

Unified Simplified Grapheme Acoustic Modeling for Medieval Latin LVCSR

Lili Szabó, Péter Mihajlik, András Balog, Tibor Fegyó

lili@speechtex.com



Motivation

- Digitizing medieval charters when optical character recognition is not sufficient

Challenges

- Latin is not spoken natively
- There is no available speech database, and it is resource-heavy to create one
- Many variants/dialects exist, and we can only make guesses about the pronunciation
- The pronunciation mainly depends on
 - the **era** of the read text
 - the **geographical region** where the text originates from
 - the **native language** of the speaker

Text data

Regions of origin: Kingdom of Bohemia (CZ), Kingdom of Hungary (HU), Kingdom of Poland (PL)

- In-domain data (Monasterium): medieval charters (HU), 480k/35k token/type
- Background data (Latin Library): historical texts, 1.3M/115k token/type

Spelling variants

jam	iam
judex	iudex
gracia	gratia

Language model

- 3-gram language model
- Kneser-Ney smoothing
- Interpolating the two corpora
- SRILM [2]

Perplexity measures on test

Table 1: Perplexity/OOV rate (%)

Corpus	Text region			All
	CZ	HU	PL	
Monasterium	551/11.8	82/0.9	3130/18.3	479/10.5
Latin Library	3266/7.8	3549/1.6	2305/5.5	2992/9.7
Interpolated	924/3.9	82/0.9	2288/5.5	672/3.5

System diagram

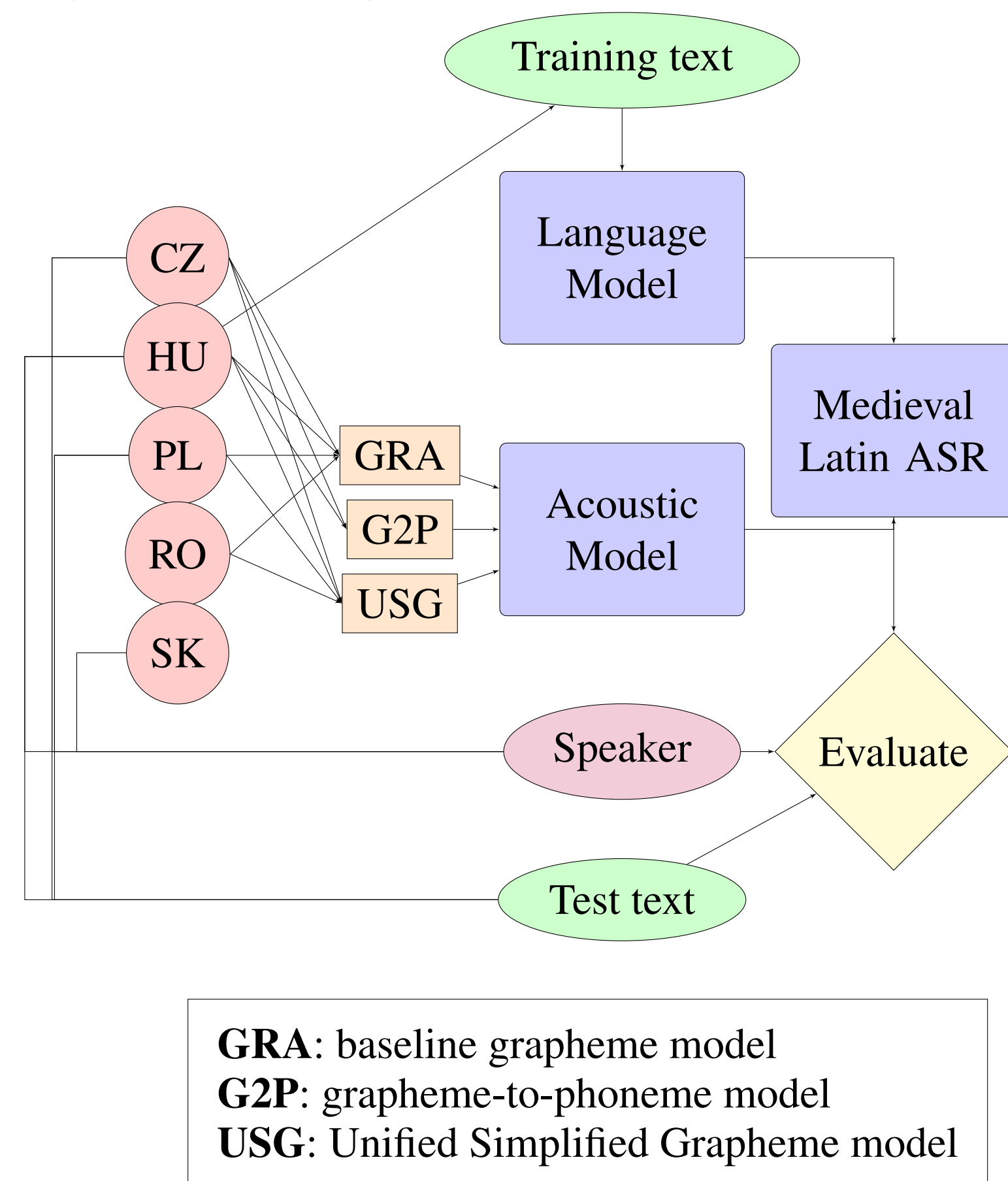


Figure 1: Medieval Latin Speech Recognizer

Speech data

- CZ: 76 hours
- HU: 567 hours (G2P) or 112 hours (grapheme and USG)
- PL: 31 hours
- RO: 35 hours

Test data

- Independent medieval charters
- Region of read text: CZ, HU, PL
- Native language of test speakers: CZ, HU, PL, SK

Acoustic model

- 6-hidden-layer DNN
- 2000 neurons per layer
- p-norm activation function
- 7000-11000 senones (softmax size)
- Kaldi toolkit [1]

Baseline Grapheme Model

- All graphemes are trained
- Only those grapheme models are retained that are part of the Latin alphabet, e.g.
 - keeping model of τ
 - throwing away model of ř

Table 2: Word Error Rate (WER[%]) results for monolingual grapheme-based acoustic models of Czech, Hungarian, Polish and Romanian (CZ, HU, PL, RO).

AM Language	Speaker				Σ
	CZ	HU	PL	SK	
CZ	53.6	73.8	62.9	45.7	59.0
HU	33.7	28.6	47.1	29.1	34.6
PL	65.0	67.6	46.4	51.1	57.5
RO	53.6	69.1	44.7	43.8	52.8

Knowledge-based grapheme-to-phoneme (G2P) mapping

Figure 2: Latin digraph context-insensitive rewrite rules and context-sensitive rewrite rules. V: vowel, VP: palatal vowel, ^VP: everything but a palatal vowel, C: consonant, *: zero or any, ^: beginning of word, [stx]: not s, t or x.

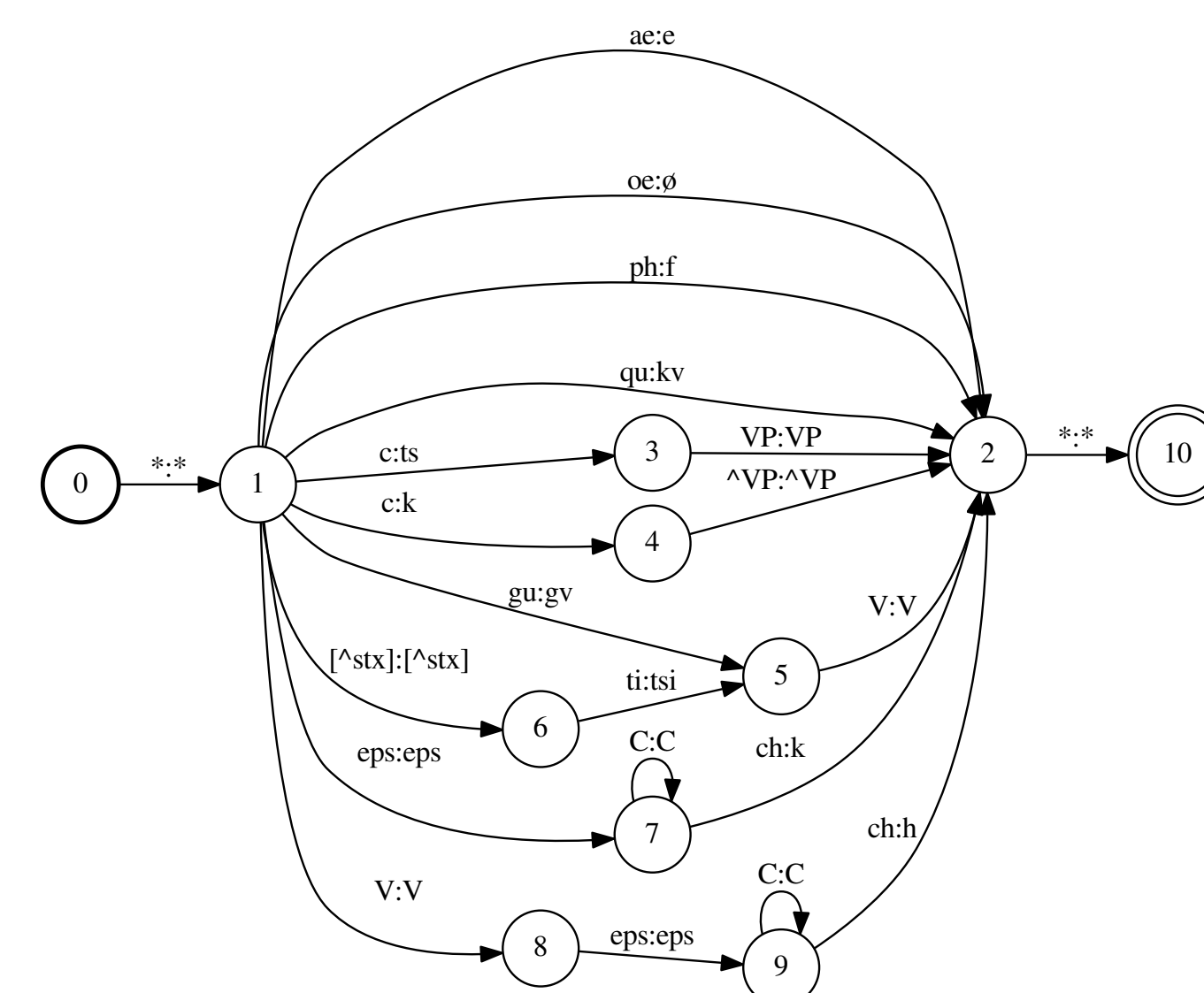


Table 3: WER[%] for Czech-Latin source-target G2P model. Acoustic model training set: 76 hours.

Speaker	Latin Test Text			
	CZ	HU	PL	Σ
CZ	43.8	28.2	49.1	40.4
HU	48.7	40.0	58.7	49.1
PL	53.3	18.2	53.2	41.6
SK	30.3	30.0	44.0	34.8
Σ	43.9	28.9	50.8	41.2

Table 4: WER[%] for Hungarian-Latin source-target G2P model. Acoustic model training set: 567 hours.

Speaker	Latin Test Text			
	CZ	HU	PL	Σ
CZ	19.4	6.4	28.0	17.9
HU	25.0	25.4	20.2	23.5
PL	28.9	15.4	41.3	28.5
SK	20.4	9.1	22.9	17.5
Σ	22.6	12.5	28.1	21.1

Unified Simplified Grapheme (USG) Model

- Utilizing many available language resources in the hopes that statistical variations help generalizing over different pronunciations

Table 5: Simplification examples for the unified model.

Language	CZ	HU	PL	RO
Orthographic form	řekl	őz	miś	apă
USG transcription	rekl	oz	mis	apa

Table 6: WER[%] for all the three-language USG models.

AM Language	Speaker				Σ
	CZ	HU	PL	SK	
CZ+HU+PL	28.2	28.2	27.7	22.4	26.6
CZ+HU+RO	23.3	21.4	23.9	19.2	21.9
CZ+PL+RO	24.6	33.1	25.6	19.8	25.8
HU+PL+RO	24.8	21.5	25.7	20.7	23.2

Table 7: WER[%] for USG model of Czech, Hungarian, Polish and Romanian (CZ+HU+PL+RO).

Speaker	Latin Test Text			
	CZ	HU	PL	Σ
CZ	20.4	11.8	30.7	21.0
HU	21.1	14.6	25.7	20.5
PL	23.0	10.0	33.0	22.0
SK	14.5	12.7	24.8	17.3
Σ	19.9	12.2	29.0	20.4

Conclusions

- Knowledge-based G2P modeling is good, but time consuming and restricted
- Four-language USG modeling is the best
 - It is able to generalize over different speaker test sets

References

- [1] Povey, D., Ghoshal, A., Boulianne, G., Burget, L., Glembek, O., Goel, N., Hannemann, M., Motlicek, P., Qian, Y., Schwarz, P., Silovsky, J., Stemmer, G., Vesely, K.: The Kaldi speech recognition toolkit. In: IEEE 2011 Workshop on Automatic Speech Recognition and Understanding. IEEE Signal Processing Society (2011)
- [2] Stolcke, A.: SRIlm – an extensible language modeling toolkit. In: In Proceedings of the 7th International Conference on Spoken Language Processing (ICSLP). pp. 901–904 (2002)