4.2.2 General-domain analogies

No entity-level encyclopedic analogy dataset is available, so we follow Cao et al. (2017) in evaluating the effect of joint training on words using the Google analogy set (Mikolov et al., 2013). As shown in Table 7, our Wikipedia embeddings roughly match MPME embeddings (which use annotated entity links) on the semantic portion of the dataset, but our ability to train on unannotated Gigaword boosts our results on all relations except city-in-state. Overall, we find that jointly-trained word embeddings split performance with word-only skipgram training, but that word-only training tends to get consistently closer to the correct answer. This suggests that terms and entities may conflict with word-level semantic signals.

### 4.3 Entity disambiguation

Finally, to get a picture of the impact of our embedding method on downstream applications, we investigated entity disambiguation. Given a named entity occurrence in context, the task is to assign a canonical identifier to the entity being referred to: e.g., to mark that “New York” refers to the city in the sentence, “The mayor of New York held a press conference.” It bears noting that in unambiguous cases, a terminology alone is sufficient to link the correct entity: for example, “Barack Obama” can only refer to a single entity, regardless of context. However, many entity strings (e.g., “cold”, “New York”) are ambiguous, necessitating the use of alternate sources of information such as our embeddings to assign the correct entity.

#### 4.3.1 Biomedical abstracts

We evaluate on the MSH WSD dataset (Jimeno-Yepes et al., 2011), a benchmark for biomedical word sense disambiguation. MSH WSD consists of mentions of 203 ambiguous terms in biomedical literature, with over 30,000 total instances. Each sample is annotated with the set of UMLS entities the term could refer to. We adopt the unsupervised method of Sabbir et al. (2016), which combines cosine similarity and projection magnitude of an entity representation $e$ to the averaged word embeddings of its contexts $C_{avg}$ as follows:

$$f(e, C_{avg}) = \cos(C_{avg}, e) \cdot \frac{||P(C_{avg}, e)||}{||e||}$$

(4)

The entity maximizing this score is predicted.

We compare against concept embeddings learned by Sabbir et al. (2016). They used MetaMap (Aronson and Lang, 2010) with the disambiguation module enabled on a curated corpus of 5 million Pubmed abstracts to create a UMLS concept cooccurrence corpus for word2vec training. As shown in Table 8, our method lags behind theirs, though it clearly beats both random (49.7% accuracy) and majority class (52%) baselines. In addition, we leverage our jointly-embedded entities and words by adding in the definition-based model used by Pakhomov et al. (2016), which calculates an entity’s embedding as the average of definitions of its neighbors in the UMLS hierarchy (McInnes et al., 2011). We use this alternate

<table>
<thead>
<tr>
<th>Method</th>
<th>Capital (common)</th>
<th>Capital (all)</th>
<th>Currency</th>
<th>City in State</th>
<th>Family</th>
</tr>
</thead>
<tbody>
<tr>
<td>word2vec (W)</td>
<td>89.1</td>
<td>86.0</td>
<td>15.0</td>
<td>55.5</td>
<td>82.4</td>
</tr>
<tr>
<td>word2vec (G)</td>
<td>90.9</td>
<td>89.7</td>
<td><strong>18.4</strong></td>
<td>38.4</td>
<td>81.0</td>
</tr>
<tr>
<td>MPME (W)</td>
<td>83.6</td>
<td>80.5</td>
<td>11.9</td>
<td>50.6</td>
<td>78.9</td>
</tr>
<tr>
<td>Proposed (W)</td>
<td>90.1</td>
<td>78.7</td>
<td>9.1</td>
<td>42.5</td>
<td>75.5</td>
</tr>
<tr>
<td>Proposed (G)</td>
<td><strong>92.7</strong></td>
<td><strong>92.3</strong></td>
<td>16.4</td>
<td>31.3</td>
<td>81.6</td>
</tr>
</tbody>
</table>

Table 7: Analogy completion accuracy % on the semantic relations in the Google analogy dataset. W=Wikipedia, G=Gigaword.
entity embedding in Equation 4 to calculate a second score that we add to the direct entity embedding score. This yields a large performance boost of over 60% absolute, indicating that using entities and words together makes up much of the gap between our distantly supervised embeddings and the external resources used by Sabbir et al. (2016). Using the definition-based method alone with our jointly-embedded words, we see a significant increase over Pakhomov et al. (2016), indicating the benefits of joint training. However, the combined entity and definition model still yields a significantly different 2% boost in accuracy over definitions alone. Finally, we evaluate an oracle combination that reports correct if either entity or definition embeddings achieve the correct result; as shown in the last row of Table 8, this combination outperforms the entity-only method of Sabbir et al. (2016), and approaches their state-of-the-art result that combines entity embeddings with a knowledge-based approach from the structure of the UMLS.

Specific errors shed more light on these differences. The definition-based method performs better in many cases where the surface form is a common word, such as coffee (68% definition accuracy vs 28% entity accuracy) and iris (93% definition accuracy vs 35% entity accuracy). Entities outperform on some more technical cases, such as potassium (74% entity accuracy vs 49% definition accuracy). Combining both approaches in the joint model recovers performance on several cases of low entity accuracy: for example, joint accuracy on coffee is 68%, and on lupus (53% entity accuracy), joint performance is 60%.

Table 8: MSH WSD disambiguation accuracy. Definitions is comparable to Pakhomov et al. (2016), using jointly-embedded words. All differences are significant (McNemar’s test, \( p \ll 0.01 \)).

<table>
<thead>
<tr>
<th>Method</th>
<th>Accuracy %</th>
</tr>
</thead>
<tbody>
<tr>
<td>Baselines</td>
<td></td>
</tr>
<tr>
<td>Sabbir et al. (2016) (entities; +MetaMap)</td>
<td>89.3</td>
</tr>
<tr>
<td>Sabbir et al. (2016) (+MetaMap, UMLS)</td>
<td>92.2</td>
</tr>
<tr>
<td>Pakhomov et al. (2016) (words)</td>
<td>77.7</td>
</tr>
<tr>
<td>Proposed</td>
<td></td>
</tr>
<tr>
<td>Entities</td>
<td>76.4</td>
</tr>
<tr>
<td>Definitions (joint words)</td>
<td>80.8</td>
</tr>
<tr>
<td>Entities+Definitions</td>
<td>82.7</td>
</tr>
<tr>
<td>Oracle (Entities—Definitions)</td>
<td>90.9</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>Method</th>
<th>Accuracy %</th>
</tr>
</thead>
<tbody>
<tr>
<td>MPME (entities; +graph structure)</td>
<td>89.0</td>
</tr>
<tr>
<td>Wikipedia</td>
<td>40.9</td>
</tr>
<tr>
<td>Wikipedia + mentions</td>
<td>44.6</td>
</tr>
<tr>
<td>Gigaword</td>
<td>58.0</td>
</tr>
<tr>
<td>Gigaword + mentions</td>
<td>63.9</td>
</tr>
</tbody>
</table>

Table 9: AIDA linking accuracy, using entity embeddings trained on Wikipedia and Gigaword. All differences are significant (McNemar’s test, \( p \ll 0.01 \)).

4.3.2 Newswire entities

AIDA (Hoffart et al., 2011) is a standard dataset for entity linking in newswire, consisting of approximately 30,000 entities linked to Wikipedia page IDs. To reduce the search space, Pershina et al. (2015) provided a set of candidate entities for each mention, which we use for our experiments. The MPME model of Cao et al. (2017) achieves near state-of-the-art performance accuracy on AIDA with this candidate set, using the mention sense distributions and full document context included in the model. As our embeddings are trained without explicit entity annotations, we instead use the same cosine similarity and projection model discussed in Section 4.3.1 for this task. In contrast to our results on the biomedical data, we see performance far below the baseline on these data, as shown in Table 9.

However, we improve this performance slightly by multiplying by the similarity between the entity embedding and the average word embedding of the mention itself; this gives us roughly a further 4% accuracy for both Wikipedia and Gigaword embeddings. Using the surface form recovers several cases where entities alone yield unlikely options, e.g. Roman-era Britain instead of the United Kingdom for Britain. However, it also introduces lexical errors: for example, British in several cases refers to the United Kingdom, but the British people are often selected instead. We note that this extra score actually hurts performance on MSH WSD, where the terms are curated to be highly ambiguous, in contrast to the shorter contexts and clearer terms used in AIDA.

Two other issues bear consideration in this evaluation. Prior approaches to the AIDA dataset, including MPME, make use of the global context of entity mentions within a document to improve predictions; by using local context only, we observe some inconsistent predictions, such as selecting the cricket world cup instead of the FIFA com-
petition for world cup, in a document discussing football. Additionally, in contrast to the MSH WSD dataset, many instances in AIDA have several highly-related candidates that introduce some confusion in our results. For example, Ireland could refer to the United Kingdom of Great Britain and Ireland, the island of Ireland, or the Republic of Ireland. As our embedding training does not include gold entity links, cases like this are often errors in our predictions.

5 Analysis of joint embeddings

To get a more detailed picture of our joint embedding space, we investigate nearest neighbors for each point by cosine similarity. As entities in the UMLS are assigned one or more of over 120 semantic types, we first examine how intermixed these types are in our biomedical embeddings. Figure 1 shows how often an entity’s nearest neighbor shares at least one semantic type with it, across the three biomedical embedding methods we evaluated. As each set of embeddings has a different vocabulary, we also restrict to the entities that all three can embed (approximately 11,000).

We see that our method puts entities of the same type together nearly 40% of the time, despite embedding over 270 thousand entities. On an even footing, our method puts types together significantly more often Mencia et al. (2016) (McNemar’s; $p < 0.05$), and equivalently with De Vine et al. (2014), despite using less entity-level information in training. Within our embeddings, major biological types such as bacteria, eukaryotes, mammals, and viruses all have more than 60% of neighbors with the same type, while less structured clinical types such as Clinical Attribute and Daily or Recreational Activity are in the 10-20% range. Corpus polysemy does not appear to have any effect on this type matching (mean polysemy of 1.5 for both matched and non-matched entities).

Expanding to include the words and terms in the joint embedding space, however, we see definite qualitative effects of corpus polysemy on entity nearest neighbors. Table 10 gives nearest word, term, entity, and joint neighbors to two biomedical entities: C0009443 (the common cold; $CP = 6.71$) and C0242797 (home health aides; $CP = 1$). For the more polysemous C0009443, where 95% of its mentions are of the word “cold” (polysemy=7), word-level neighbors are mostly nonsensical, while term neighbors are more logical, and entity neighbors reflect different senses of “cold”. By contrast, the non-polysemous C0242797, which is represented by 14 different unambiguous strings, words, terms, and entities are all very clearly in line with the theme of home health aides. Notably, the common and unambiguous terms for C0242797 are its nearest neighbors out of all points, while only two of the top 10 neighbors to C0009443 are terms.

<table>
<thead>
<tr>
<th>Entity</th>
<th>Words</th>
<th>Terms</th>
<th>Entities</th>
<th>Joint</th>
</tr>
</thead>
<tbody>
<tr>
<td>C0009443 (common cold)</td>
<td>k(+)-grown legionella-contaminated hyperinflating</td>
<td>cold short periods changed</td>
<td>C0041912 (upper respiratory infections) C0234192 (cold sensation) C0719425 (“Cold” pharmaceutical brand)</td>
<td>C0041912 (upper respiratory infections) C0234192 (cold sensation) C0719425 (“Cold” pharmaceutical brand)</td>
</tr>
<tr>
<td>C0242797 (home health aides)</td>
<td>homemaker-home voluntary-sector health/social</td>
<td>home health aide home health aides home health</td>
<td>C1553498 (home health encounter) C0019855 (home care services) C1317851 (home health care specialty)</td>
<td>home health aide home health aides</td>
</tr>
</tbody>
</table>

Table 10: Top 3 nearest neighbors to two UMLS entities, using words, terms, entities, or all three.

Figure 1: Percentage of UMLS entities whose nearest neighbor shares a semantic type, with no vocabulary restriction (vocab size in parentheses) and in a shared vocabulary subset.
6 Discussion

Faruqui et al. (2016) observe that similarity and relatedness are not clearly distinguished in semantic embedding evaluations, and that it is unclear exactly how vector-space models should capture them. We see more evidence of this, as cosine similarity seems to be capturing a mix of the two properties in our data. This mix is clearly informative, but it empirically favors relatedness judgments, and cosine similarity is insufficient to separate the two properties.

Corpus polysemy plays a qualitative role in our embedding model, but less of a quantitative one. It does not correlate with similarity and relatedness judgments or entity disambiguation decisions, but it clearly affects the organization of the embedding space, by embedding entities with high corpus polysemy in less coherent areas than those with low polysemy. Linzen (2016) points out that for analogy completion, local neighborhood structure can interfere with standard methods; how this neighborhood structure affects predictions in more complex tasks is an open question.

Overall, we find two main advantages to our model over prior work. First, by only using a terminology and an unannotated corpus, we are able to learn entity embeddings from larger and more diverse data; for example, embeddings learned from Gigaword (which has no entity annotations) outperform embeddings learned on Wikipedia in most of our experiments. Second, by embedding entities and text into a joint space, we are able to leverage complementary information to get higher performance in both intrinsic and extrinsic tasks; an oracle model nearly matches a state-of-the-art ensemble vector and knowledge-based model for biomedical word sense disambiguation. However, our other entity disambiguation results demonstrate that there is additional entity-level information that we are not yet capturing. In particular, it is unclear whether our low performance on disambiguating newswire entities is due to a disambiguation model mismatch, a lack of information in our embeddings, or a combination of both.

7 Conclusions

We present a method for jointly learning embeddings of entities and text from an arbitrary unannotated corpus, using only a terminology for distant supervision. Our learned embeddings better capture both biomedical and encyclopedic similarity and relatedness than prior methods, and approach state-of-the-art performance for unsupervised biomedical word sense disambiguation. Furthermore, entities and words learned jointly with our model capture complementary information, and combining them improves performance in all of our evaluations. We make an implementation of our method available at github.com/OSU-slatelab/JET, along with the source code used for our evaluations and our pretrained entity embeddings. Our novel Wikipedia similarity and relatedness datasets are available at the same source.

Acknowledgments

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Serguei V S Pakhomov, Greg Finley, Reed McEwan, Yan Wang, and Genevieve B Melton. 2016. Corpus domain effects on distributional semantic modeling of medical terms. Bioinformatics, 32(August):btw529.


A WikiSRS construction details

We followed a similar process to Pakhomov et al. (2010) in selecting the entity pairs to be used in our dataset. We first filtered the full list of Wikipedia pages to the subset that we learned embeddings for, and then used the entity types assigned to these pages in YAGO (Mahdisoltani et al., 2015) to restrict to only entities labeled with WordNet types organization or person, or with the YAGO type geoEntity. For each pairing of these categories (Organization-Organization, Organization-Place, Organization-Person, Place-Place, Place-Person, and Person-Person), we manually selected 30 pairs of entities for each of the following relatedness categories: Completely Unrelated, Somewhat Unrelated, Somewhat Related, and Highly Related. These produced the list of 720 entity pairs we used for our Mechanical Turk surveys.

We augmented each survey of 30 questions with 4 manually-created validation pairs using common entities (e.g., London, New York), each of which was categorized as Highly Related or Completely Unrelated. We included these validation questions at random indices in our surveys. To evaluate if participants were reading the questions, we binned their ratings on these validation questions into 0-25 (Completely Unrelated), 26-50 (Somewhat Unrelated), 51-75 (Somewhat Related), and 76-100 (Highly Related). If a participant’s ratings disagreed with ours on multiple validation questions, we discarded their data (we allowed disagreement on a single question, as some validation questions had high variance in responses among reliable annotators).

We recruited 6 participants for each survey, for a total of 34 unique participants across the 48 HITs. Participants were presented with a message describing the survey and stating that by clicking the button at the bottom of the message to begin the survey, they were providing informed consent to participate. Identifying participant data was not collected, and we used only the anonymous worker IDs provided by the Mechanical Turk interface to collate our data and remunerate workers. Participants were asked optional demographic questions about their age bracket and native language at the end of the survey; we did not end up using age information, but filtered our participants for those that self-reported English reading proficiency. The majority responded to a single HIT, while 3 completed more than 20. We discarded all submissions from 3 participants, as they did not report English reading proficiency (1) or did not satisfy the validation questions (2). All participants were paid state minimum wage at the time of the study for their time, regardless of whether they answered demographic questions or if we used their data in the final sample. Collection of this data was approved under Ohio State University IRB protocol 2017E0050.

To generate the final dataset, we assessed each participant’s responses to the validation questions in each survey. We kept surveys for which we had at least 4 participants with satisfactory answers to the validation questions; this resulted in discarding 1 of the 24 HITs for each task. Due to 2 repeated pairs, this gave us final dataset sizes of 688 pairs for each of similarity and relatedness, 658 of which were shared between the tasks.

Following Pakhomov et al. (2010), we assessed inter-annotator agreement using the intraclass correlation coefficient (ICC). Table 11 gives the values for our datasets. The numbers reported are within the moderate range, and they correspond to the ICC numbers reported by Pakhomov et al. on the UMNSRS datasets.

The source code of our Mechanical Turk interface and data files used to generate the tasks are available at github.com/OSU-slatelab/WikiSRS.

<table>
<thead>
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<th># of raters</th>
<th>Similarity ICC</th>
<th># pairs</th>
<th>Relatedness ICC</th>
<th># pairs</th>
</tr>
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<td>419</td>
<td>0.467</td>
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<tr>
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<td>299</td>
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<tr>
<td>&gt; 6</td>
<td></td>
<td>2</td>
<td></td>
<td>2</td>
</tr>
<tr>
<td>Total</td>
<td></td>
<td>688</td>
<td></td>
<td>688</td>
</tr>
</tbody>
</table>

Table 11: The intraclass correlation coefficient (ICC) among Amazon Mechanical Turk worker judgments of similarity and relatedness of pairs of Wikipedia entities. As ICC requires a fixed number of raters, but we had variable numbers of responses to each HIT, we break down the datasets by the number of workers who rated each item.
A Sequence-to-Sequence Model for Semantic Role Labeling

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Abstract

We explore a novel approach for Semantic Role Labeling (SRL) by casting it as a sequence-to-sequence process. We employ an attention-based model enriched with a copying mechanism to ensure faithful regeneration of the input sequence, while enabling interleaved generation of argument role labels. Here, we apply this model in a monolingual setting, performing PropBank SRL on English language data. The constrained sequence generation set-up enforced with the copying mechanism allows us to analyze the performance and special properties of the model on manually labeled data and benchmarking against state-of-the-art sequence labeling models. We show that our model is able to solve the SRL argument labeling task on English data, yet further structural decoding constraints will need to be added to make the model truly competitive. Our work represents a first step towards more advanced, generative SRL labeling setups.

1 Introduction

Semantic Role Labeling (SRL) is the task of assigning semantic argument structure to constituents or phrases in a sentence, to answer the question: Who did what to whom, where and when? This task is normally accomplished in two steps: first, identifying the predicate and second, labeling its arguments and the roles that they play with respect to the predicate. SRL has been formalized in different frameworks, the most prominent being FrameNet (Baker et al., 1998) and PropBank (Palmer et al., 2005). In this work we focus on argument identification and labeling using the PropBank (PB) annotation scheme.

Recent end-to-end neural models considerably improved the state-of-the-art results for SRL in English (He et al., 2017; Marcheggiani and Titov, 2017). In general, such models treat the problem as a supervised sequence labeling task, using deep LSTM architectures that assign a label to each token within the sentence.

SRL training resources for other languages are more restricted in size and thus, models suffer from sparseness problems because specific predicate-role instances occur only a handful of times in the training set. Since annotating SRL data in larger amounts is expensive, the use of a generative neural network model could be beneficial for automatically obtaining more labeled data in low-resource settings. The model that we present in this paper is a first step towards a joint label and language generation formulation for SRL, using the sequence-to-sequence architecture as a starting point.

We explore a sequence-to-sequence formulation of SRL that we apply, as a first step, in a classical monolingual setting on PropBank data, as illustrated in Figure 1. This constrained monolingual setting will allow us to analyze the suitability of a sequence-to-sequence architecture for SRL, by benchmarking the system performance against existing sequence labeling models for SRL on well known labeled evaluation data.
Sequence-to-sequence (seq2seq) models were pioneered by Sutskever et al. (2014), and later enhanced with an attention mechanism (Bahdanau et al., 2014; Luong et al., 2015). They have been successfully applied in many related structure prediction tasks such as syntactic parsing (Vinyals et al., 2015), parsing into Abstract Meaning Representation (Konstas et al., 2017), semantic parsing (Dong and Lapata, 2016), and cross-lingual Open Information Extraction (Zhang et al., 2017).

When applying a seq2seq model with attention in a monolingual SRL labeling setup, we need to restrict the decoder to reproduce the original input sentence, while in addition inserting PropBank labels into the target sequence in the decoding process (see Figure 1). To achieve this, we encode each input sentence into a suitable representation that will be used by the decoder to regenerate word tokens as given in the source sentence and introducing SRL labels in appropriate positions to label argument spans with semantic roles. In order to avoid lexical deviations in the output string, we add a copying mechanism (Gu et al., 2016) to the model. This technique was originally proposed to deal with rare words by copying them directly from the source when appropriate. We apply this mechanism in a novel way, with the aim of guiding the decoder to reproduce the input as closely as possible, while otherwise giving it the option of generating role labels in appropriate positions in the target sequence.

Our main contributions in this work are:

(i) We propose a novel neural architecture for SRL using a seq2seq model enhanced with attention and copying mechanisms.

(ii) We evaluate this model in a monolingual setting, performing PropBank-style SRL on standard English datasets, to assess the suitability of this model type for the SRL labeling task.

(iii) We compare the performance of our model to state-of-the-art sequence labeling models, including detailed (also comparative) error analysis.

(iv) We show that the seq2seq model is suited for the task, but still lags behind sequence labeling systems that include higher-level constraints.

2 Model

We propose an extension to the Sequence-to-Sequence model of (Bahdanau et al., 2014) to perform SRL. The model will learn to map an unlabeled source sequence of words \(x_{1:T_x}\) into a target sequence \(y_{1:T_y}\) consisting of word tokens and SRL label tokens (see Figure 2). The source sentence, represented as a sequence of dense word vectors, is fed to an LSTM encoder to produce a series of hidden states that represent the input. This information is used by the decoder to recursively generate tokens step-by-step, conditioned on the previous generated tokens and the source by attending the encoder’s hidden states as proposed in Bahdanau et al. (2014). On top of this architecture, we add the copying mechanism (Gu et al., 2016), which helps the model to avoid lexical deviations in the output while still having the freedom of generating words and SRL labels based on the context. The attention-based generation and copying mechanism will be competing with each other so that the model learns when to copy directly from the source and when to generate the next token.

In our current setup we restrict role labeling to a single predicate per sentence. If a sentence has more than one predicate, we create a separate copy for each predicate; the same setting was applied in Zhou and Xu (2015). In each sentence copy the predicate whose roles are to be labeled is preceded by a special token \(<PRED>\) that marks the position of the predicate under consideration. This helps the decoder to focus on generating argument labels for that specific predicate (see Table 1.)

2.1 Vocabulary

We assume a unique vocabulary for both encoder and decoder that comprises the words occurring during training, the out-of-vocabulary token, and the special symbol used to mark the position of the predicate, thus \(V = \{v_1, ..., v_N\} \cup \{UNK, <PRED>\}\). In addition, we employ a set \(L = \{l_1, ..., l_M\}\) with all the possible labeled brackets and a set \(X = \{x_1, ..., x_{T_x}\}\), a per-instance set containing the \(T_x\) words from the current source sequence. Thus, our total vocabulary is defined for each instance as \(V \cup L \cup X\).

The label set \(L\) contains one common opening bracket (\# for all argument types to indicate the beginning of an argument span, and several label-specific closing brackets, such as \(PO:A1\), which indicates in this case that the span for argument \(A1\) is ending (see also Table 1).
2.2 Encoder

We use a two-layer bi-RNN encoder with LSTM cells (Hochreiter and Schmidhuber, 1997) that outputs a series of hidden states \( h_j = [h_j^1; h_j^2] \) where each \( h_j \) contains information about the surrounding context of the word \( x_j \). We refer to the complete matrix of encoder hidden states as \( M \), since it acts as a memory that the decoder can use to copy words directly from the source.

2.3 Attention Mechanism

We use the global dot product attention from Luong et al. (2015) to compute the context vector \( c \).

2.4 Decoder

The role of the decoder (a single-layer recurrent unidirectional LSTM) is to emit an output token \( y_t \) from a learned distribution over the vocabulary at each time step \( t \) given its state \( s_t \), the previous output token \( y_{t-1} \), the attention context vector \( c_t \), and the memory \( M \). To get this distribution it is necessary to compute two separate modes: one for generating and one for copying.

To obtain the probability of generating \( y_t \) we use the context vector produced by the attention to learn a score \( \psi_g \) for each possible token \( v_i \) of being the next generated token. We define \( \psi_g \) as:

\[
\psi_g(y_t = v_i) = W_v[s_t; c_t], \quad v_i \in \mathcal{V} \cup \mathcal{L} \quad (2)
\]

where \( W_v \in \mathbb{R}^{N \times 2d_e} \) is a learnable parameter and \( s_t, c_t \) are the current decoder state and context vector respectively. This means that the model computes a generation score for both words and labels, based on what it is attending on at the current step.

For the probability of copying \( y_t \) we compute the score \( \psi_c \) of copying a token directly from the source as:

\[
\psi_c(y_t = x_j) = \sigma(h_j^T W_c)s_t, \quad x_j \in \mathcal{X} \quad (3)
\]

where \( W_c \in \mathbb{R}^{d_h \times d_e} \) is a learnable parameter, \( h_j \) is the encoder hidden state representing \( x_j \), \( s_t \) is the current decoder state, and \( \sigma \) is a non-linear transformation; we used \( \tanh \) for our experiments.

Using the two scoring methods, the decoder will have two competing modes: the generation mode, used to generate the most probable subsequent token based on attention; and the copying, used to choose the next token directly from the encoder memory \( M \), which holds both positional and content information of the source. A final mixed distribution is calculated by adding the probability of generating \( y_t \) and the probability of copying \( y_t \):

\[
p(y_t \mid s_t, y_{t-1}, c_t, M) = p(y_t \mid g \mid s_t, y_{t-1}, c_t) + p(y_t \mid c \mid s_t, y_{t-1}, M) \quad (4)
\]

We use a softmax layer to convert the two scores into a joint distribution that represents the mixed
Table 1: A single sentence with three labeled predicates is converted into three different source-target pairs. The symbol $<_{PRED}>$ in each source marks the predicate for which the model is expected to generate a correct predicate-argument structure.

Table 1: A single sentence with three labeled predicates is converted into three different source-target pairs. The symbol $<_{PRED}>$ in each source marks the predicate for which the model is expected to generate a correct predicate-argument structure.

$$p(y_t, g|\cdot) = \begin{cases} \frac{1}{Z} e^{\psi_g(y_t)} & y_t \notin V \cup L \\ 0 & \text{otherwise} \end{cases}$$

$$p(y_t, c|\cdot) = \begin{cases} \frac{1}{Z} \sum_{j} e^{\psi_c(x_j)} & y_t \in X \\ 0 & \text{otherwise} \end{cases}$$

(5)

where $Z$ is the normalization term shared by the two modes, $Z = \sum_{v \in V} e^{\psi_g(v)} + \sum_{x \in X} e^{\psi_c(x)}$. Since a single softmax is applied over the copying and generating modes, the network learns by itself when it is proper to copy a word from the source and when it needs to generate a label.

During training, the objective is to minimize the negative log-likelihood of the target token $y_t$ for each time-step for both generate mode (given previous generated tokens) and copy mode (given source sequence $X$). We calculate the loss for the whole sequence as:

$$loss = - \frac{1}{T_y} \sum_{t=0}^{T_y} \log P(y_t | y_{<t}, X)$$

(6)

3 Experimental Setup

3.1 Datasets and Evaluation Measures

We test the performance of our system on the span-based SRL datasets CoNLL-05\(^2\) and CoNLL-12.\(^3\) These datasets provide the gold predicate as part of the input. Since we focus on argument identification and classification, we provide this information in the input to the system. We use the standard training, development

\(^2\)http://www.lsi.upc.edu/~srlconll/home.html

\(^3\)http://conll.cemantix.org/2012/data.html

and test splits and use the official CoNLL-05 evaluation script on both datasets. We compare our results with Collobert et al. (2011); FitzGerald et al. (2015); Zhou and Xu (2015) and He et al. (2017) who use the same datasets and evaluation script. We show results separately for the Brown and WSJ portion of the CoNLL-05 test dataset.

The CoNLL-05 Shared Task\(^4\) evaluation script computes precision, recall and F1 measure (the harmonic mean of precision and recall) for the predicted arguments. The script expects prediction-gold pairs that have the same number of words in order to consider them comparable, and only if this is the case, it computes a score. Furthermore, an argument is only considered correct if the words spanning the argument as well as its role label match with gold (Carreras and Márquez, 2005). This means that it is essential to predict perfect argument spans besides the correct role label.

3.2 Pre-processing

For our seq2seq model we need to provide sources and targets in a linearized manner. The sequences are sentences with zero or more predicates. Following Zhou and Xu (2015), if a sentence has $n_p$ predicates we process the sentence $n_p$ times, each one with its corresponding predicate-argument structure. As shown in Table 1, we linearize the target side by converting the CoNLL format into sequences of tokens that include brackets indicating the span of the argument and the argument label on the closing bracket. We inform the model about the predicate that it should focus on by adding the special token $<_{PRED}>$ to the source sequence immediately before the predicate word.

This process is entirely reversible and thus we convert the system outputs back to CoNLL format and evaluate the results with the official script.

\(^4\)http://www.lsi.upc.edu/~srlconll/soft.html
### 3.3 Training

Since we process as many copies of sentences as it has predicates, the final amount of sequences is approximately 94K for CoNLL-05 and 185K for CoNLL-12 training sets. We keep linearized sequences up to 100 tokens long and lowercase all tokens. Given this limit, we omit 30 (CoNLL-05) and 900 (CoNLL-12) sequences from training. We initialize the model with pre-trained 100-dimensional GloVe embeddings (Pennington et al., 2014) and update them during training. All the tokens that are not covered by GloVe or that appear less frequently than a given threshold in the training dataset are mapped to the UNK embedding. We keep track of this mapping to be able to post-process the sequence and recover the rare tokens. Our vocabulary size is set to $|V| \approx 20K$ words for CoNLL-05 and $|V| \approx 18K$ words for CoNLL-12.

We use Adam optimizer (Kingma and Ba, 2014), a learning rate $\ell_r = 0.001$ and gradient clipping at 5.0. Both encoder and decoder have hidden layer of 512 LSTMs. We use dropout (Srivastava et al., 2014) of 0.4 and train for 4 epochs with batch size of 6.

### 4 Evaluation and Results

Initially, we trained a model using attention only, and it learned to generate balanced brackets (every opening bracket has a corresponding closing bracket within the sequence) without further constraints. Yet, due to its generative nature, many target sequences diverged from the source in both length and token sequences. This was expected, because the system has to learn to generate not only the labels at the correct time-step but also to re-generate the complete sentence accurately. This is a disadvantage compared to the sequence labeling models where the words are already given.

By adding copying mechanism the model successfully regenerates the source sentence in the majority (up to 99%) of cases, as shown in Table 2. Such behavior also enables us to measure the performance of the model as an argument role classifier against the gold standard. Thus, we can benchmark its labeling performance against previous architectures built to solve the SRL task.

Table 3 displays the overall labeling performance of our copying-enhanced seq2seq model in comparison to previous neural sequence labeling architectures. For sequences that do not fully reproduce the input, we cannot compute appropriate scores against the gold standard. We compute two alternative scores for these cases: oracle-min, by setting the score for these sentences to 0.0 F1, and oracle-max, by setting their results to the scores we would obtain with perfect (= gold) labels. With these scores, we can better estimate the loss we are experiencing by non-perfectly reproduced sequences (see Table 2.)

As seen in Table 3, our model achieves an F1 score of 76.05 on the CoNLL-05 development set, and 73.4 on CoNLL-12 (min-oracle), and 77.29 and 75.05 (max-oracle), respectively. While these scores are still low compared to the latest neural SRL architectures, they are above the relatively simple model of Collobert et al. (2011).
that in contrast to the stronger models of FitzGerald et al. (2015); Zhou and Xu (2015) and He et al. (2017), our architecture is very lean and does not (yet) employ structured prediction (e.g., Conditional Random Field), to impose structural constraints on the label assignment. While this is certainly an extension we are going to explore in future work, here we will conduct deeper investigation to learn more about the kind of errors that our unconstrained seq2seq model makes. We report the analysis on CoNLL-05 development set.

4.1 Analysis

Argument Spans The model needs to generate labeled brackets at the appropriate time-step, in other words, the prediction of correct spans for arguments. To verify how well it is doing this, we measure how much overlap exists between the generated spans and the gold ones. This is equivalent to computing unlabeled argument assignment. We found that 77.5% of the spans match the gold spans completely, 21.2% of spans are partially overlapping with gold spans, and only 1.2% of the spans do not overlap at all with gold.

Argument Labels Recall from Section 2 that our model is labeling the sentences as in a translation task. It learns to use information from relevant words in the source sequence, aligning the labels to the argument words via learned attention weights as it is shown in Figure 3. This allows us to see where the model is looking when generating the labeled bracket. The confusion matrix in Figure 4 shows predicted vs. gold labels for all correctly assigned argument spans (i.e., the spans that match the gold boundaries). We observe that the model does very well for A0 and A1 gold roles, and that it causes only few misclassifications for A2. However, it frequently predicts core argument roles A0–A3 for non-argument roles, and also tends to mix predictions among non-core arguments. Since A0 and A1 roles are most frequent in the data, this indicates that the seq2seq model would benefit from more training data, particularly for less frequent roles, to better differentiate roles, and this is more prominent for the ones that are marked with prepositions.

Role co-occurrence and role set constraints Despite the absence of more refined decoding constraints, our model learns to avoid generating duplicated argument labels in most of the sequences. We find duplicated argument labels in less than 1% of the sequences. Figure 5 shows that the majority (about 70%) of sentences do not involve any missing or excess arguments; about 24/20% of sentences experience a single missing/excess role, and only 5/4% of the sentences experience a higher amount of missed/excess roles. Overall,
Figure 6: Performance of the model based on the number of tokens that the sequence has.

Figure 7: F1 score of arguments in buckets of increasing distances from their predicate, with distance normalized by sentence length (CoNLL-05, dev). We compare our model with He et al. (2017).

missed vs. excess arguments are balanced.

**Sequence Length** Another characteristic of the seq2seq model is that it encodes within a single sequence both words and labeled brackets. This increases the length of the sequences that need to be processed, and it is a well known problem that sequence length affects performance of recurrent neural models, even with the use of attention.

To measure the labeling performance difficulty with increasing sequence length, we partitioned the system outputs in six different bins containing groups of sentences of similar length (see Figure 6). As expected, the F1 score degrades proportionally to the length of the sequence, especially in sentences with more than 30 tokens.

**Distance to predicate** He et al. (2017) show that the surface distance between the argument and the predicate is also proportional to the amount of labeling errors. In our model, the distance between argument words and the predicate is even longer because of labeled brackets embedded in the sequence. Figure 7 displays the F1 score for different token distances between predicate and the respective argument. We see that the seq2seq model follows the same trend as the sequence labeling model, despite the fact that our model has access to the hidden states from the encoded input sentence; however, the real distance between predicate and argument in the decoder is also bigger.

**Distance from sentence beginning.** With each token that the model generates in decoding, the distance to the end position of the encoded sentence representation grows. While intuitively we would expect the model performance to degrade with larger distance to the input, it is also true that the model could be more prone to making mistakes at the beginning of the sequence, when the decoder has not yet generated enough context. To investigate this, we traced the ratio of errors that occur in several ranges of the sequence. We can see in Figure 8 that the first intuition was correct, the distance to the encoded representation is proportional to the mistakes that the model makes. We compare the error ratio to He et al. (2017) and show that the seq2seq system follows a similar trend but degrades faster with sequence length.

5 Related Work

**Semantic Role Labeling.** Traditional approaches to SRL relied on carefully designed features and expensive techniques to achieve global consistency such as Integer Linear Programming (Punyakanok et al., 2008) or dynamic programming
(Täckström et al., 2015). First neural SRL attempts tried to mix syntactic features with neural network representations. For example, FitzGerald et al. (2015) created argument and role representations using a feed-forward NN, and used a graphical model to enforce global constraints. Roth and Lapata (2016), on the other hand, proposed a neural classifier using dependency path embeddings to assign semantic labels to syntactic arguments.

Collobert et al. (2011) proposed the first SRL neural model that did not depend on hand-crafted features and treated the task as an IOB sequence labeling problem. Later, Zhou and Xu (2015) proposed a deep bi-directional LSTM model with a CRF layer on top. This model takes only the original text as input and assigns a label to each individual word in the sentence. He et al. (2017) also treat SRL as a IOB tagging problem, and use again a deep bi-LSTM incorporating highway connections, recurrent dropout and hard decoding constraints together with an ensemble of experts. This represents the best performing system on two span-based benchmark datasets so far (namely, CoNLL-05 and CoNLL-12). Marcheggiani et al. (2017) show that it is possible to construct a very accurate dependency-based SRL system without using any kind of explicit syntactic information. In subsequent work, Marcheggiani and Titov (2017) combine their LSTM model with a graph convolutional network to encode syntactic information at word level, which improves their LSTM classifier results on the dependency-based benchmark dataset (CoNLL-09).

**Sequence-to-sequence models.** Seq2seq models were first discovered as powerful models for Neural Machine Translation (Sutskever et al., 2014; Cho et al., 2014) but soon proved to be useful for any kind of problem that could be represented as a mapping between source and target sequences. Vinyals et al. (2015) demonstrate that constituent parsing can be formulated as a seq2seq problem by linearizing the parse tree. They obtain close to state-of-the-art results by using a large automatically parsed dataset. Dong and Lapata (2016) built a model for a related problem, semantic parsing, by mapping sentences to logical form. Seq2seq models have also been widely used for language generation (e.g. Karpathy and Li (2015); Chisholm et al. (2017)) given their ability to produce linguistic variation in the output sequences.

More closely related to SRL is the AMR parsing and generation system proposed by Konstas et al. (2017). This work successfully constructs a two-way mapping: generation of text given AMR representations as well as AMR parsing of natural language sentences. Finally, Zhang et al. (2017) went one step further by proposing a cross-lingual end-to-end system that learns to encode natural language (i.e. Chinese source sentences) and to decode them into sentences on the target side containing open semantic relations in English, using a parallel corpus for training.

6 Conclusions

In this paper we explore the properties of a Sequence-to-Sequence model for identifying and labeling PropBank roles. This is motivated by the fact that using a seq2seq model gives more flexibility for further tasks such as constrained generation and cross-lingual label projection. Another advantage is that our model is a very lean architecture compared to the deep Bi-LSTM of the recent SRL models.

To our knowledge, this is the first attempt to perform SRL using a seq2seq approach. Specific challenges emerged by formulating the problem in this way, such as: (i) the decoding of labels and words within a single sequence; (ii) generating balanced labeled brackets at the correct position; (iii) avoiding repetition of tokens, and especially, (iv) generating labeled sequences that perfectly match the source sentence in order to make the labeled sequence absolutely comparable. Despite these difficulties, we could show that a sequence-to-sequence model with attention and copying achieves quite respectable labeling performance with a lean architecture and without yet considering structural constraints. For future work we consider extensions towards joint semantic role labeling and constrained generation, to produce new variations of existing labeled data.

Acknowledgements

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Predicting Concreteness and Imageability of Words
Within and Across Languages via Word Embeddings

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Abstract
The notions of concreteness and imageability, traditionally important in psycholinguistics, are gaining significance in semantic-oriented natural language processing tasks. In this paper we investigate the predictability of these two concepts via supervised learning, using word embeddings as explanatory variables. We perform predictions both within and across languages by exploiting collections of cross-lingual embeddings aligned to a single vector space. We show that the notions of concreteness and imageability are highly predictable both within and across languages, with a moderate loss of up to 20% in correlation when predicting across languages. We further show that the cross-lingual transfer via word embeddings is more efficient than the simple transfer via bilingual dictionaries.

1 Introduction
Concreteness and imageability are very important notions in psycholinguistic research, building on the theory of the double, verbal and non-verbal, modality of representation of concrete words in the mental lexicon, contrasted to single verbal representation of abstract words (Paivio, 1975, 2010). Although often correlated with concreteness, imageability is not a redundant property. While most abstract things are hard to visualize, some call up images, e.g., torture calls up an emotional and even visual image. There are concrete things that are hard to visualize too, for example, abbey is harder to visualize than banana (Tsvetkov et al., 2014).

Both notions have proven to be useful in computational linguistics as well. Turney et al. (2011) present a supervised model that exploits concreteness to correctly classify 79% of adjective-noun pairs as having literal or non-literal meaning. Tsvetkov et al. (2014) exploit both the notions of concreteness and imageability to perform metaphor detection on subject-verb-object and adjective-noun relations, correctly classifying 82% and 86% instances, respectively.

The aim of this paper is to investigate the predictability of concreteness and imageability within a language, as well as across languages, by exploiting cross-lingual word embeddings as our available signal.

2 Related Work
While much work has been done on exploiting word embeddings in expanding sentiment lexicons (Tang et al., 2014; Amir et al., 2015; Hamilton et al., 2016), there is little work on predicting other lexical variables, concreteness and imageability included.

Tsvetkov et al. (2014) performed metaphor detection, using, among others, concreteness and imageability as their features. To propagate these features, obtained from the MRC psycholinguistic database (Wilson, 1988) to the entire lexicon, they used a supervised learning algorithm on vector space representations, where each vector element represented a feature. Performance of these classifiers was 0.94 for concreteness and 0.85 for imageability. They also applied the concreteness and imageability features to other languages by projecting features with bilingual dictionaries.

Broadwell et al. (2013) extended imageability scores to the whole lexicon by using the MRC...
imageability scores and hyponym and hyperonym links from WordNet.

Rothe et al. (2016) trained an orthogonal transformation to reorder word embedding dimensions into one-dimensional ultradense subspaces, the output thereby being a lexicon. They trained the transformations for sentiment, concreteness and frequency. For obtaining training data for concreteness, they used the BWK database (Brysbaert et al., 2014). They showed that concreteness and sentiment can be better extracted from embedding spaces than frequency, with a Kendall $\tau$ correlation coefficient of 0.623 for concreteness. Rothe and Schütze (2016) further exploited this method to perform operations over the extracted dimensions, such as given a concrete word like friends, find the related, but abstract word friendship.

**Contributions** In this paper we perform a systematic investigation of transfer of two lexical notions, concreteness and imageability, (1) to the remainder of the lexicon not covered in an annotation campaign, and (2) to other languages.

While there were already successful transfers within a language based on word embeddings (Tsvetkov et al., 2014; Rothe and Schütze, 2016), the only cross-lingual transfer was based on transfer via bilingual dictionaries (Tsvetkov et al., 2014). In this paper we compare the effectiveness of cross-lingual transfer via word embeddings and via bilingual dictionaries.

A byproduct of this research is a lexical resource in 77 languages containing per-word estimates for concreteness and imageability.

### 3 Data

#### 3.1 Lexicons

In our experiments we use two existing English and one Croatian lexicon with concreteness and imageability ratings.

For English we use the MRC database (Wilson, 1988) (MRC onwards), consisting of 4,293 words with ratings for concreteness and imageability. The ratings range from 100 to 700 and were obtained by merging three different resources (Wilson, 1988).

We also use the BWK database consisting of 39,954 English words (Brysbaert et al., 2014) (BWK onwards) with concreteness ratings summarized through arithmetic mean and standard deviation. The ratings were collected in a crowdsourcing campaign in which each word was labeled by 20 annotators on a 1–5 scale.

For Croatian we use the MEGAHR database (MEGA onwards), consisting of 3,000 words, with concreteness and imageability ratings summarized through arithmetic mean and standard deviation. The ratings were collected in an annotation campaign among university students, with each word obtaining 30 annotations per variable on a 1–5 scale.

For performing cross-lingual transfer via a dictionary, we use data from a large popular online Croatian-English dictionary containing around 100 thousand entries.

#### 3.2 Embeddings

For both in-language and cross-lingual experiments we use the aligned Facebook collection of embeddings, trained with fastText (Bojanowski et al., 2016) on Wikipedia dumps, with embedding spaces aligned between languages with a linear transformation learned via SVD (Smith et al., 2017) on a bilingual dictionary of 500 out of the 1000 most frequent English words, obtained via the Google Translate API.

We also experimented with another cross-lingual embedding collection (Conneau et al., 2017), obtaining similar results and backing all our conclusions. This is in line with recent work on comparing cross-lingual embedding models which suggests that the actual choice of monolingual and bilingual signal is more important for the final model performance than the actual underlying architecture (Levy et al., 2017; Ruder et al., 2017). Given that one of our goals is to transfer concreteness and imageability annotations to as many languages as possible, using cross-lingual word embeddings based on Wikipedia dumps and dictionaries obtained through a translation API is the most plausible option.

### 4 Experiments

#### 4.1 Setup

We perform two sets of experiments: one within each language, and another across languages.
While in-language experiments are always based on supervised learning, in cross-lingual experiments we compare two transfer approaches: one based on a simple dictionary transfer, and another on supervised learning on the word embeddings in the source language, and performing predictions on word embeddings in the target language, with the two embedding spaces being aligned.

We perform our prediction experiments by training SVM regression models (SVR) and deep feedforward neural networks (FFN) over standardized (zero mean, unit variance) embeddings and each specific response variable. We experiment with all available gold annotations as our response variables, namely both the arithmetic mean and standard deviation of concreteness and imageability.

We tuned the hyperparameters of each of the regressors on a subset of the Croatian, MEGA dataset in the case of the in-language experiments, and another subset of the BWK dataset for the cross-lingual experiments. Given that we perform the final experiments on the whole datasets, and that we have two additional English datasets at our disposal for the in-language experiments and three additional dataset pairs for the cross-lingual experiments, we consider our approach to be resistant to the overfitting of the hyperparameters going unnoticed.

While the SVR proved to work well with the RBF kernel, the $C$ hyperparameter of 1.0 and the $\gamma$ hyperparameter of 0.003, the feedforward network obtained strong results with two fully-connected hidden layers, consisting of 128 and 32 units each and ReLU activation functions, with a dropout layer after each of the hidden layers, and an output layer with a linear activation function. We optimized for the mean squared error loss function and ran 50 epochs on each of the datasets, with a batch size of 32.

While we used the same regressor setup for the SVR system for both the in-language and cross-lingual experiments, for the FFN system the dropout probability in the in-language experiments was 0.5, while in the cross-lingual setting the dropout probability was set to 0.8, obtaining thereby a more general model which transfers better to the other language.

We perform in-language experiments via 3-fold cross-validation, while we train models on our source language dataset and evaluate the models on our target language dataset for cross-lingual experiments. We evaluate each approach via the Spearman rank and Pearson linear correlation coefficients. In the paper we report the Spearman correlation coefficient only as the relationships across both metrics in all the experiments are identical. We perform our experiments with the scikit-learn (Pedregosa et al., 2011) and keras (Chollet et al., 2015) toolkits.

4.2 In-language Experiments

We start our experiments in the in-language setting, running cross-validation experiments over each of our three datasets on all available variables. The results of these experiments, with some basic information on the size of the datasets, are given in Table 1. Aside from the three lexicons introduced in Section 3.1, we experiment with another lexicon, BWK.3K, which is a randomly downsampled version of the BWK lexicon to the size of the two remaining lexicons. We introduce this additional resource (1) to control for dataset size when comparing results on our different datasets and (2) to measure the impact of training data size by comparing the results on the two flavours of the BWK dataset.

The results in Table 1 show that the support vector regressor consistently performs better than the feedforward neural network at predicting almost all values, with relative error reduction lying between 7% and 12%. The bold results are statistically significantly better than the corresponding non-bold ones given the approximate randomization test (Edgington, 1969) with $p < 0.05$. Our assumption is that the stronger FFN model does not show a positive impact primarily due to the small size of the datasets and the simplicity of the modeling problem.

We can further observe that the arithmetic mean is much easier to predict than standard deviation on both variables in all the datasets. This can be explained by the fact that standard deviation on the two phenomena can partially be explained with the level of ambiguity of a specific word, and this type of information is at least not directly available in context-based word embeddings.

Furthermore, imageability seems to be consistently slightly harder to predict than concreteness. Our initial assumption regarding this difference was that imageability is a more vague notion for
human subjects, and therefore their responses are more dispersed, adding to the complexity of the prediction. However, analyzing standard deviations over concreteness and imageability showed that these are rather the same. We leave this open question for future research.

When comparing the results on predicting mean concreteness on the full BWK and the trimmed BWK.3K datasets, we see a significant improvement of the predictions on the larger dataset, showing that having 10 times more data for learning can produce significant improvements in the prediction quality.

4.3 Cross-lingual Experiments

In cross-lingual experiments we compare our two approaches to cross-lingual transfer: dictionary lookup (DIC onwards) and supervised learning on aligned word embedding spaces via the two methods introduced in Section 4.2, SVR and FFN.

The DIC method simply looks up for each word in the source language resource all possible translations to the target language and directly transfers the concreteness and imageability ratings to the target language words. In case of collisions in the target language (two source language words being translated to the same word in the target language), we perform averaging over the transferred ratings. In our experiments, the arithmetic mean showed to be a better averaging method than the median, we therefore report the results on that averaging method.

The SVR and FFN methods use supervised learning in a very similar fashion to the in-language experiments described in Section 4.2. We train a supervised regression model on the whole source language dataset, using word embedding dimensions as features and the variable of choice as our target. We obtain estimates of our variable of choice in the target language by applying the source-language model on the target-language word embeddings since the two embedding spaces are aligned.

For both approaches we compare the target-language estimates with the gold data available from our lexicons.

We present the results of the cross-lingual experiments in Table 2. Our first observation is that, while in the in-language setting the SVR method has regularly outperformed the FFN method, in the cross-lingual setting this is not the case any more, with SVR and FFN obtaining very similar results, in five out of six cases in the range of no statistically significant difference. Our explanation for the loss of the positive impact in using the weaker, support vector regression model, is that with the noisy alignment of the two embedding spaces the prediction problem became harder, now both models performing similarly. While the strong point of SVR is that it performs very well on small datasets, the strong point of the FFN method is that it generalizes better.

That higher generalization is beneficial in case of the cross-lingual problem is observable in the difference in the hyperparameter tuning results on the FFN method, where in the in-language setting the optimal dropout was 0.5, while in the cross-lingual setting it is 0.8.

Our second observation is that all the predicted ratings suffer in the cross-lingual setting, when compared to the in-language results presented in Table 1, observing for the SVR method a drop of around 5 to 15%. While standard deviation was already poorly predicted in the in-language set-

<table>
<thead>
<tr>
<th>dataset</th>
<th>MEGA</th>
<th>BWK</th>
<th>BWK.3K</th>
<th>MRC</th>
</tr>
</thead>
<tbody>
<tr>
<td>lang</td>
<td>hr</td>
<td>en</td>
<td>en</td>
<td>en</td>
</tr>
<tr>
<td>size</td>
<td>2,682</td>
<td>22,797</td>
<td>3,000</td>
<td>4,061</td>
</tr>
<tr>
<td>method</td>
<td>SVR</td>
<td>FFN</td>
<td>SVR</td>
<td>FFN</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>C.M</th>
<th>0.760</th>
<th>0.742</th>
<th>0.887</th>
<th>0.834</th>
<th>0.872</th>
<th>0.863</th>
</tr>
</thead>
<tbody>
<tr>
<td>C.STD</td>
<td>0.265</td>
<td>0.274</td>
<td>0.484</td>
<td>0.461</td>
<td>0.376</td>
<td>0.364</td>
</tr>
<tr>
<td>I.M</td>
<td>0.645</td>
<td>0.602</td>
<td>-</td>
<td>-</td>
<td>0.803</td>
<td>0.787</td>
</tr>
<tr>
<td>I.STD</td>
<td>0.439</td>
<td>0.415</td>
<td>-</td>
<td>-</td>
<td>-</td>
<td>-</td>
</tr>
</tbody>
</table>

Table 1: Results of the in-language experiments on predicting mean (.M) and standard deviation (.STD) of concreteness (C) and imageability (I), either using a support vector regressor (SVR) or feed-forward network (FFN). Evaluation metric is the Spearman correlation coefficient.
Table 2: Results of the cross-lingual experiments, either using supervised learning (SVR, FFN), or simple dictionary lookup (DIC). Evaluation metric is the Spearman correlation coefficient. Results in bold are best results per problem with no statistically significant difference.

<table>
<thead>
<tr>
<th></th>
<th>source</th>
<th>target</th>
<th>SVR</th>
<th>FFN</th>
<th>DIC</th>
<th>SVR</th>
<th>FFN</th>
<th>DIC</th>
<th>SVR</th>
<th>FFN</th>
<th>DIC</th>
<th>SVR</th>
<th>FFN</th>
<th>DIC</th>
</tr>
</thead>
<tbody>
<tr>
<td>C.M</td>
<td>MEGA (hr)</td>
<td>MEGA (hr)</td>
<td>0.791</td>
<td>0.793</td>
<td>0.728</td>
<td>0.724</td>
<td>0.719</td>
<td>0.641</td>
<td>0.797</td>
<td>0.794</td>
<td>0.611</td>
<td>0.694</td>
<td>0.683</td>
<td>0.523</td>
</tr>
<tr>
<td>C.STD</td>
<td>BWK (en)</td>
<td>MEGA (hr)</td>
<td>0.178</td>
<td>0.141</td>
<td>0.224</td>
<td>0.185</td>
<td>0.145</td>
<td>0.137</td>
<td>-</td>
<td>-</td>
<td>-</td>
<td>0.648</td>
<td>0.531</td>
<td>0.503</td>
</tr>
<tr>
<td>I.M</td>
<td>MEGAB (en)</td>
<td>MRC (en)</td>
<td>-</td>
<td>-</td>
<td>-</td>
<td>-</td>
<td>-</td>
<td>-</td>
<td>-</td>
<td>-</td>
<td>-</td>
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<td>-</td>
</tr>
</tbody>
</table>

5 Conclusion

In this paper we have shown that concreteness and imageability ratings can be successfully transferred both to non-covered portions of the lexicon and to other languages via (cross-lingual) word embeddings.

With the in-language experiments we have shown that the arithmetic mean of both notions is much easier to predict than their standard deviation, the latter probably encoding word ambiguity, type of information not directly present in word embeddings.

Our experiments across languages have shown that the loss in comparison to in-language experiments on predicting the means of both concreteness and imageability are around 15%, a reasonable price to pay given the applicability of the method to all of the 77 languages present in the word embedding collection. The predictions of concreteness and imageability obtained in the 77 languages are available at http://hdl.handle.net/11356/1187.

Comparing the two methods of transfer – dictionary vs. cross-lingual embeddings, shows regularly better (5%–15%) results of the latter, proving once more the usefulness of word embeddings, especially in the currently expanding cross-lingual setup.

Acknowledgements

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*Ongoing developments are stored at https://github.com/clarinsi/megahr-crossling.*
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