

Example-Based Automatic Phonetic Transcription

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Motivation

Why use automatic phonetic transcription?

- Phonetic transcriptions are an essential resource in speech technologies and linguistics.
 - Speech recognizers
 - Speech synthesis
 - Labelling of corpora
- Manual transcription is time-consuming, expensive and error-prone.



Motivaton (2)

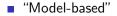
Benefits of automatic phonetic transcription

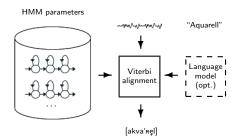
- Creation of draft transcriptions
 - Correction by human transcribers instead of creation from scratch
 - Faster and cheaper
- More objective than transcriptions of a team of human transcribers
- Consistency check of already transcribed material



Existing approaches

■ Mostly based on Hidden Markov Models (HMMs)



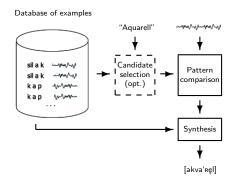




Our approach

 Inspired by concatenative speech synthesis and template-based speech recognition

"Example-based"





2 scenarios

Constrained phone recognition

Unconstrained phone recognition



2 scenarios

- Constrained phone recognition
 - Decision based on audio sample and intermediate transcription derived from orthographic transcription by letter-to-sound rules

Unconstrained phone recognition



2 scenarios

- Constrained phone recognition
 - Decision based on audio sample and intermediate transcription derived from orthographic transcription by letter-to-sound rules

$$+$$
 "Bäcker" /b e k 6/ \rightarrow [beke]

Unconstrained phone recognition



2 scenarios

- Constrained phone recognition
 - Decision based on audio sample and intermediate transcription derived from orthographic transcription by letter-to-sound rules

- Unconstrained phone recognition
 - Decision based on audio sample only





Database of examples

- Three-phone speech samples
- Phone boundaries determined by doing forced alignment with the Hidden Markov Toolkit (HTK)
- 12 Mel Frequency Cepstral Coefficients (MFCCs) plus overall energy, delta and acceleration coefficients: 39 parameters per frame

Pattern matching

- Measure for similarity between two utterances
- Dynamic time warping (DTW) algorithm
- Segmental and open-begin-end DTW



Transcription synthesis

- Constrained phone recognition
 - Number of phones fixed
 - Most frequent phones from best matching three-phone samples
- Unconstrained phone recognition
 - Number of phones unknown
 - List of n best matching samples for each frame
 - Nearest neighbor classification



Transcription synthesis

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 - Number of phones fixed
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"Bäcker" /b e k 6/

```
sil b e_o k 6 sil
```



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[beke]

Unconstrained phone recognition

sil b b b e_o e_o e_o e_o k k 6 6 6 sil

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Transcription synthesis

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"Bäcker" /b e k 6/
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[beke]

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sil b b b e_o e_o e_o e_o k k 6 6 6 sil

b e_o k 6

[beke]



Evaluation

Evaluation database: ADABA

- Austrian pronunciation database
- 6 professional speakers: Austrian, German and Swiss
- Narrow transcriptions: 89 phonemes instead of 45 in SAMPA German
- About 12 000 utterances per speaker (\sim 5h speech)
- Recordings in studio quality
- Provided by Rudolf Muhr, Research Center for Austrian German
 http://adaba.at/
 - http://adaba.at/



Evaluation (2)

Data set specification

- Restriction to a single speaker
- 85% training data, 5% development data, and 10% test data

Evaluation measures

Percentage of correct phones and phone accuracy

$$PC = \frac{N - D - S}{N} \times 100\%$$
 $PA = \frac{N - D - S - I}{N} \times 100\%$

N ... total number of phones in the reference transcriptionD ... number of deletions, S ... number of substitutionsI ... number of insertions.



Evaluation (3)

Benchmark: Comparison to a model-based transcriber

- Trained with Hidden Markov Toolkit (HTK)
- Same data and acoustic frontend
- 5-state left-to-right context-dependent triphone models with up to 16 GMMs
- For constrained phone recognition:
 Use of intermediate transcription for language model



Results

Constrained phone recognition

| | Int. Tr. | Model-based | Example-based |
|----|----------|-------------|---------------|
| PC | 83.36% | 90.88% | 91.95% |
| PA | 81.22% | 88.83% | 89.89% |

Performance differences are significant at the 0.1% level using the Matched-Pairs test.



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Unconstrained phone recognition

| | Model-based | Example-based |
|----|-------------|---------------|
| PC | 88.10% | 85.21% |
| PA | 86.96% | 82.38% |

Performance differences are significant at the 0.1% level using McNemar's test.



Implementations

EXTRA

- Standalone Java application
 - Evaluation and analysis of transcriptions
 - Batch transcription mode

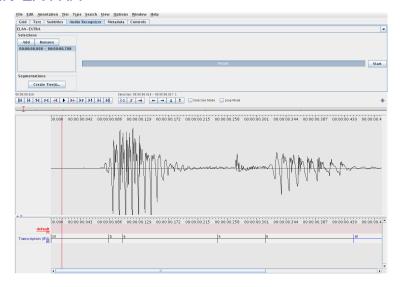
ELAN-EXTRA

Extension for the ELAN linguistic annotation software

http://www.spsc.tugraz.at/people/stefan-petrik/project-extra

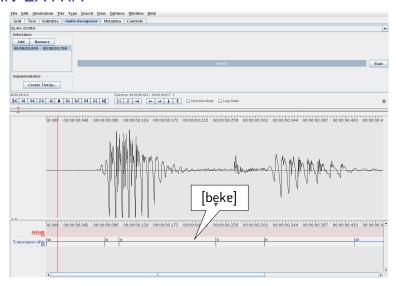


ELAN-EXTRA



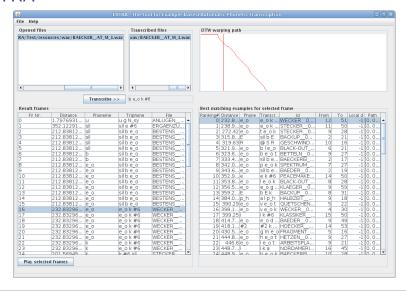


ELAN-EXTRA





EXTRA





Conclusion

- Example-based approach to automatic phonetic transcription
 - Comparison to concrete audio samples instead of model
 - Detection of rare pronunciation variants possible
- Useful support for transcription of speech corpora
 - Manual transcription of part of corpus rest automatically
 - Consistency check easily feasible
- Evaluation on the ADABA database
 - Comparable to an HMM-based transcription system
 - Best results with a combination of rule-based and example-based APT



Discussion

Thank you for your attention!



References I



C. Cucchiarini and H. Strik, "Automatic phonetic transcription: An overview," *Proceedings of ICPhS*, pp. 347–350, 2003.



M. De Wachter, M. Matton, K. Demuynck, P. Wambacq, R. Cools, and D. Van Compernolle, "Template-based continuous speech recognition," *IEEE Transactions on Audio, Speech, and Language Processing*, pp. 1377–1390, 2007.



C. Leitner, "Data-based automatic phonetic transcription," Master's thesis, Graz University of Technology, 2008.



R. Muhr, "The Pronouncing Dictionary of Austrian German (AGPD) and the Austrian Phonetic Database (ADABA) – Report on a large phonetic resources database of the three major varieties of German," *Proceedings of LREC*, 2008.



L. Rabiner and B.-H. Juang, Fundamentals of Speech Recognition. Prentice Hall PTR, 1993.



A. Park and J. R. Glass, "Towards unsupervised pattern discovery in speech," *IEEE Workshop on Automatic Speech Recognition and Understanding, 2005*, pp. 53–58, 2005.



References II



P. Tormene, T. Giorgino, S. Quaglini, and M. Stefanelli, "Matching incomplete time series with dynamic time warping: an algorithm and an application to post-stroke rehabilitation," *Artificial Intelligence in Medicine*, vol. 45, no. 1, pp. 11–34, January 2009.



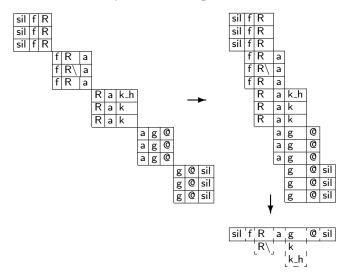
P. Wittenburg, H. Brugman, A. Russel, A. Klassmann, and H. Sloetjes, "ELAN: a professional framework for multimodality research," *In Proceedings of Language Resources and Evaluation Conference (LREC)*, 2006.



S. Young, G. Evermann, M. Gales, T. Hain, D. Kershaw, X. A. Liu, G. Moore, J. Odell, D. Ollason, D. Povey, V. Valtchev, and P. Woodland, *The HTK Book*. Cambridge University Engineering Department, 2006.



Synthesis - constrained phone recognition





Synthesis - unconstrained phone recognition

