

Unsupervised Language Acquisition Learning the Components of a Language

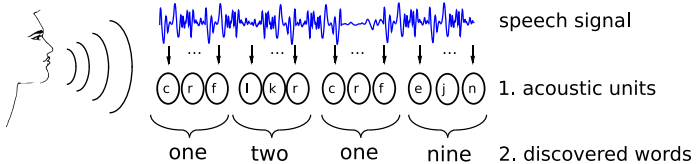
Dipl.-Ing. Oliver Walter

Department of Communications Engineering - University of Paderborn

June 24, 2014

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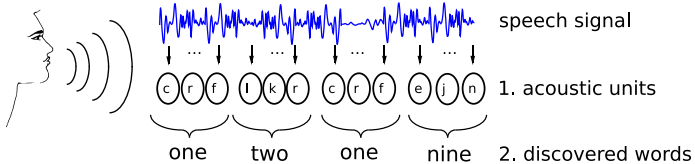
Unsupervised Language Acquisition



Unsupervised Learning

- Only speech features available: Zero resource setup
- No transcription of speech signal in terms of words and acoustic units available

Unsupervised Language Acquisition



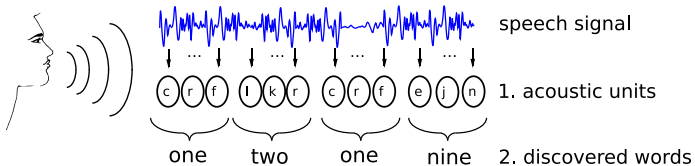
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Objective

- Unsupervised language acquisition
- "Learn a language like a child"
- Different approaches:
 - ▶ Exemplar based pattern discovery
 - ▶ Statistical model based pattern discovery
 - ▶ Flat and hierarchical approaches

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- Unsupervised language acquisition
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 - Different approaches:
 - ▶ Exemplar based pattern discovery
 - ▶ Statistical model based pattern discovery
 - ▶ Flat and hierarchical approaches
- ⇒ Use discovered word sequence for unsupervised speech recognizer training

Exemplar based Pattern Discovery

Goal: automatically find recurring acoustical patterns in audio recordings

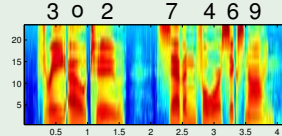
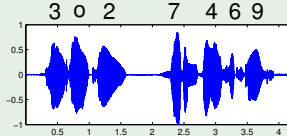
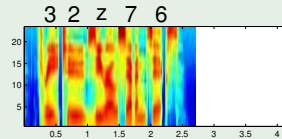
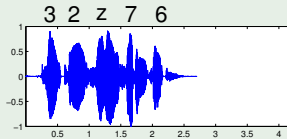
- Given: continuous audio stream
 - Exemplar based method: Find similarities by comparing sequences
- ⇒ Number and segmentation of audio patterns unknown

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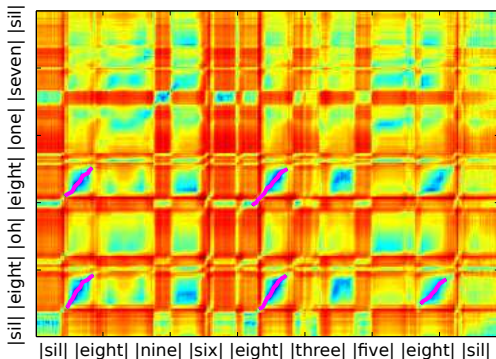
Example Sequences: Time and Spectral Domain Representation



Dynamic Time Warping

Dynamic Time Warping (DTW) based pattern search

- **Goal:** find similar exemplars in two sequences
 - Calculate distance between each pair of feature vectors of two sequences
 - Each region of low distance maps two similar exemplars
- ⇒ Find connected regions with low distance



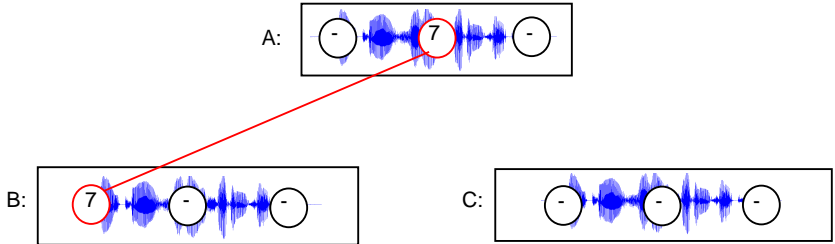
Clustering Algorithm

- **Goal:** Form clusters of similar exemplars in multiple sequences
- **Input:** Comparison of sequences A, B and C only delivers exemplar pairs

Exemplar Clustering

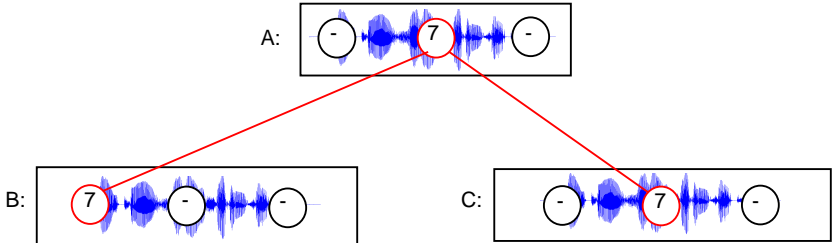
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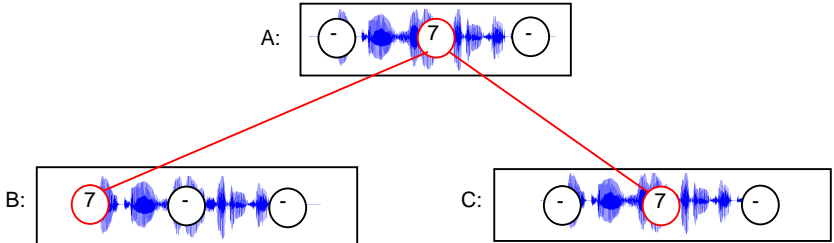
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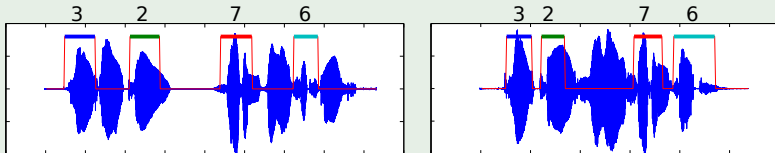
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 - ▶ The resulting graph is clustered using an unsupervised graph clustering algorithm



Example results

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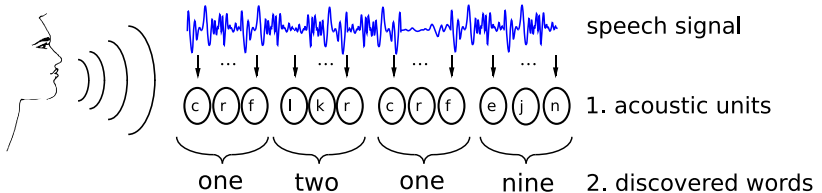
- Detection of single digits in Sequences of digits



Some conclusions

- Simple to implement
- Computationally expensive
- No statistical modeling

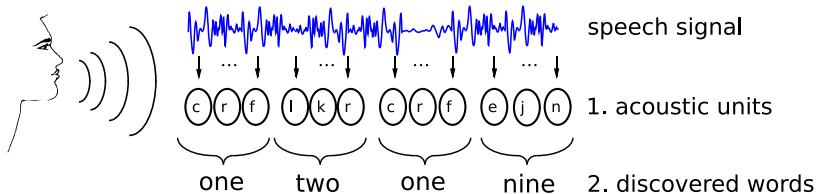
Hierarchical System for Unsupervised Word Discovery



Hierarchy

- Two-level hierarchical approach:
 - ▶ 1. Model speech signal as sequence of acoustic units
 - ▶ 2. Model recurring sequences of acoustic units as words

Hierarchical System for Unsupervised Word Discovery



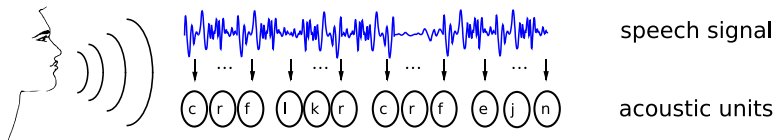
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Statistical Model based approach

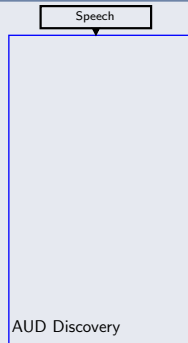
- Learning of different statistical models:
 - ▶ Acoustic model
 - ▶ Probabilistic pronunciation lexicon
 - ▶ Language model

Acoustic Unit Discovery (Overview)

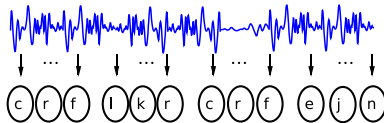


Three steps to acoustic unit discovery

- **Goal:** Learn acoustic units representing repeating sequences of speech features
- **Key Idea:** Speech signal consist of small number of building blocks, e.g. phones



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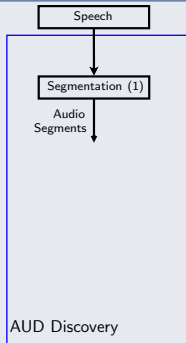


speech signal

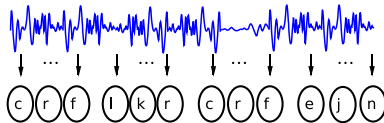
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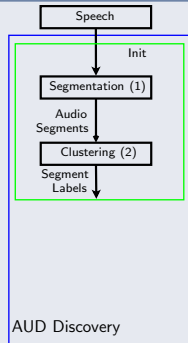


speech signal

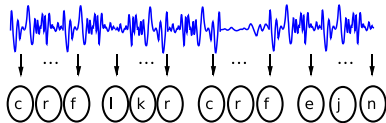
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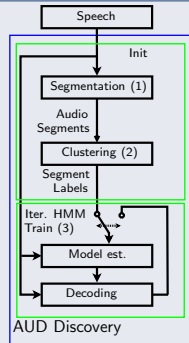


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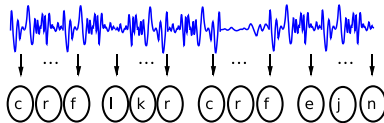
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 - ▶ 3. Iterative HMM training for each acoustic unit



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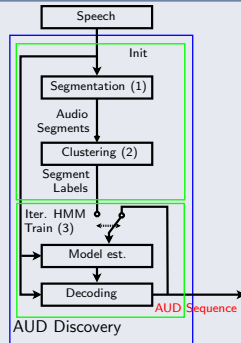


speech signal

acoustic units

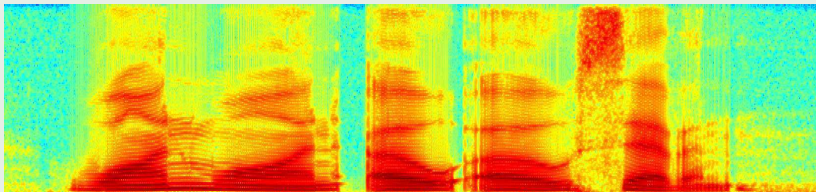
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 - ▶ 3. Iterative HMM training for each acoustic unit
- **Output:** transcription of speech signal in terms of a sequence of acoustic units



Step 1: Segmentation

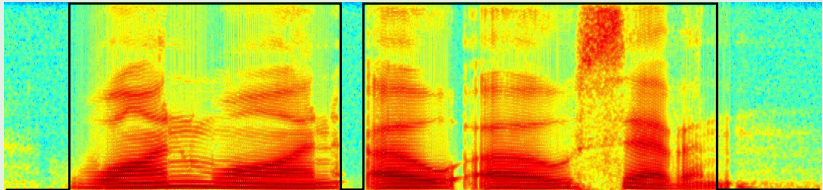
Spectrogram („one, one, oh, oh, seven“)



Step 1: Segmentation

- Use Voice Activity Detection (VAD) to support segmentation

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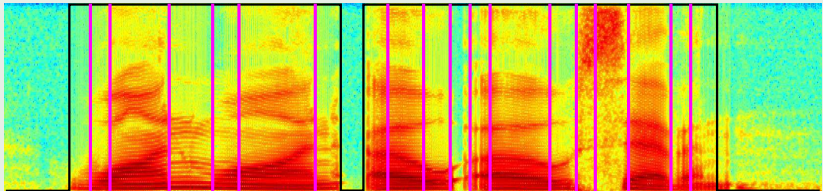


- VAD: black line, low: inactive, high: active

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Spectrogram („one, one, oh, oh, seven“)

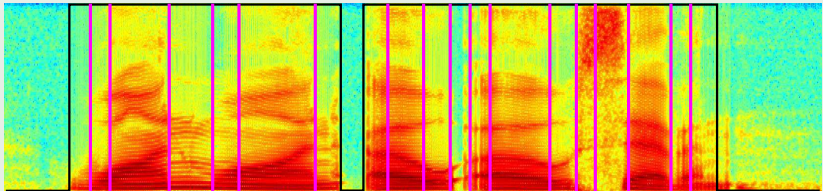


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- **Segmentation:** magenta line, indicating segment borders

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- ⇒ **Output:** Initial transcriptions in terms of segment numbers

Spectrogram („one, one, oh, oh, seven“)



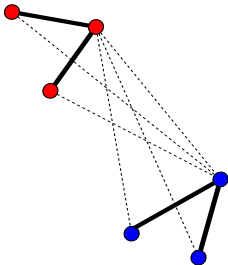
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Cluster on **sparse** distance matrix

- Build adjacency matrix according to DTW distances between segments
- Calculation of all distances too costly
- Calculate distances only between seeds and all segments
- Use kmeans++ like seed selection
- Use unsupervised graph clustering algorithm to cluster the graph

Iterative Hidden Markov Model (HMM) Training

Step 3: Acoustic unit model training and refinement

- Train HMM for each acoustic unit
- Left to right 3-state HMM
- Gaussian Mixture Model emission distributions
- Iterate between model estimation and decoding until convergence

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Training Algorithm

- Iterative HMM training using the resulting sequence of cluster labels from the clustering step as an initial transcription for the input signal:

$$\text{Model estimation: } \Lambda^{(\kappa+1)} = \operatorname{argmax}_{\Lambda} \prod_{d=1}^D p(\mathbf{X}_d | T_d^{(\kappa)}; \Lambda^{(\kappa)})$$

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(iteration index κ , HMM parameters Λ and transcriptions T)

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⇒ **Output:** Refined transcription in terms of acoustic units

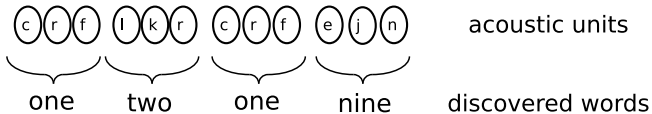
Example transcriptions: TiDigits - Digit Sequences

33031: sil F **CF BB** G sil C **CF BB** C sil AA B D I sil C **CF BB** D DC EG EJ BD I sil
 3533: sil C **CF BB** G AE AA DE FA AH sil C **CF BB** I G C **CF BB** I sil

Example transcriptions: Domotica 3 - Dysarthric Speech

- Two Repetitions of the sentence: ALADIN hoofdeinde op stand 1
- Repetition 1:
 - ▶ AJ AE AA AC B AF F BJ C H H AH AB AF AC AD BJ C AC F
 F AD E I AC H AH AB AF F
- Repetition 2:
 - ▶ AJ AE AA AC B AF F BJ C H AH AB AF AC AD E C H BB
 F AD E I AC H AH AB AF F

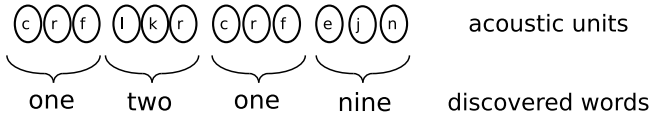
Word Discovery (Overview)



Unsupervised Word discovery

- **Input:** Acoustic unit sequence
- **Goal:** Learn word models representing repeating sequences of acoustic units
- **Key Idea:** Segmentation of input sequence

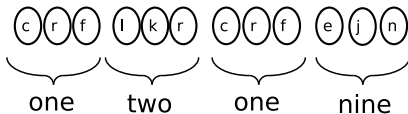
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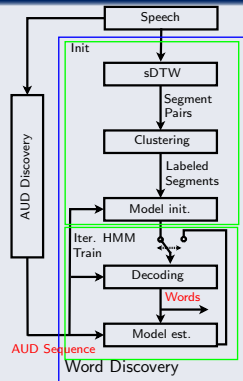
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 - ▶ **Words:** Probabilistic pronunciation lexicon
 - ▶ **Language Model:** Power Law distribution
 - ▶ **Semi-Supervised learning:** Initialization of pronunciation lexicon

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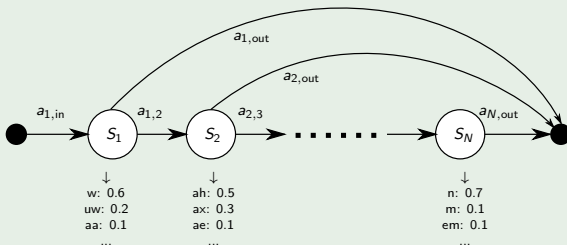
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- ⇒ **Output:** Sequence of words



Word Model: Probabilistic Pronunciation Lexicon

- One HMM with discrete emission distributions per word
- Length modeling: product of transition probabilities delivers probability for length

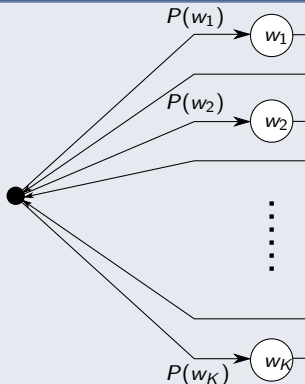
Example Sequences: One \rightarrow (w ah n|uw ax m|. . .)



- Parametric: How many HMMs?
- Parameter space grows with each HMM: $N \times N_{AUD}$ for emission distributions

Language Model connects Words to ergodic Markov chain

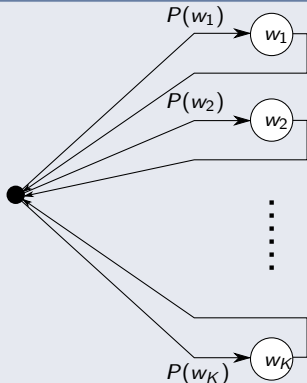
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- Word models connected by language model to form ergodic HMM
- Language model: power law distribution over words \rightarrow Zipf's Law

$$P(w_k; s) = \frac{k^{-s}}{\sum_{i=1}^K i^{-s}}$$

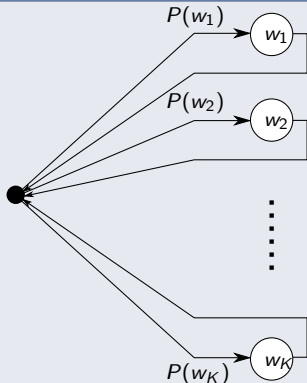


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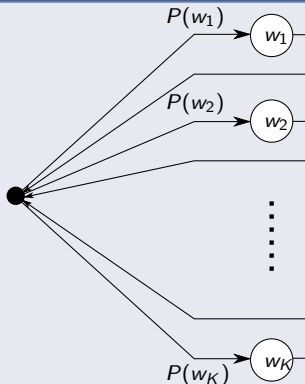


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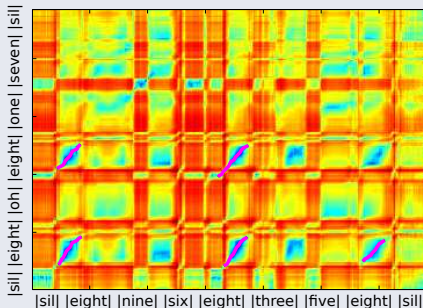
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Semi-Supervised Initialization of word models

- EM Algorithm is sensitive to local maxima and requires initialization
- **Initialization without knowledge:** Draw parameters randomly
- **Semi-Supervised initialization:**
 - ▶ DTW-based pattern discovery algorithm delivers clusters of patterns in the input signal
 - ⇒ Run DTW algorithm on subset of input data to find words (9% of data → 3.5% coverage)



- ▶ Labels can be assigned to discovered clusters by listening to exemplars
- ▶ For each discovered cluster initialize the emission distributions of a word HMM

Experimental Results

Experimental Results

- **Input:** Acoustic unit sequence learned from TiDigits
- **Performance measure:** Word Accuracy in %
- **Random initialization** of 11 word HMMs: 67.9%
- **DTW-based initialization:** 8 of the 11 word HMMs initialized: 81.9%
- **Unsupervised speech recognizer training:** iterative training of GMM-HMM speech recognizer using discovered word sequence as initial transcription:

Iteration	0	1	3	5	7
Random initialization	67.9	80.8	82.9	84.4	84.7
DTW-based initialization	81.9	96.6	98.4	98.5	98.5

⇒ The performance of semi-supervised training is close to the supervised training

Some Conclusions

- Delivers good results on small databases when number of words known
- Standard HMM training algorithms can be used for parameter estimation
- Parametric in terms of the number of words
- Parameter space (e.g. for pronunciation lexicon) grows with number of words

Conclusion, further Research and Outlook

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- Statistical model based pattern discovery
- Hierarchical structure of language
- Learning of Acoustic Units, Words and Language Models

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Further Research

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- Unsupervised Segmentation of error free text and noisy input (lattices)
- Joint learning of higher order phoneme/word language models and segmentation

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Outlook

- Model variation in pronunciation and errors/noise in segmentation algorithm
- Nonparametric acoustic model discovery
- Integration of acoustic model, word and language model discovery



Thank you for your attention!

Questions ?

Dipl.-Ing Oliver Walter

University of Paderborn
Department of Communications
Engineering

walter@nt.uni-paderborn.de
nt.uni-paderborn.de