

Thai Spelling Recognition Using a Continuous Speech Corpus

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Abstract

Spelling recognition is an approach to enhance a speech recognizer's ability to cope with incorrectly recognized words and out-of-vocabulary words. This paper presents a general framework for Thai speech recognition enhanced with spelling recognition. In order to implement Thai spelling recognition, Thai alphabets and their spelling methods are analyzed. Based on hidden Markov models, we propose a method to construct a Thai spelling recognition system by using an existing continuous speech corpus. To compensate the difference between spelling utterances and continuous speech utterances, the adjustment of utterance speed is taken into account. Assigning different numbers of states for syllables with different durations is helpful to improve the recognition accuracy. Our system achieves up to 79.38% accuracy.

1 Introduction

Nowadays, several works on automatic speech recognition (ASR) for continuous speech are being developed, not only system that rely on dictionary, but also the recognition on out of vocabulary circumstances. In a situation of misrecognition and out-of-vocabulary words, a practical and efficient solution that would assist the ASR is to equip the system with a spelling recognition subsystem, where users can spell out a word letter by letter. Spelling recognition is a challenging task with a high interest for directory assistance services, or other applications where a large number of proper names or addresses are handled. Many works that focus on spelling recognition were widely developed in several languages, for instance, English, Spanish, Portuguese and German. In (San-Segundo et al., 2001) the hypothesis-verification Spanish continuous spelled proper name recognition over the telephone was proposed. In this work, several feature sets were investigated by using models of neural networks. In their succeeding work (San-Segundo et al., 2002), three different recognition architectures, including the two-level architecture, the integrated architecture and the hypothesis-verification architecture, are analyzed and compared. In (Rodrigues et al.,

1997), a Portuguese speaker -independent system for recognizing an isolated letter was introduced. The system dealt with speech utterances over a telephone line using Hidden Markov Model (HMM). A number of experiments were made over four different perplexity language models. Mitchell and Setlur (1999) proposed a fast list matcher to select a name from a name list that was created from an n -best letter recognizer on spelling over a telephone line recognition task. In (Bauer and Junkawitsch, 1999), an approach is proposed to combine word recognition with spelling recognition in a user-friendly manner as a fall back strategy. As a German city name recognizer, the system was applied to directory assistance services.

Unlike other languages, spelling in Thai has several styles. One of them is similar to spelling in English, i.e., /d-ii//z-oo//g-ii/ for "dog". There are three more methods in Thai spelling, where some syllables are inserted to make it clearer for the hearer. One is to spell out a letter followed by its representative word's utterance. Another way is to mix the former two types. The third method is to spell out a set of letters that form a syllable, followed by its corresponding utterance. So far spelling recognition for Thai language has not been explored yet. One of the main reasons is that there is no standard corpus for this purpose. Creating a corpus of spelled utterances is a time consuming task. In this work we use the NECTEC-ATR Thai Speech Corpus, a standard continuous Thai speech corpus, for our spelling recognition system. Another objective of this work is to examine how a spelling system can be implemented using a normal Thai continuous speech corpus. That is, as the preliminary stage, we investigate the effects of spelling using such existing corpus.

This paper is organized as follows. In section 2, language characteristics in Thai are introduced. Section 3 presents our recognition framework. The spelling styles for Thai words are discussed in section 4. The experimental results and analysis are shown in section 5. Finally, the conclusion and future works are given in section 6.

5 Experimental Results and Analysis

5.1 Experimental Environment

As mentioned above, the corpus for a spelling recognition task is unfortunately not available at this time. Therefore, this work applies the NECTEC-ATR Thai Speech Corpus, constructed by NECTEC (National Electronics and Computer Technology Center) incorporated with ATR Spoken Language Translation Laboratories. In Thai language speech recognition, this corpus is normally used for a continuous speech recognition task. This speech corpus is used as the training set for our spelling recognition system. The corpus contains 390 sentences gathered by assigning 42 speakers (21 males and 21 females) to read all sentences for a trail. So, there are totally 16,380 read utterances.

In the first place, by the reason of computation time, only utterances of 5 males and 5 females, are used, i.e., totally 3,900 trained utterances. In addition, as our preliminary work, the effects of spelling result with a normal continuous training corpus are investigated. Even though, the training corpus has quite different characteristics compared to the test utterances, we can expect a reasonable result. The test utterance is constructed by recording the spelling of 136 proper names by a female participant.

The speech signals were digitized by 16-bit A/D converter of 16 kHz. A feature vector used in our experiment is a 39-feature vector, consists of 12 PLP coefficients and the 0th coefficient, as well as their first and second order derivatives. Therefore, there are totally 39 elements.

The language model used in this task is a bigram language model, trained from totally 6,107 proper names, i.e., 5,971 Thai province, district and subdistrict names, as well as 136 proper names from the test transcription.

A phone-based HMM is applied as the recognition system. The acoustic units used in this experiment are defined in the same manner as in (Pisarn and Theeramunkong, 2003). All experiments, including automatic transcription labelling, are performed using HTK toolkit (Young et al., 2002). The word correctness is given by the percentage of numbers of correct words divided by total number of words and the accuracy is computed by the percentage of subtracted the numbers of correct words by the number of insertion errors, which are then divided by total number of words.

5.2 Setting a Baseline

In the first experiment, we investigate the spelling results using the original training and testing data

as they are. This will be a baseline for all of our experiment. In this initial stage, the context-independent method (CI), achieves 79.94 and 57.99 for correctness and accuracy, respectively. The system with context-dependent method (CD) gains 70.80 and 46.09 for correctness and accuracy respectively. In principle, low accuracy is triggered by a large number of insertion errors. Because of this figure, two possible assumptions can be made (1) there is in compatible duration between the training and the test set, and (2) Our HMM models are inappropriate.

5.3 Adjusting the Duration

To investigate the results of the first assumption, the utterance speed of the utterances from the training and testing are measured in the form of the number of phone per second. The speed can be computed by dividing the number of total phones in each utterance transcription by its utterances duration in seconds. As a result, the average utterance speed of the training set is 11.7 phones/sec while the average utterance speed of the test set is only 4.6 phones/sec. This indicates that the speed of test utterances are approximately 2.5 times slower than that of train utterances. This difference may cause low accuracy.

To compensate for this duration difference among the training utterance and the testing utterance, a method to shrink and stretch a speech signal, by preserving pitch and auditory features of the original signal, is applied in our signal preprocessing. The experiments are done in two environments; stretching the training utterances and shrinking the test utterances. By adjusting the duration of the training and testing utterances, insertion errors could be reduced. Stretching the training utterances and shrinking the test utterances are performed using various scale factors in order to investigate the effectiveness. Table 4 shows the recognition results of stretched training utterances with various scale factors. Here, the original test utterances are used.

Duration	Model	%Correct	Accuracy
1.25Train	CI	81.91	62.49
	CD	82.05	66.36
1.43Train	CI	85.43	68.54
	CD	85.86	70.09
1.67Train	CI	86.42	63.34
	CD	84.59	63.97

Table 4. Recognition Results of Stretched Training Utterances with Various Scale Factors.

In principle, stretching training utterances causes the original utterances to be distorted. The more

scale the utterances are stretched, the more distorted the utterances we obtain. As stated in the previous section, utterances training are approximately 2.5 times faster than the test utterances. However, they are expected to achieve a very low accuracy. The experimental results show that by adjusting training utterances 1.43 times slower than the original one (1.43Train) can improve the correctness to 85.86 % and the accuracy to 70.09% in a context-dependent method. But with more stretching, the accuracy drops to 63.97%.

Reversely we also examine the system accuracy when the test utterances are shrunk on various scale factors. The original training utterances are used for training our system. The recognition results are shown in Table 5.

Duration	Model	%Correct	Accuracy
0.71Test	CI	86.28	74.88
	CD	82.41	73.12
0.43Test	CI	82.97	77.34
	CD	80.93	75.93

Table 5. Recognition Results of Shrunked Test Utterances with Various Scale Factors.

Shrinking test utterances can improve accuracy. Especially, the test utterances with 0.43 scaling factor can reduce the accuracy error to 19.35%, from 57.99% to 77.34%.

No.of states	Model	%Correct	Accuracy
3	CI	82.97	77.34
	CD	80.93	75.93
4	CI	80.01	76.78
	CD	79.73	76.85
5	CI	80.79	79.38
	CD	79.31	78.25
6	CI	78.04	76.99
	CD	76.92	76.21

Table 6. Recognition Accuracy with Various Numbers of States for a Long Vowel Phoneme.

5.4 Acoustic Models with Different Numbers of States

In fact, phone durations of each phoneme in Thai language do not have the same duration. Especially in the vowel class, there are vowels pairs, where one has a longer phone while the other has a shorter phone. For example, the vowel pair, *a* and *aa*, have a similar phone but different durations. The phoneme *a* has a shorter duration than the phoneme *aa*. The other vowel pairs are *i-ii*, *v-vv*, *u-uu*, *e-ee*, *x-xx*, *o-oo*, *@-@@*, *q-qq*, *ia-iaa*, *va-vva*, and *ua-uaa*. The shorter phone should not have the

same number of state as the longer one. Therefore, we examined the recognition rate on different numbers of HMM states. The experiment is examined using the 0.43Test set since it is the best one in the previous experiment. The results are shown in Table 6. In this experiment, the number of states for a long vowel phoneme is varied from 3 to 6 states. However, the numbers of states for the other phonemes are set to 3 states.

Table 6 shows that a 5-state HMM for a long vowel phoneme and a 3-state HMM for the other phonemes achieve the highest recognition accuracy, i.e., 79.38. This is, 2.04% error rate reduction compared with the 3-state HMM.

6 Conclusion

In this paper, we present a general framework for Thai speech recognition enhanced with spelling recognition. Four styles for spelling Thai words were discussed. To recognize spelling utterances, HMMs were constructed using a continuous speech corpus. To achieve higher correctness and accuracy, we compensated the utterance speed among the training and test utterances by stretching the training utterances or shrinking the test utterances. The experimental results indicated promising performance of 79.38% recognition accuracy after this adjustment. With a good scaling factor, the system achieved 19.35% improvement compared with the baseline where the training and test utterances were used as they are. Assigning a larger number of states to a longer syllable (i.e., long vowel) could improve recognition accuracy by 2.04 %. Our further works include (1) to construct a system that deals with several kinds of spelling methods, and (2) to explore the incorporation of spelling recognition into the conventional speech recognition system.

Acknowledgements

The authors would like to thank National Electronics and Computer Technology Center (NECTEC) for allowing us to use the NECTEC-ATR Thai Speech Corpus. We also would like to thank Dr. Virach Sornlertlamvanich for his useful suggestions through this work. This work has partly been supported by NECTEC under project number NT-B-22-I5-38-47-04.

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