

Target Language-Independent Acoustic Modeling for Medieval Latin

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Abstract. We present a medieval Latin charter speech recognition system which can be of great help for preserving Latin language documents from the same era, as optical character recognition systems are often challenged by historic documents. Our goal is to develop a medieval Latin recognizer that is built from multiple source languages but the target language. Our source languages for the acoustic model are the Visegrad languages (based on a geographical grouping): Czech, Hungarian and Polish. The baseline systems we start with are separately trained grapheme-based acoustic models for all the above three languages. We introduce two pronunciation modeling techniques to outperform the separately trained models. The first one is using grapheme-to-phoneme (G2P) mapping with Latin-specific pronunciation rules implemented. The second one is training a Unified Simplified Grapheme (USG) acoustic model that can deal with cross-native-language variations. We show that our methods outperform our baseline system, reducing the WER by ...% and ...% respectively. We also observe, that adding more languages to our joint acoustic model further reduces the WER.

Keywords: G2P, medieval Latin, low resource speech recognition, unified simplified grapheme modeling

1 Introduction

The pronunciation of Latin texts mainly depends on the era and region of their origin [2]. Apart from the two widely studied classical and ecclesiastical pronunciation styles [1], regional pronunciations exist emerging after the post-classical era. One of these pronunciation groups is the Slavic-Uralic [2] one, which uses the same pronunciation rules, described in detail in Section 3.2. Although the target pronunciation is considered to be uniform for this group, it also has to be taken into account, that the acoustic base of the different source languages varies, which can lead to different speakers pronouncing the same words differently. It also has to be noted, that apart from the variations in the pronunciations, orthographic and linguistic variations of Latin are also exhibited through regions.

This raises the question of how to create a speech recognition system which has to deal with uniform pronunciations for native speakers of different languages reading linguistically different texts. We propose a system that is suitable to recognize medieval

Latin speech spoken by speakers from the Visegrad region.¹ The system we develop is a joint system that can deal with both the variability in the speakers’ pronunciations when speaking medieval Latin, as well as the grammatical/lexical variabilities of the target texts. Therefore, it is important to collect in-domain textual/language data for the language model from the relevant geographican regions and time. We describe the data acquisition process in section 2.1.

Our baseline system consists of separately trained grapheme-based acoustic models for the three Visegrad languages (Czech, Hungarian and Polish). These separately trained models work good with their respective native speakers, but perform poorly with speakers of different native languages. We apply two different pronunciation modeling techniques to develop models that are superior to the baseline. The first one, dicussed in detail in Section 3.2, is based on the assumption that The second method we use is USGM (Unified Simplified Grapheme Modeling), where a joint/minimal/common grapheme inventory is established for all the languages paricipating in the joint acoustic model training. We describe the USG method in Section 3.3. Evaluation of the baseline systems and both above approaches is presented in section 4.

1.1 Related work

Similar work has been done for multi-dialectal languages such as Arabic in [10] where jointly trained acoustic models were outperformed by methods that unify dialect specific-acoustic models using knowledge distillation and multitask learning.

To our knowledge, no previous work has been done on medieval Latin speech recognition, nor on classical Latin for that matter.

2 Data

2.1 Textual data

As part of our inquiry was to cover linguistic variability accross the Visegrad region, aquiring textual data posed a few challenges. First of all, textual data are scarce for medieval Latin, and texts originating from this geographical region are even more scarce. Additionally, most of the available sources mix local languages and Latin, with no meta-data to separate them. For the scope of this paper, we collected monolingual texts only.

Training data A smaller amount of in-domain data (medieval charters) were collected from [8] (Monasterium), with an overall of 480k tokens. These documents are originating from the Hungarian Kingdom, from 1000 to 1524 AD. To increase the vocabulary size of the language model, we collected a relatively larger (but still small, compared to state-of-the-art language models used in speech recognition) 1.3 token corpus from [9] (LatinLibrary). This corpus consists of literary and historical texts from the post-classical era. In spite of our efforts, at the time of writing this paper, we could not gather textual data from the age and area of the Kingdoms of Bohemia and Poland.

¹ Sadly, but for this paper, although it belongs to the Visegrad region, the Slovakian language is not included in the experiments, due to the lack of speech data for acoustic model training.

Test data Using independent sources three-three charters were selected from the Kingdoms of Bohemia (CZ), Hungary (HU) and Poland (PL), from around 1200-1300 AD, for development and test data. The dev set was used for evaluating the language model, and to test the performance of our recognizers. Both dev and test sets were read out loud by historians fluent in medieval Latin.

Alternate spellings One interesting feature of the acquired corpora is that they contain a significant number of spelling variants. Having spelling variants in the corpus with identical pronunciation introduces noise, and thus has a negative effect on recognition results. To detect the spelling variants we took all pairs in the pronunciation dictionary whose pronunciation were identical, and used context and expert knowledge to decide whether the pair of equivalent pronunciations are spelling variants or homophones. We obtained a unified spelling for these variants by favouring the more frequent variant in the corpus (e.g. *maiestati* to *majestati*). Resolving spelling variants resulted in a more consistent corpus in terms of perplexity (reducing it from 775 to 672), and reduced the OOV rate by 0.8%.

Language model The language models we built from the two corpora were estimated with the SRI Language Modeling toolkit (SRILM) [5] using modified Kneser-Ney smoothing method. After estimating the mixture parameter, linear interpolation was used.

The perplexity measures on the dev data showed, that the Monasterium corpus originating from the time and era of the Hungarian Kingdom was indeed best fitting with the Hungarian subset of the test data with a perplexity of 82, and 0.9% OOV rate. Adding the LatinLibrary corpus increased the perplexity significantly, but reduced the OOV rate by 7% on the overall test data, as well as the WER, so we decided to use the interpolated language model.

2.2 Speech data

Training data For Czech and Hungarian the Speecon databases [7] and [6] and broadcast news speech data was used. For Polish, only broadcast news data [?] was available, comprising 31 hours of manually transcribed speech.

Test data Native speakers of Czech, Hungarian and Polish all of whom have experience with medieval Latin were asked to record the three dev and test sets described in Section 2.1. The recording conditions were accurately controlled: closetalking microphones, quiet, non reverberant acoustic environment, fluent, flawless speech, and at least 16 kHz, 16 bit (linear PCM) encoding. No instructions were given regarding the pronunciation, the speakers were using their expertise on medieval Latin pronunciation rules combined with their native language pronunciation. The overall length of the recorded test speech was around 30 minutes.

3 Acoustic modeling

Building an acoustic model for speech recognition requires long hours of transcribed speech. As of today (medieval) Latin is not spoken natively, and as to our knowledge, there is no recorded speech database. One obvious way to handle this problem is build/create a medieval Latin database; a proposition that requires lot of time, resources and trained speakers of medieval Latin. Another way of circumvent the lack of available speech data is to use speech data of spoken languages, preferably those ones whose native speakers are going to use the system.

For all the different pronunciation modeling methods, the acoustic models were trained as follows. Mel-Frequency Cepstrum + Energy features were used with Linear Discriminant Analysis (LDA) + Maximum Likelihood Linear Transformation (MLLT), with a splice context of ± 4 frames, 10 ms of frame shift. 9×40 dimensional spliced up feature vectors served as input to the feedforward, 6 hiddenlayer neural network with pnorm [16] activation function. Prior to DNN training, a Gauss Mixture Model (GMM) pre-training was performed. Clustering and Regression Tree (CART) [4] was applied to obtain acrossword context dependent shared state phone (or graph) models and their time alignment. The number of senones (and so the size of the DNN softmax output layer) was between 7.000 and 11.000 depending on the nature of the training data. The size of the hidden layers was kept constantly on 2.000. A minibatch size of 512, an initial learning rate of 0.1, and final learning rate of 0.01 was applied in 20 epochs using the KALDI toolkit [4].

3.1 Grapheme-based pronunciation modeling

For our three separately trained baseline systems grapheme-based pronunciation models were used. It was based on the same principles described in detail in Section 3.3, namely mapping the graphemes not present in the Latin grapheme set to their normalized counterparts.

3.2 Grapheme to phoneme mapping (G2P)

The Czech and Hungarian phoneme-based acoustic models we used were trained with G2P mapping between orthographic transcriptions and native phonemes. The G2P rules were composed with the language model within a WFST framework [3]. But since we were using these models for recognizing medieval Latin speech spoken by native speakers of Czech, Hungarian and Polish, Latin-specific pronunciation rules also had to be implemented. These include a few context independent digraph mappings, and a few context dependent rewrite rules, summarized in Table 3.2 and Table 3.2 respectively, for both Czech and Hungarian.

The Latin alphabet we extracted from our corpora (see Section 2.1) consisted of 24 elements.

3.3 Unified Simplified Grapheme Modeling

The second method we propose for cross-native-language Latin spech recognition is Unified Simplified Grapheme (USG) pronunciation modeling technique, which comes

Table 1. Latin digraph context-insensitive rewrite rules.

Digraph	ae	oe	ph	qu
CZ	e	oe	f	kv
HU	e	ø	f	kv

Table 2. Latin context-sensitive rewrite rules. V: vowel, VP: palatal vowel, C: consonant, *: zero or any, ^: beginning of word, [[^] *stx*]: not s, t or x.

GR	c	c	ch	ch	gu	gu	ti	ti
PH	ts	k	h	k	gv	gu	tsi	ti
rule	cVP	VNP	VC*ch	^C*ch	guV	guC	[[^] <i>stx</i>]tiV	tiC

in play when joint acoustic models are being trained to support recognition across multiple languages.

Unified The joint acoustic model requires a unified grapheme inventory for the training, so that only those graphemes are in the model that are in the intersection of the different grapheme inventory sets of the training languages. Those letters that are not in this intersection are mapped to their normalized forms, e.g. it had a diacritic mark (acute, caron, etc.) on it, we mapped it back to its normalized form (ř to r, etc.).

Simplified Since the target language was medieval Latin, the remaining unified grapheme set also had to be simplified to the Latin grapheme set, e.g. ó to o. Further than that, those graphemes that are non-native to Latin, and can straightforwardly mapped to a native Latin grapheme(s), were also replaced. These are mappings from x to ks, y to i and w to v. As a result, a unified and simplified grapheme inventory set was produced, compatible with medieval Latin. The USG units were then used as acoustic model units in the training.

4 Experimental results

We conducted experiments on medieval Latin, spoken by native speakers of three languages (Czech, Hungarian and Polish), where the test are texts originating from different regions, as described in Sections 3 and 2.1. We introduced two techniques to improve the recognition results on separately trained grapheme-based models, described in 3.1. Our experimental results showed that both methods outperformed the baseline system.

4.1 Latin-specific G2P results

4.2 USG results

The Czech-Hungarian-Polish joint acoustic model

Table 3. Results for separately trained grapheme-based models of Czech (76 hours), Hungarian (112 hours) and Polish (31 hours).

AM Language	Speaker			
	CZ	HU	PL	Avr.
CZ	51	74	94.1	73
HU	32.2	28.1	77.1	45.8
PL	60.3	68.3	73.1	67.2

Table 5. Hungarian grapheme-based baseline model. Size of acoustic model: 112 hours.

Speaker	Latin Test Text			
	CZ	HU	PL	Avr.
CZ	31.8	19.1	45.6	32.2
HU	32.2	17.3	34.9	28.1
PL	79.6	74.5	77.1	77.1
Avr.	47.9	37	52.5	45.8

Table 7. Czech Latin-specific G2P model. Acoustic model size: 76 hours.

Speaker	Latin Test Text			
	CZ	HU	PL	Avr.
CZ	39.3	28.8	47.4	38.5
HU	52	40	58.7	50.2
PL	94.1	67.3	97.2	86.2
Avr.	61.8	45.4	67.8	58.3

Table 9. USG model of Czech (76 hours), Hungarian (112 hours) and Polish (31 hours).

Speaker	Latin Test Text			
	CZ	HU	PL	Avr.
CZ	24.8	17.9	36.1	26.3
HU	32.9	20.9	30.3	28
PL	57.2	38.2	78	57.8
Avr.	38.3	25.7	48.1	37.4

Table 11. USG model of Hungarian (112 hours), Polish (31 hours) and Roman (35 hours).

Speaker	Latin Test Text			
	CZ	HU	PL	Avr.
CZ	23.2	14.2	32.7	23.4
HU	23	14.6	27.5	21.7
PL	55.3	30	68.8	51.4
Avr.	33.8	19.6	43	32.2

Table 13. USG model of Czech (76 hours), Hungarian (112 hours), Polish (31 hours) and roman (35 hours).

Speaker	Latin Test Text			
	CZ	HU	PL	Avr.
CZ	18.4	12.1	28.7	19.7
HU	20.4	14.6	25.7	20.2
PL	54.6	25.4	64.2	48.1
Avr.	31.1	17.4	39.5	29.3

Table 4. Czech grapheme-based baseline model. Size of acoustic model: 76 hours.

Speaker	Latin Test Text			
	CZ	HU	PL	Avr.
CZ	50.2	33.6	69.1	51
HU	70.4	62.7	89	74
PL	82.2	96.4	103.7	94.1
Avr.	67.6	64.2	87.3	73

Table 6. Polish grapheme-based baseline model. Size of acoustic model: 31 hours.

Speaker	Latin Test Text			
	CZ	HU	PL	Avr.
CZ	62.5	46.7	71.9	60.4
HU	72.4	57.3	75.2	68.3
PL	75	63.6	80.7	73.1
Avr.	70	55.9	75.9	67.3

Table 8. Hungarian Latin-specific G2P model. Acoustic model size: 567 hours.

Speaker	Latin Test Text			
	CZ	HU	PL	Avr.
CZ	19.7	7.3	26.3	17.8
HU	25	25.4	20.2	23.5
PL	47.4	24.6	60.5	44.2
Avr.	30.7	19.1	35.7	28.5

Table 10. USG model of Czech (76 hours), Hungarian (112 hours) and Roman (35 hours).

Speaker	Latin Test Text			
	CZ	HU	PL	Avr.
CZ	21.5	11.2	33	21.9
HU	24.3	14.6	24.8	21.2
PL	57.9	37.3	69.7	55
Avr.	34.6	21	42.5	32.7

Table 12. USG model of Czech (76 hours), Polish (31 hours) and Roman (35 hours).

Speaker	Latin Test Text			
	CZ	HU	PL	Avr.
CZ	20.8	13	35.2	23
HU	36.8	23.6	39.5	33.3
PL	55.9	32.7	67.9	52.2
Avr.	37.8	23.1	47.5	36.2

Error analysis

4.3 Conclusions

In this paper, we presented two pronunciation modeling techniques for a target-language independent medieval Latin speech recognizer to eliminate the efforts of digitizing medieval Latin charter data. With the objective of outperforming the separately trained grapheme-based models, we presented two approaches: an expert G2P modeling, and USG modeling. The results showed...

Future research directions include acquiring a considerable amount of medieval speech and textual data.

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