

Discriminative Keyword Spotting with Limited Data

Joseph Keshet

**Department of Computer Science
Bar-Ilan University**

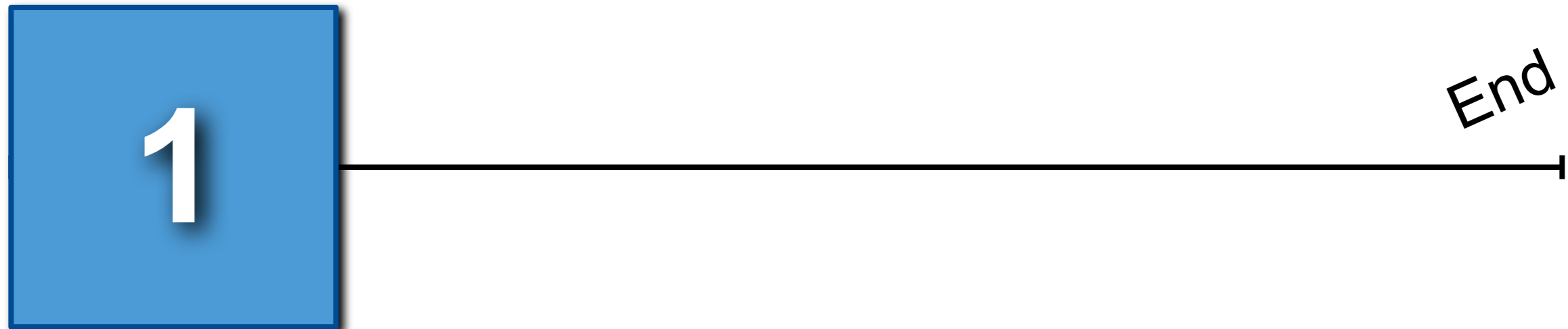
Outline

Start

End



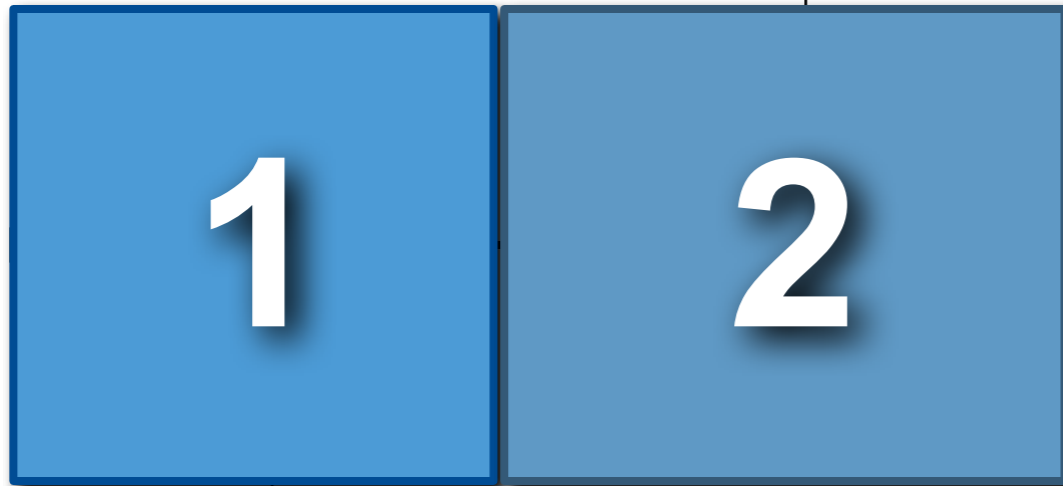
Outline



Keyword spotting
dominant paradigm and
its shortcomings

Outline

Articulatory feature-
based pronunciation
modeling

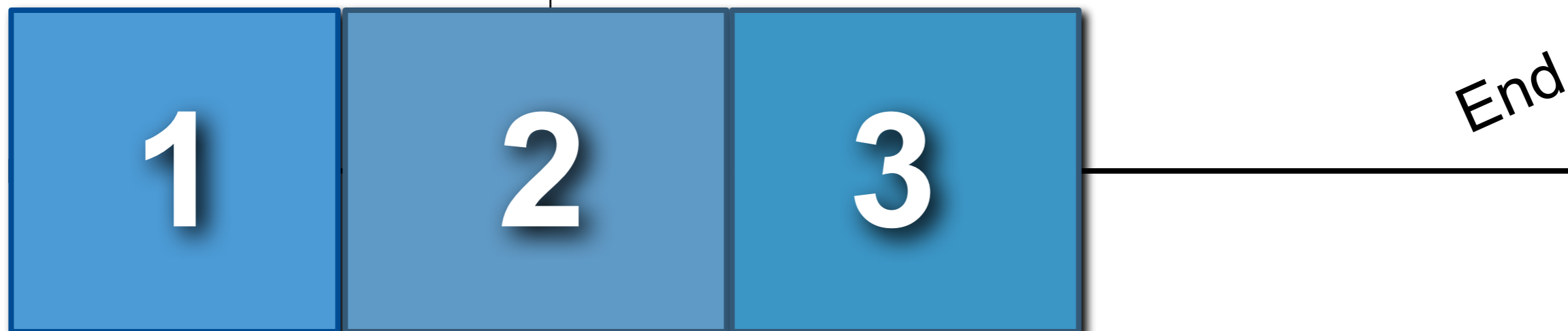


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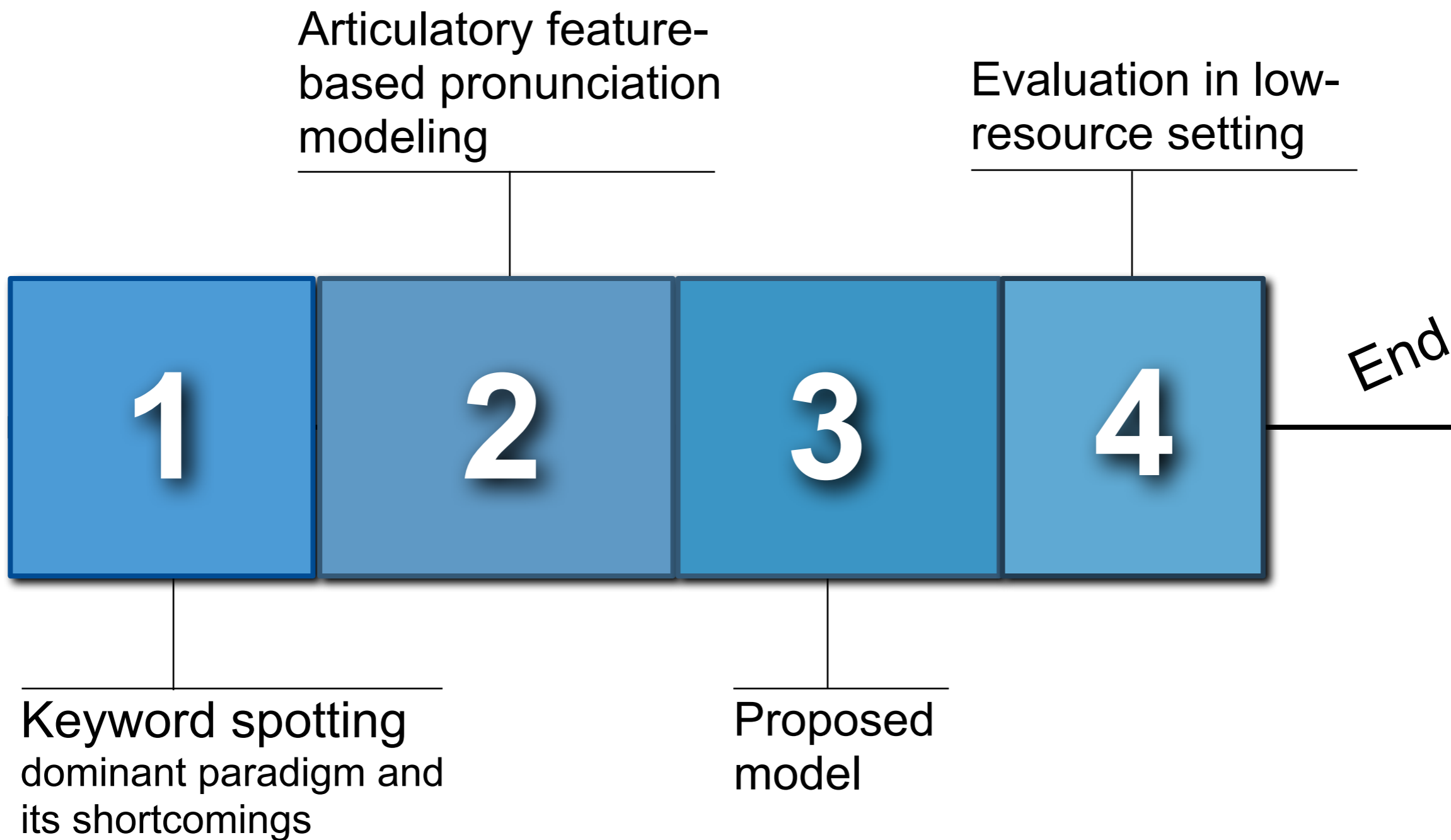
Articulatory feature-
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Keyword spotting
dominant paradigm and
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Proposed
model

Outline



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Articulatory feature-
based pronunciation
modeling

Evaluation in low-
resource setting

1

2

3

4

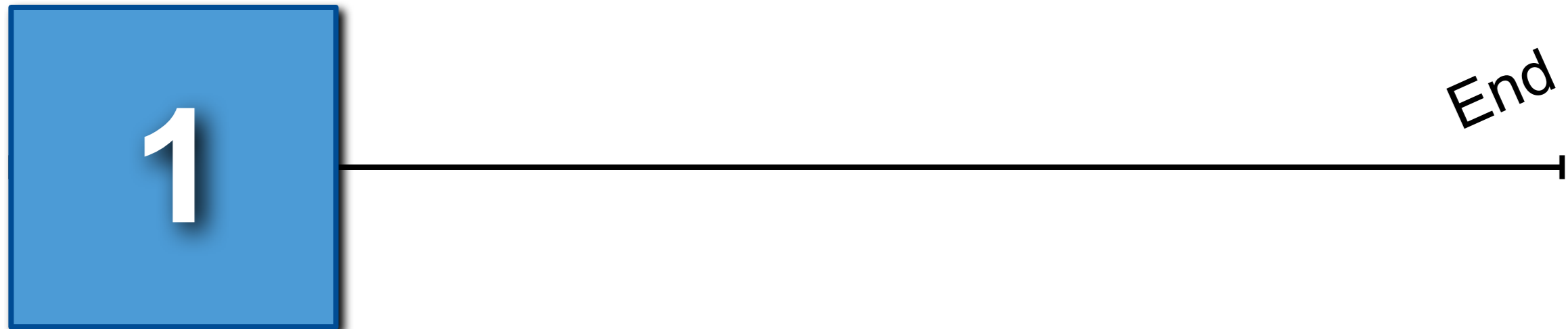
5

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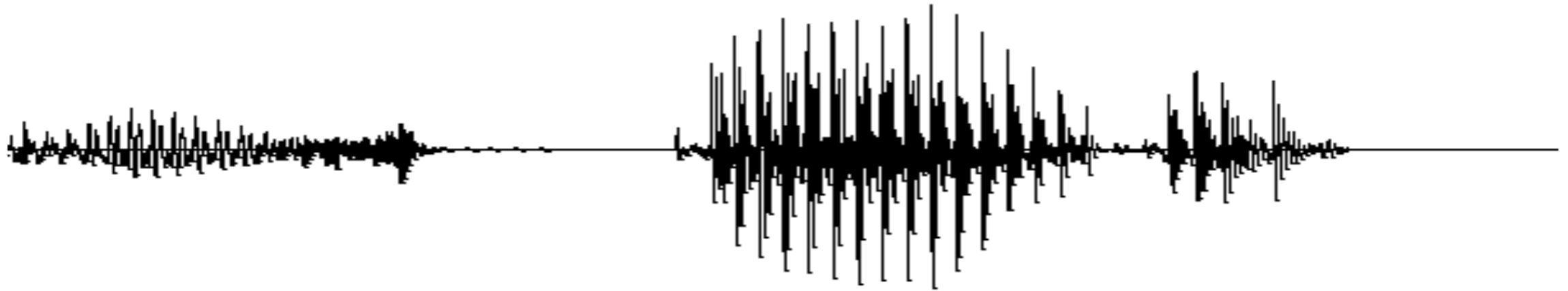
Conclusions

Outline



Keyword spotting
dominant paradigm and
its shortcomings

Given a speech signal

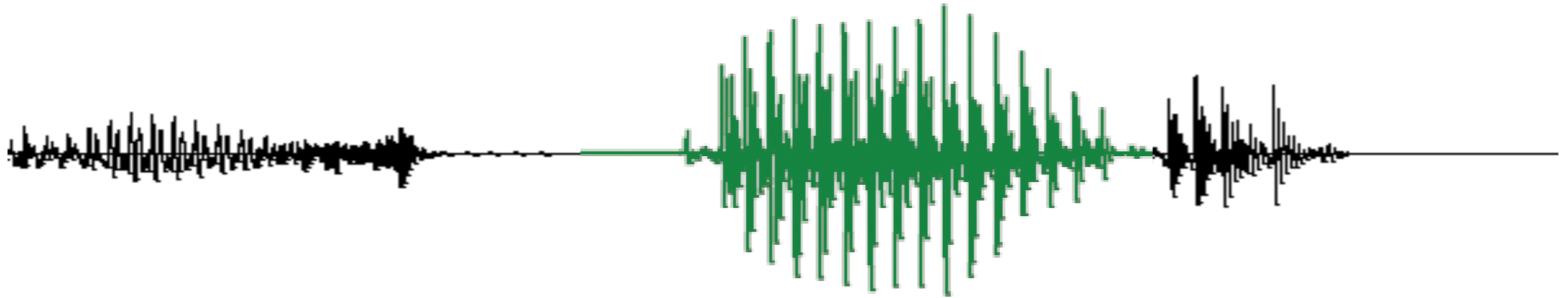


and a word

bought

Goal: find if the word is uttered in the speech signal
and where

Given a speech signal



and a word

bought

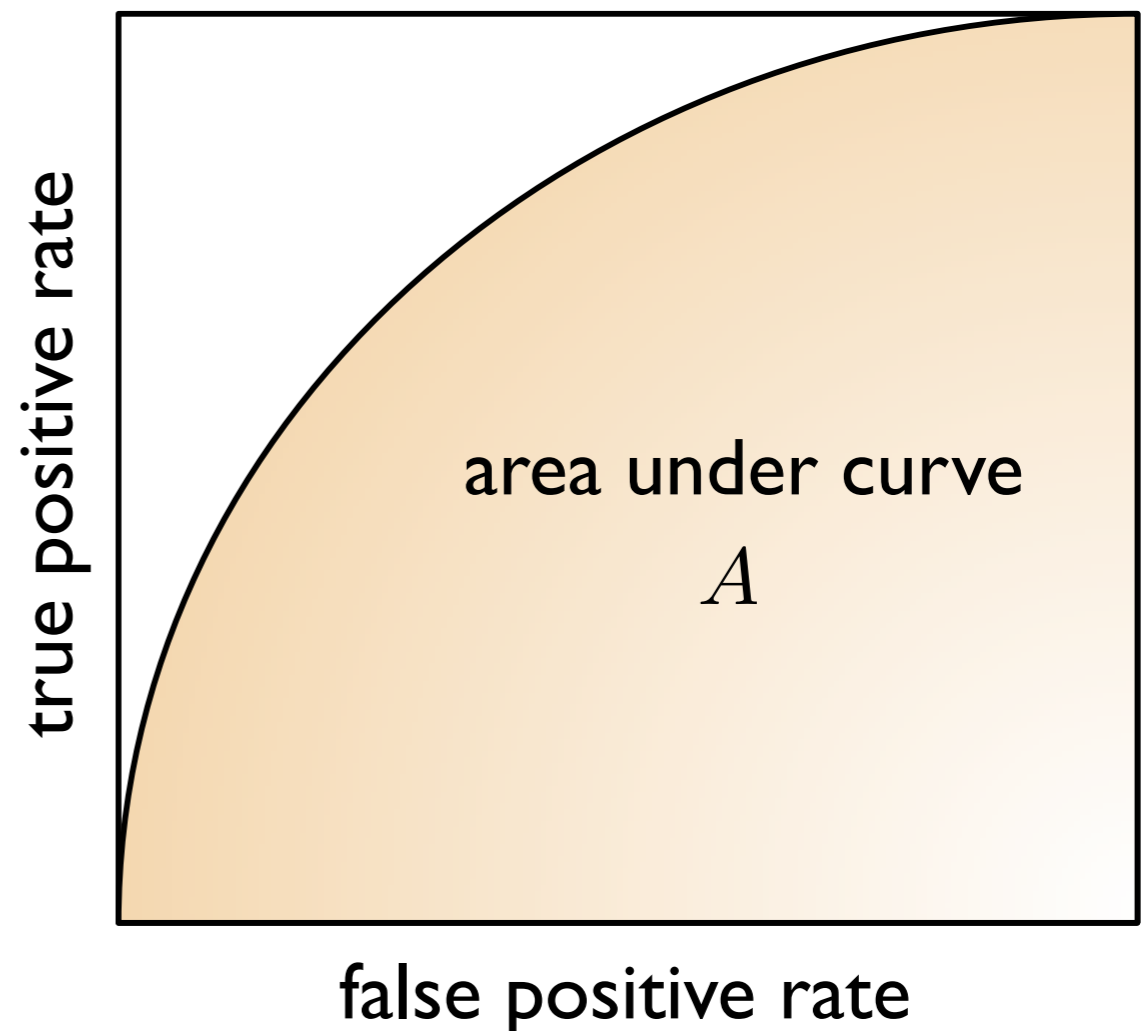
Goal: find if the word is uttered in the speech signal
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The task loss

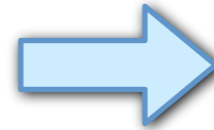
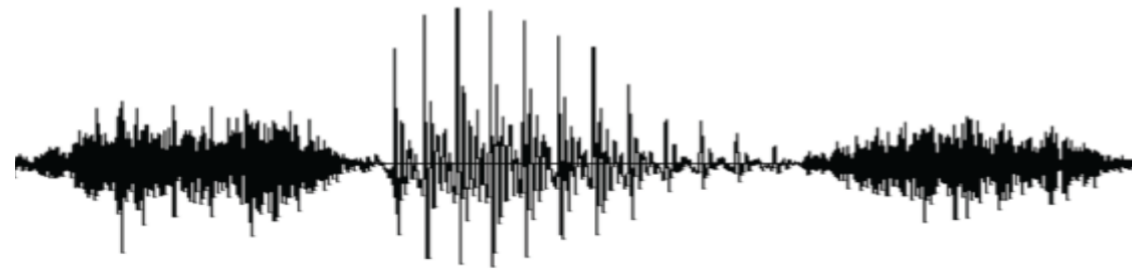
The performance of keyword spotting system is measured by Receiver Operating Characteristics (ROC) curve.

$$\text{true positive} = \frac{\text{detected utterances with keywords}}{\text{total utterances with keywords}}$$

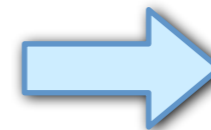
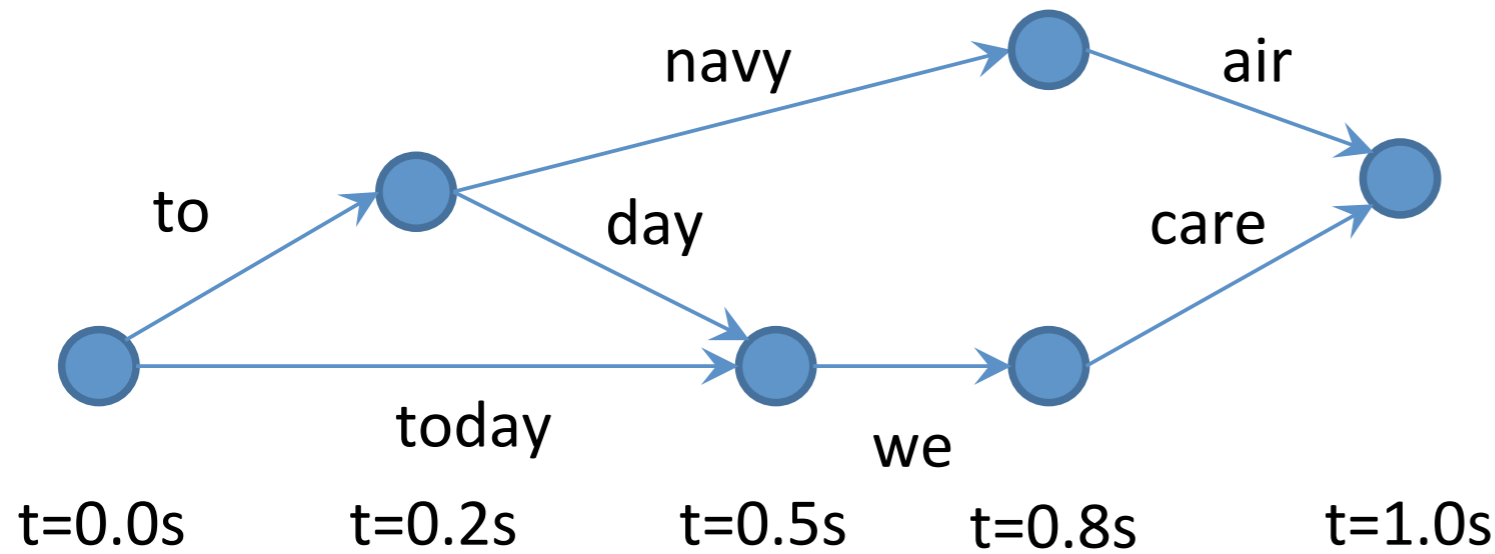
$$\text{false positive} = \frac{\text{detected utterances without keywords}}{\text{total utterances without keywords}}$$



Dominant Paradigm



Large Vocabulary
Continuous Speech
Recognizer



speech
index

Dominant Paradigm

- Common for LVCSR systems to have **millions** of free parameters
 - RWTH Gale Mandarin System $\approx 640\text{M}$ (Plahl et al. 09)
- Not always appropriate to **assume availability of large amounts of training data**
 - Rapid development of systems for low-resource languages
 - Porting keyword spotting systems to new acoustic conditions or speech styles

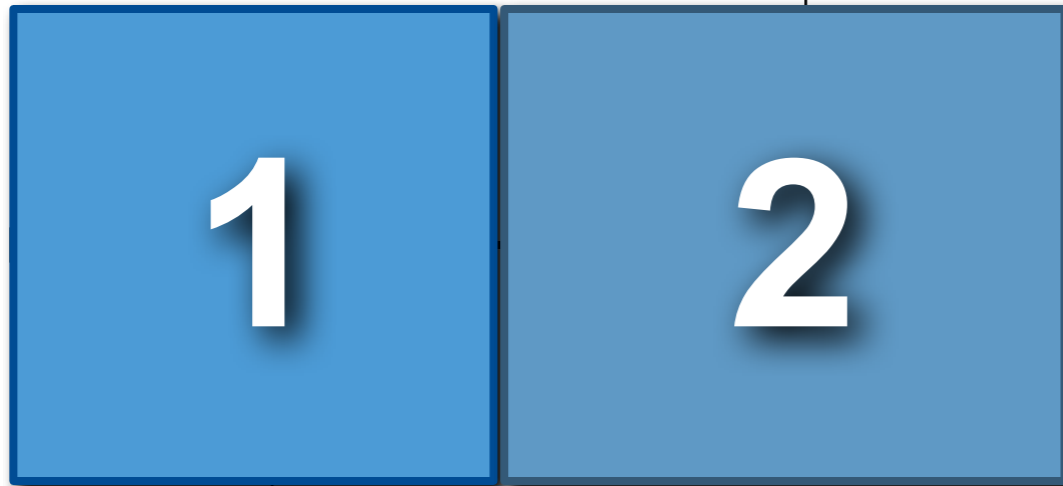
Contributions

Articulatory feature-based pronunciation modeling

Discriminative learning by maximizing the AUC
with large margin

Outline

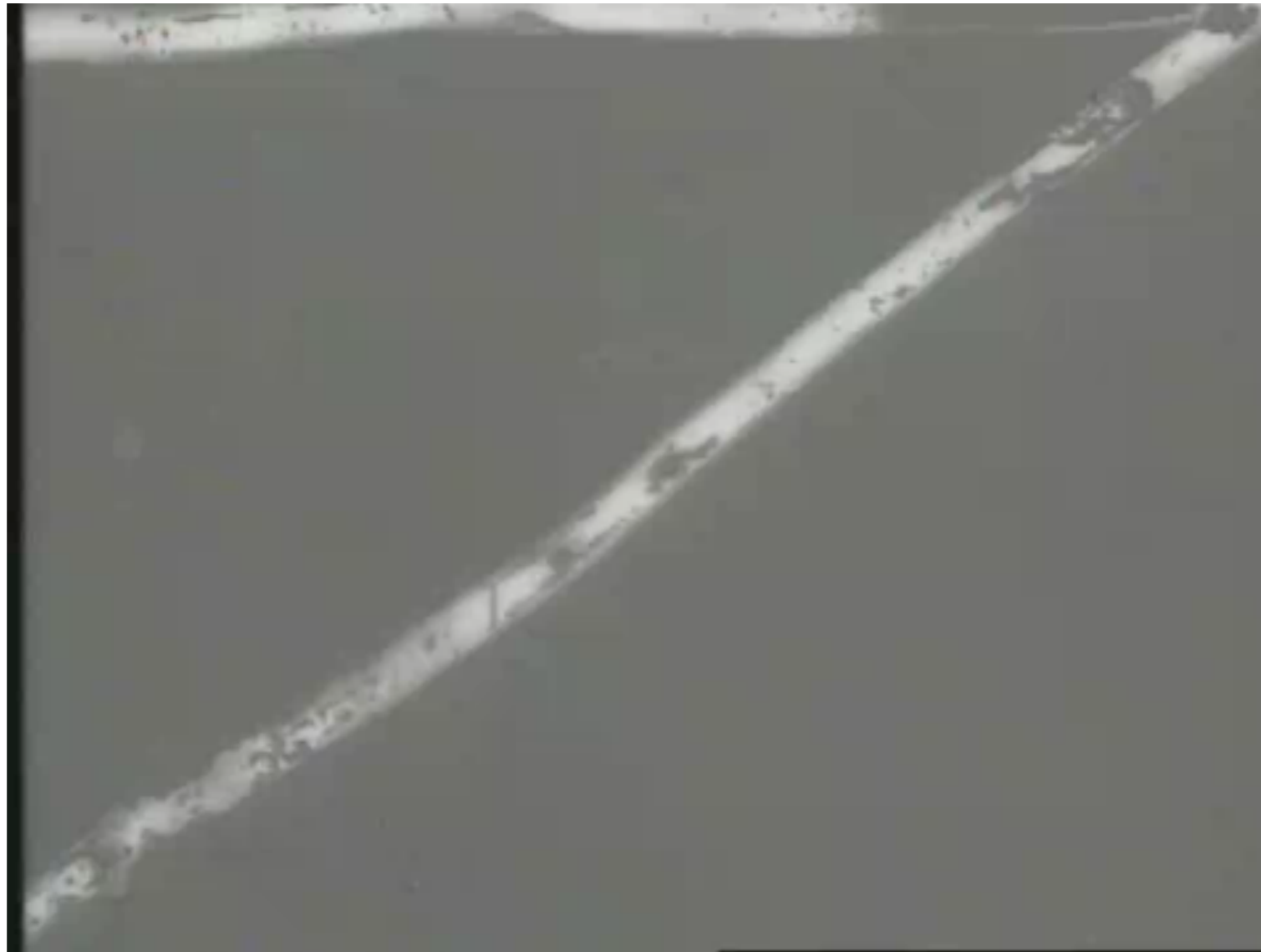
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End

What are articulatory features?



(video source: Ken Stevens, MIT)

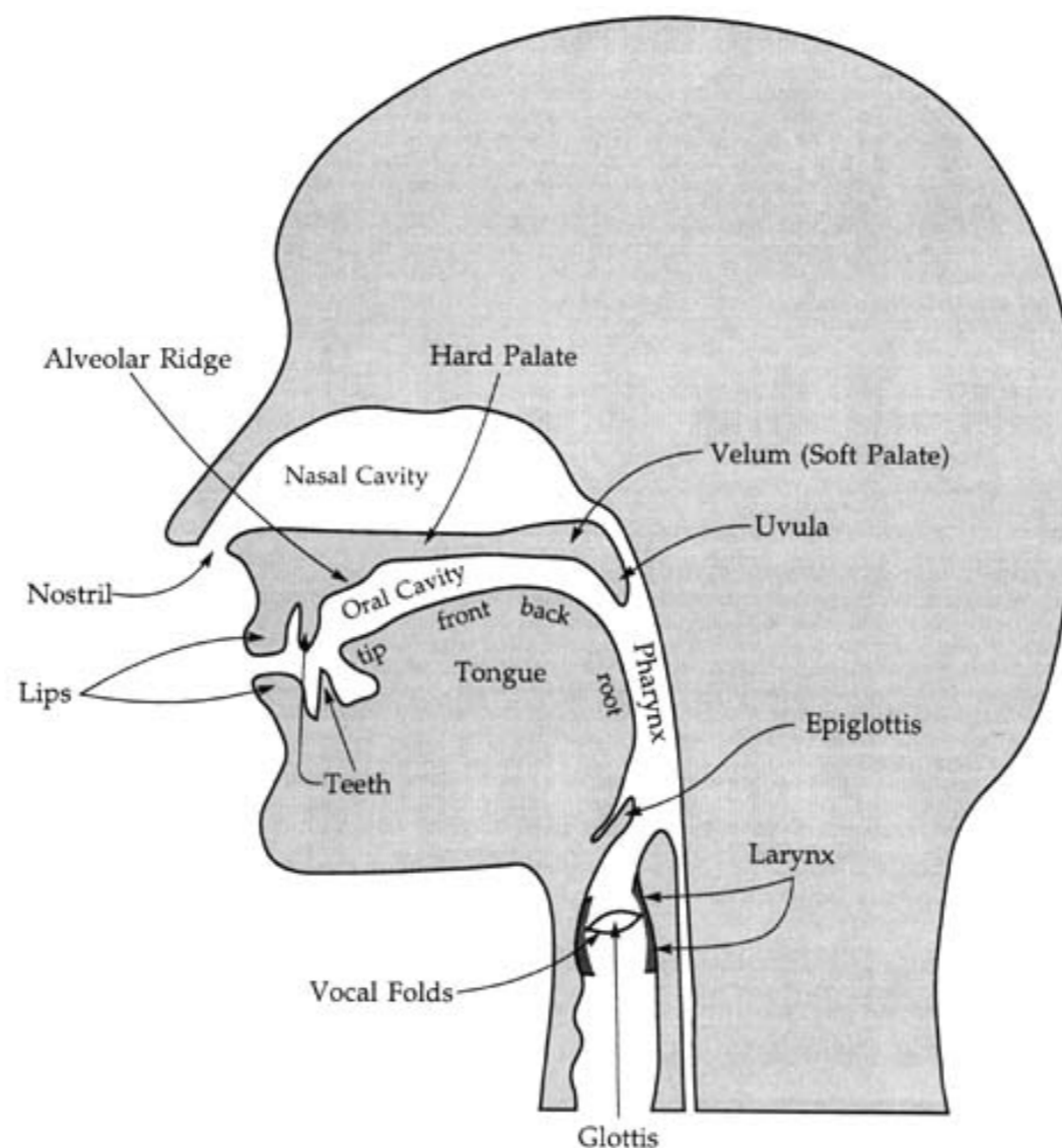
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Articulatory phonology

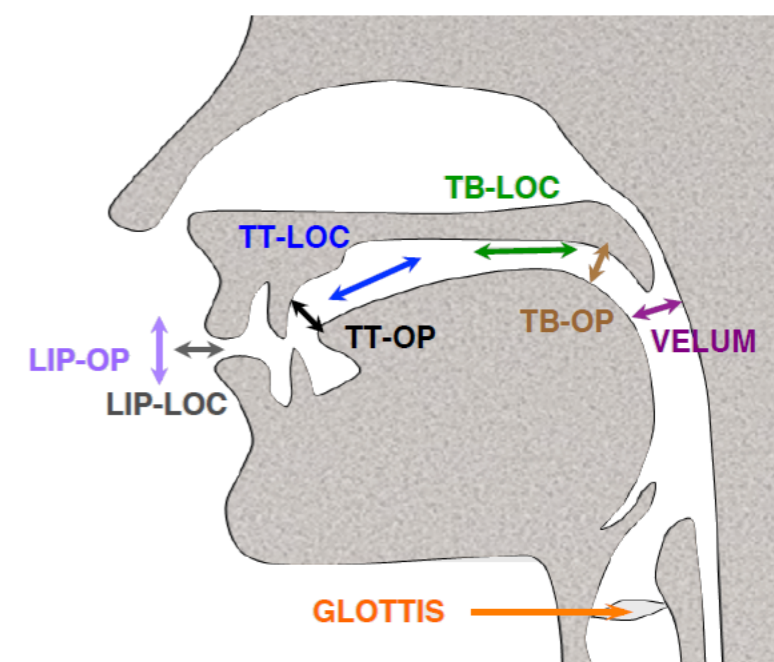
“pronunciation variations can be explained by asynchronization of the articulation”
(Browman and Goldstein, 1992)



Articulatory phonology

articulatory features (AF)

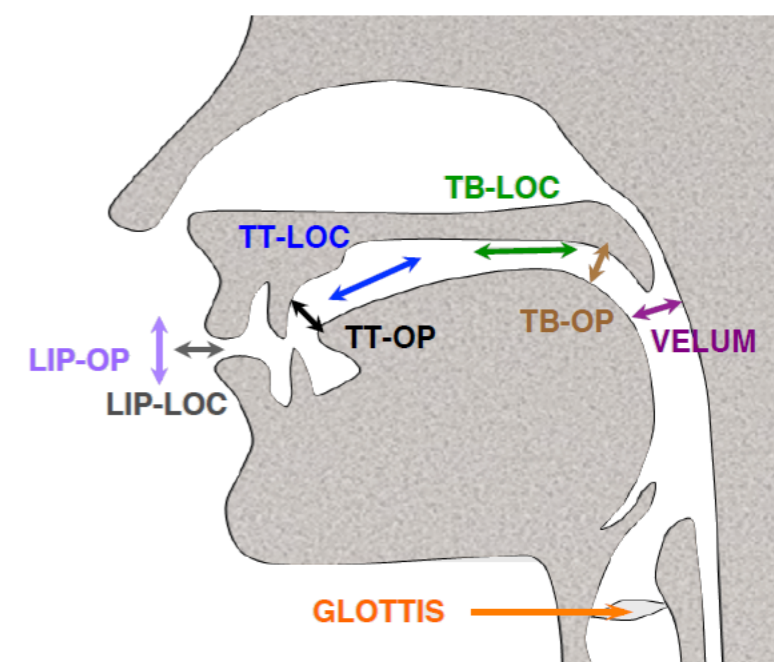
VEL	non-nasal	non-nasal	nasal	non-nasal
GLO	wide	critical	critical	wide
TB	uvular/medium	palatal/medium	uvular/medium	uvular/medium
TT	alveolar/ critical	alveolar/ medium	alveolar/closed	alveolar/critical
LIPS	wide/labial	wide/ labial	wide/labial	wide/labial
Phone	s	eh	n	s



Articulatory phonology

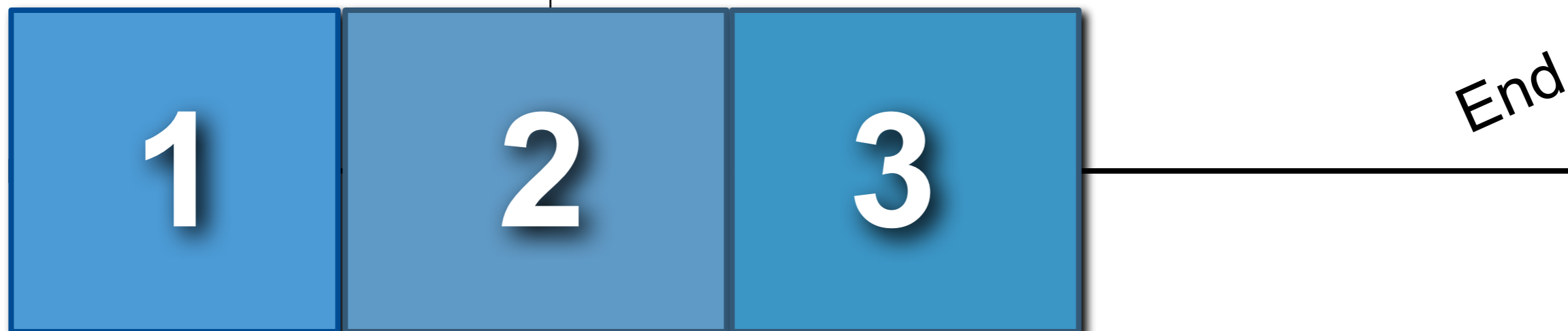
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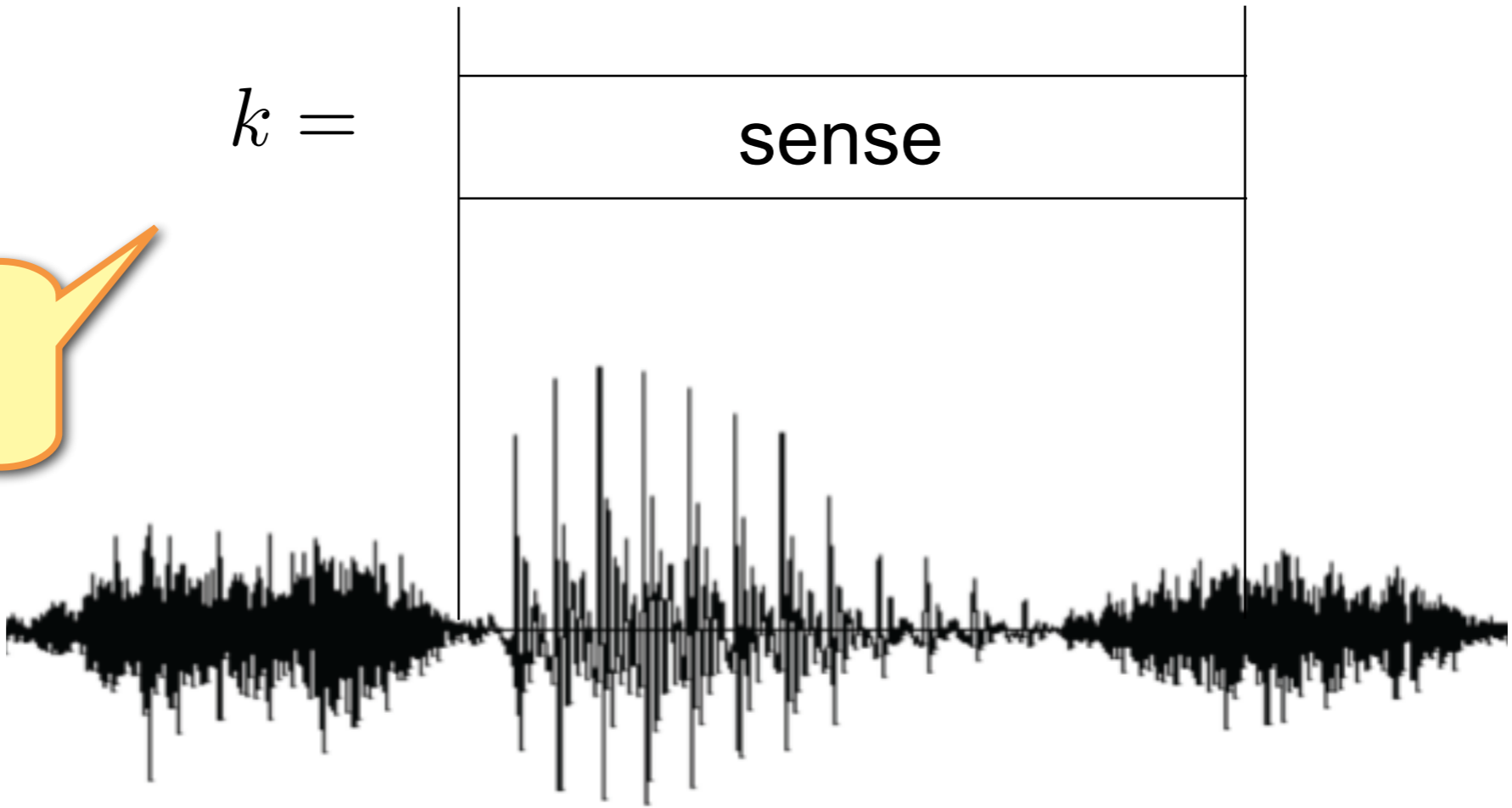
Proposed
model

timing

$$\bar{t} = (t_{\text{start}} \quad t_{\text{end}})$$



input keyword



$$\bar{\mathbf{x}} = (\mathbf{x}_1, \mathbf{x}_2 \quad \dots \quad \mathbf{x}_T)$$

sequence of acoustic vectors

speech
input signal

predicted
decision

\bar{x}

Keyword Spotter

detection (yes/no)

$k = \text{/bought/}$

$f(\bar{x}, k)$

\bar{t}'

input
keyword

predicted
timing



Model and inference

$$\bar{t}^* = f(\bar{\mathbf{x}}, k)$$

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weight
vector

$$\mathbf{w} \in \mathbb{R}^n$$

feature
map

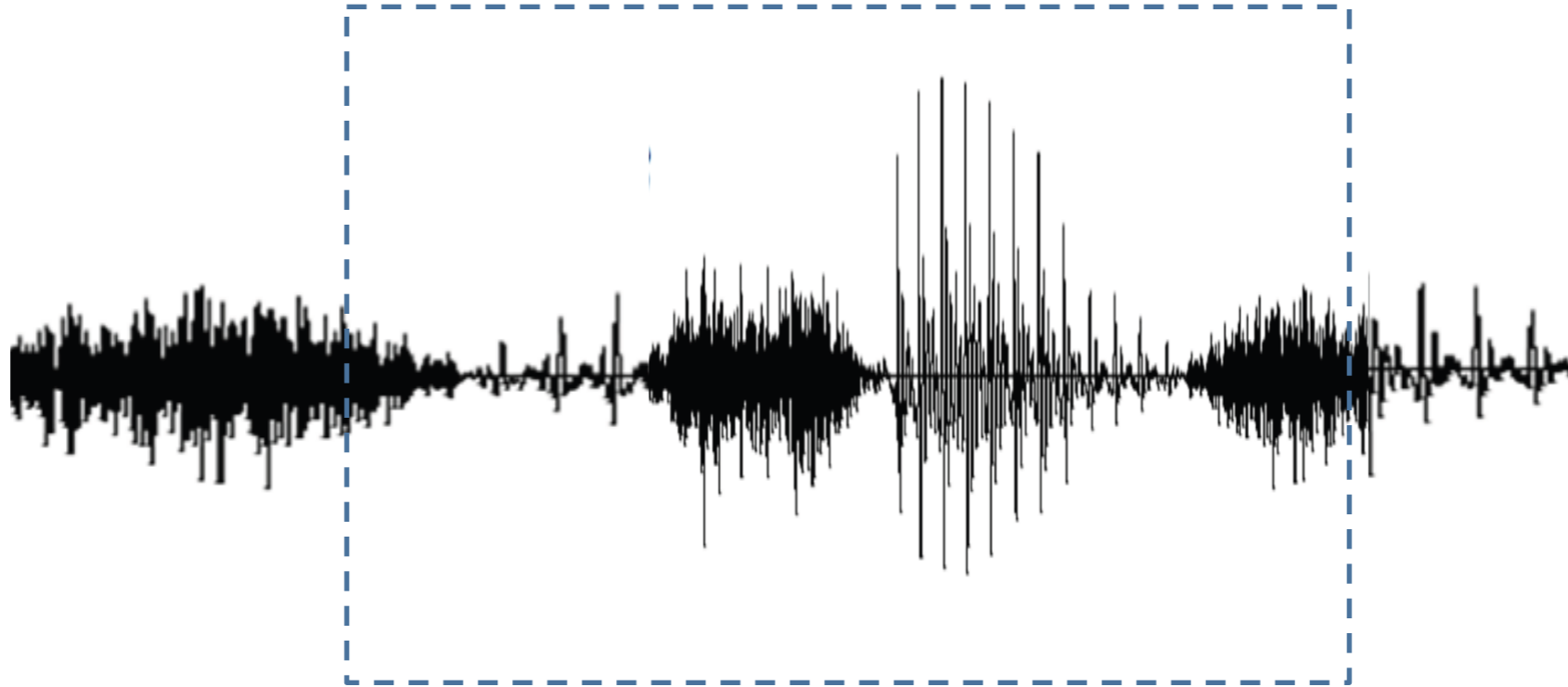
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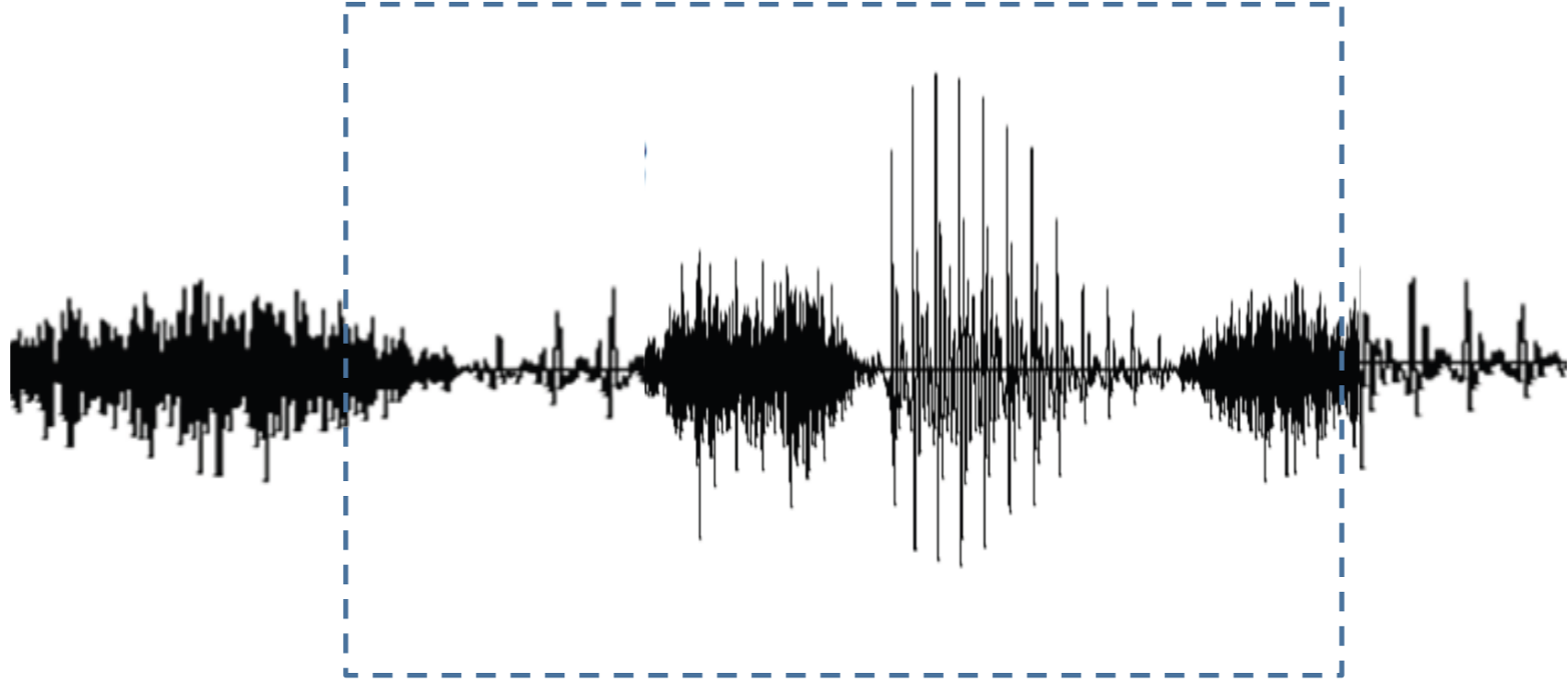
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Model and inference



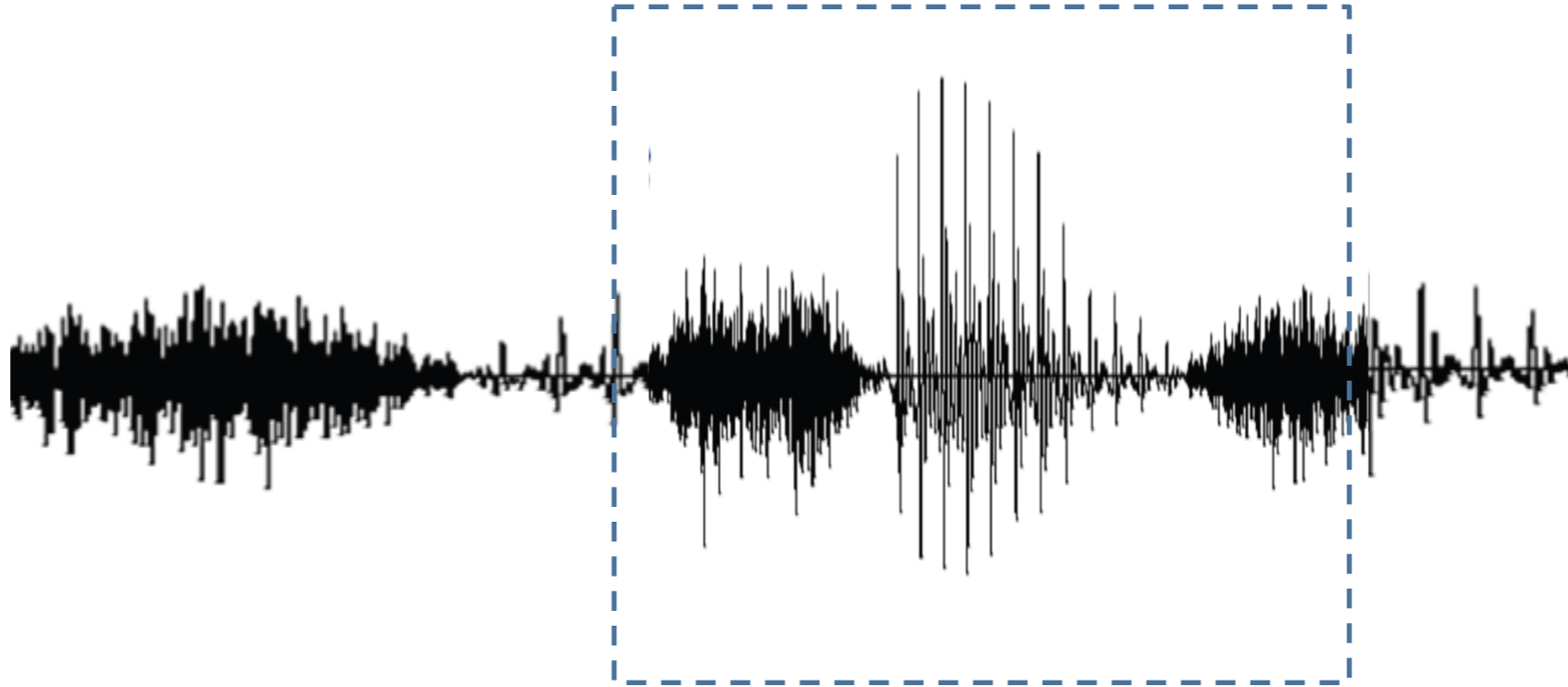
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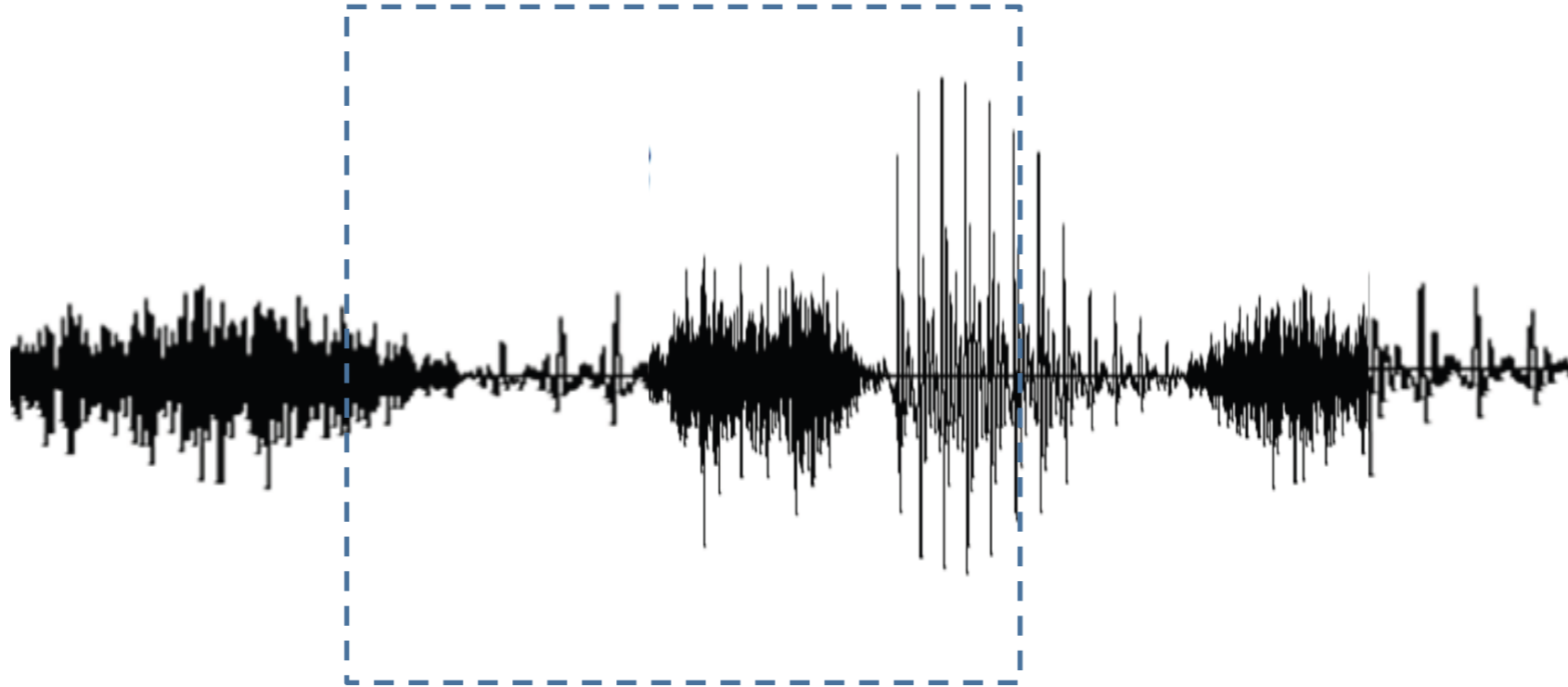
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Model and inference



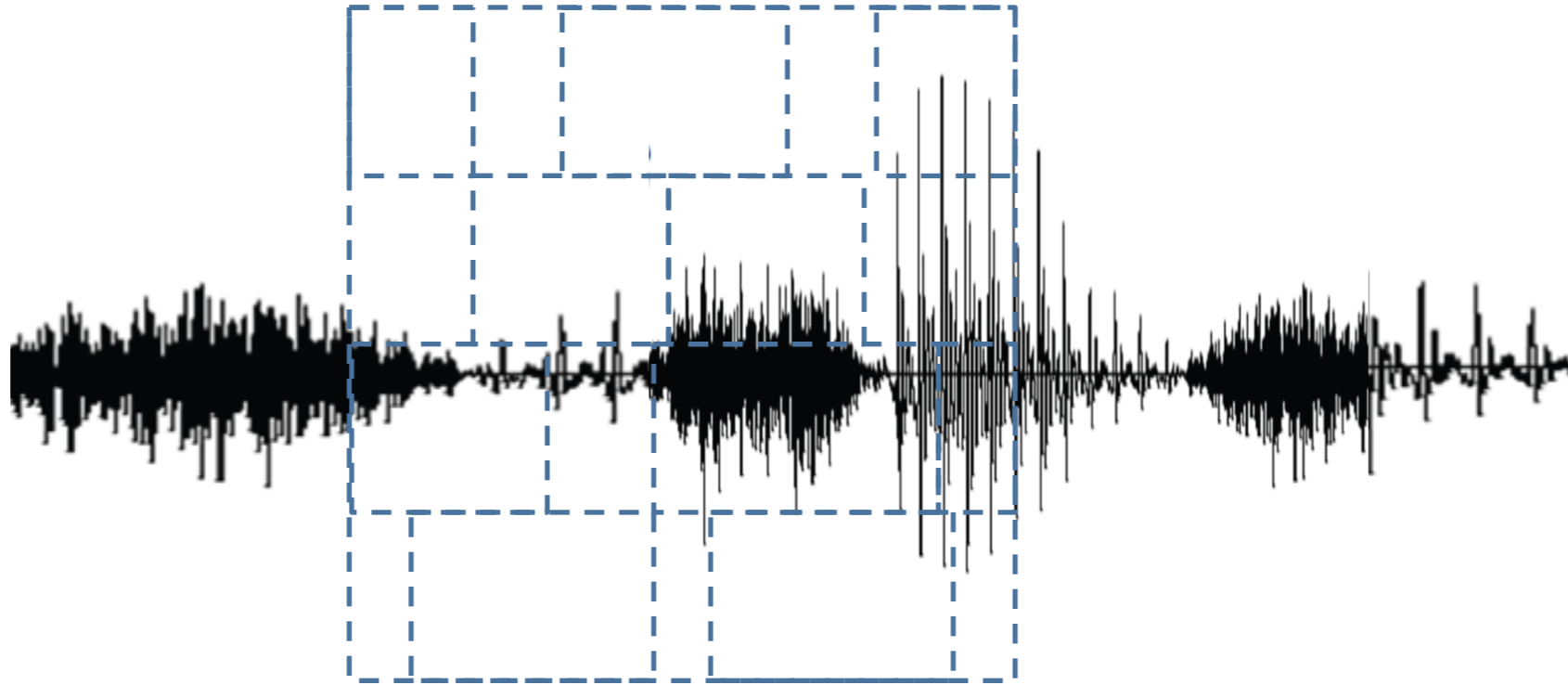
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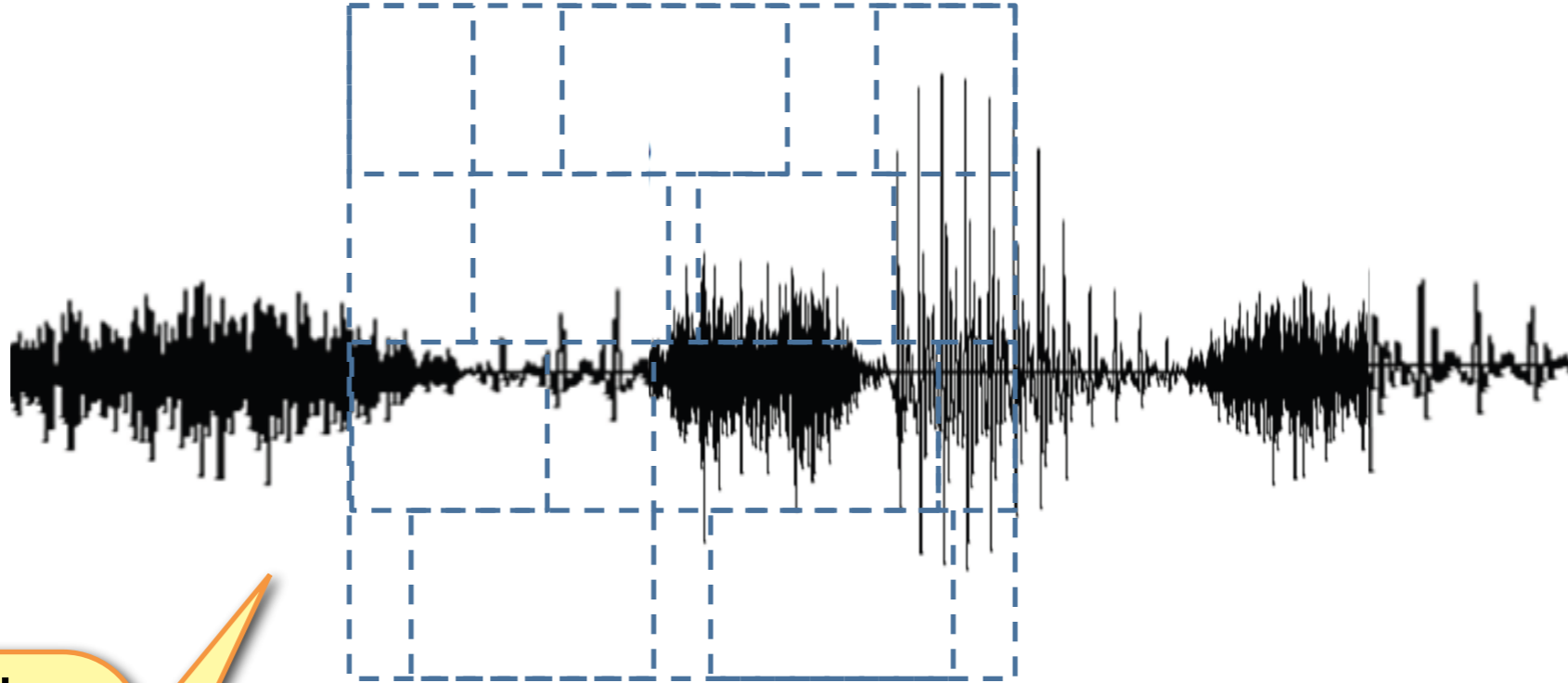
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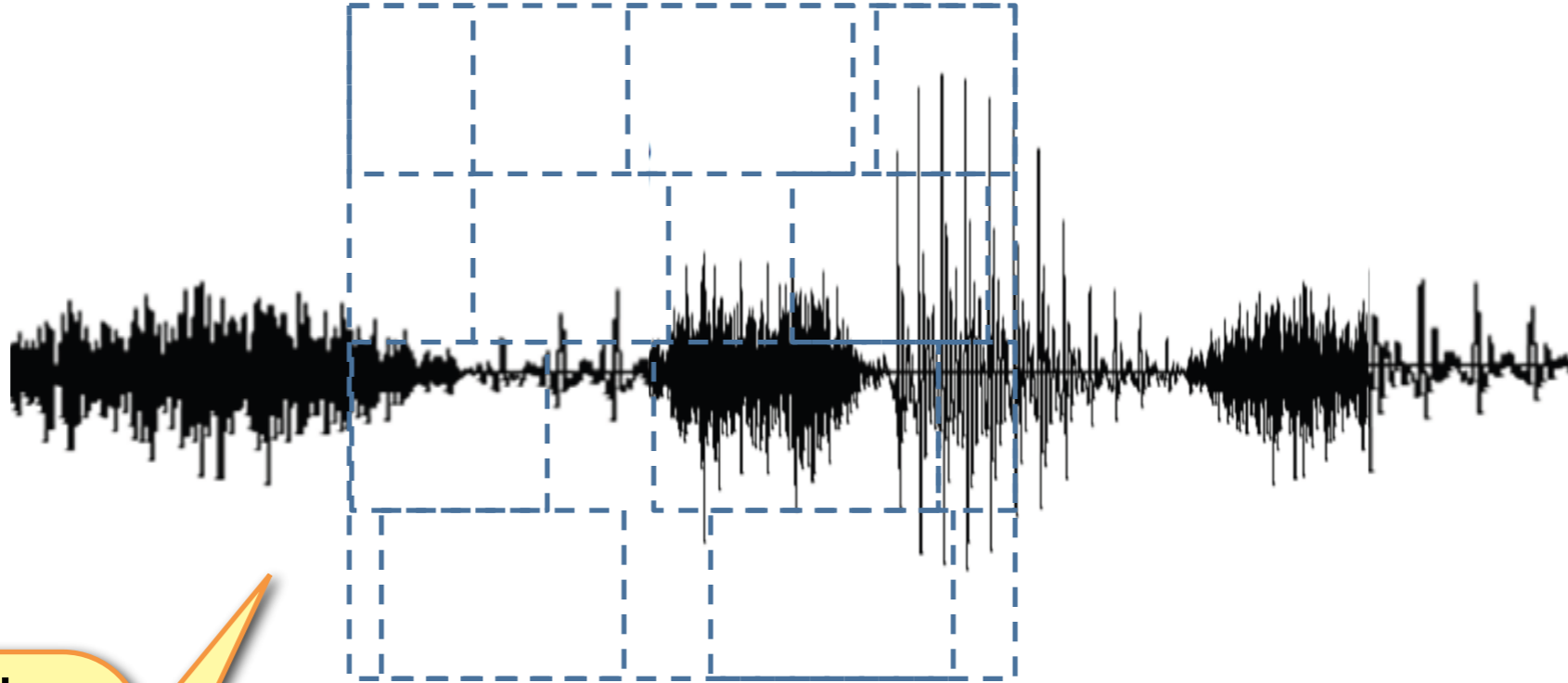
Model and inference



several
streams in the
case of
articulatory
features

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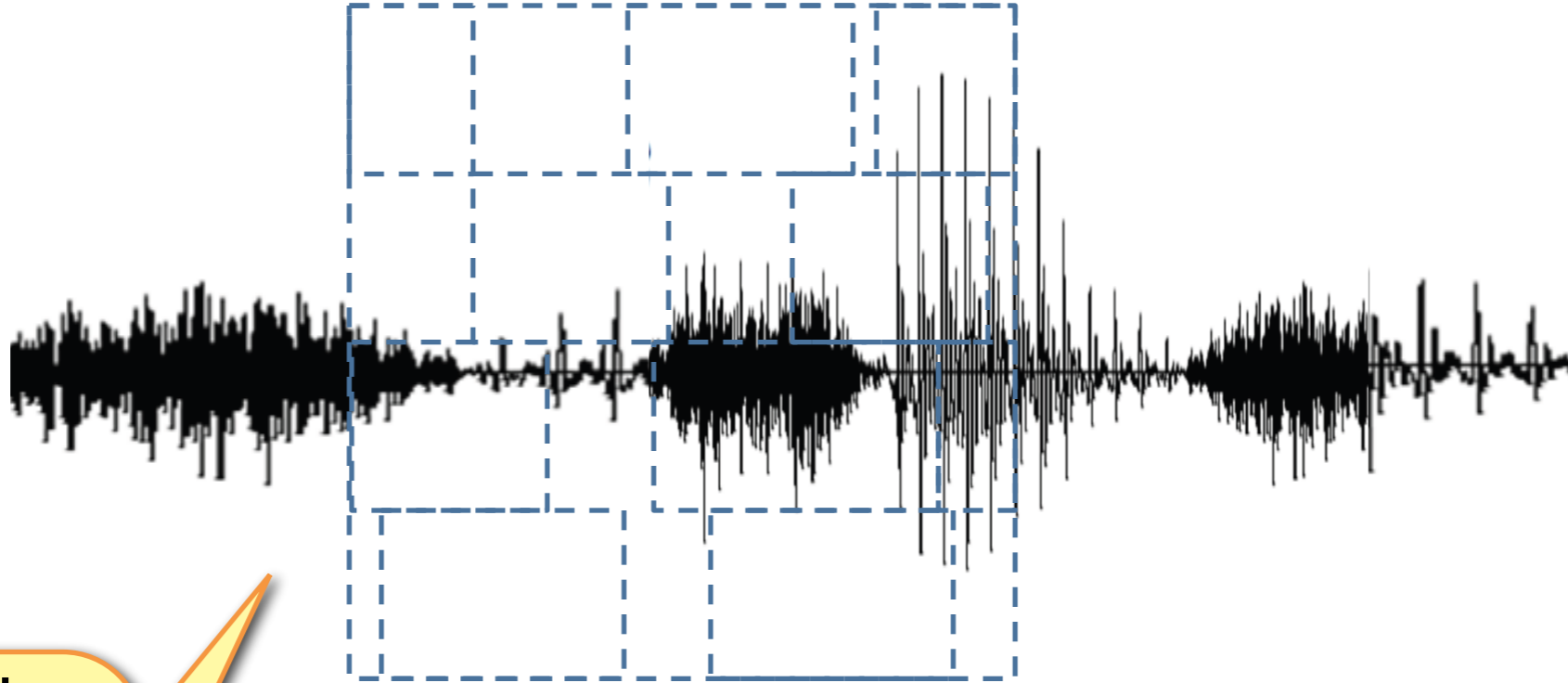
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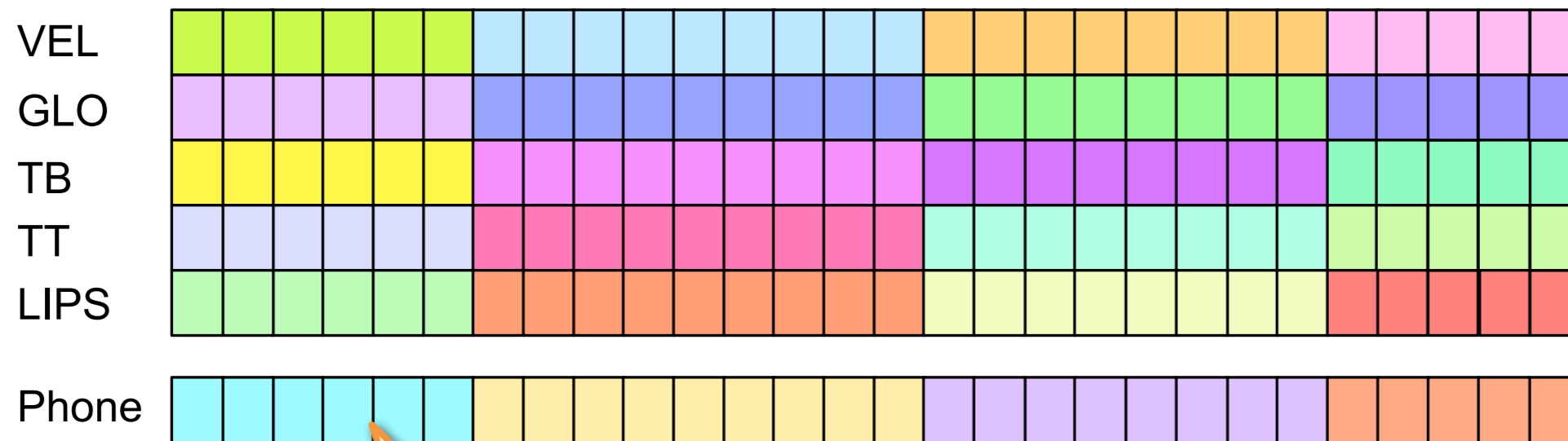
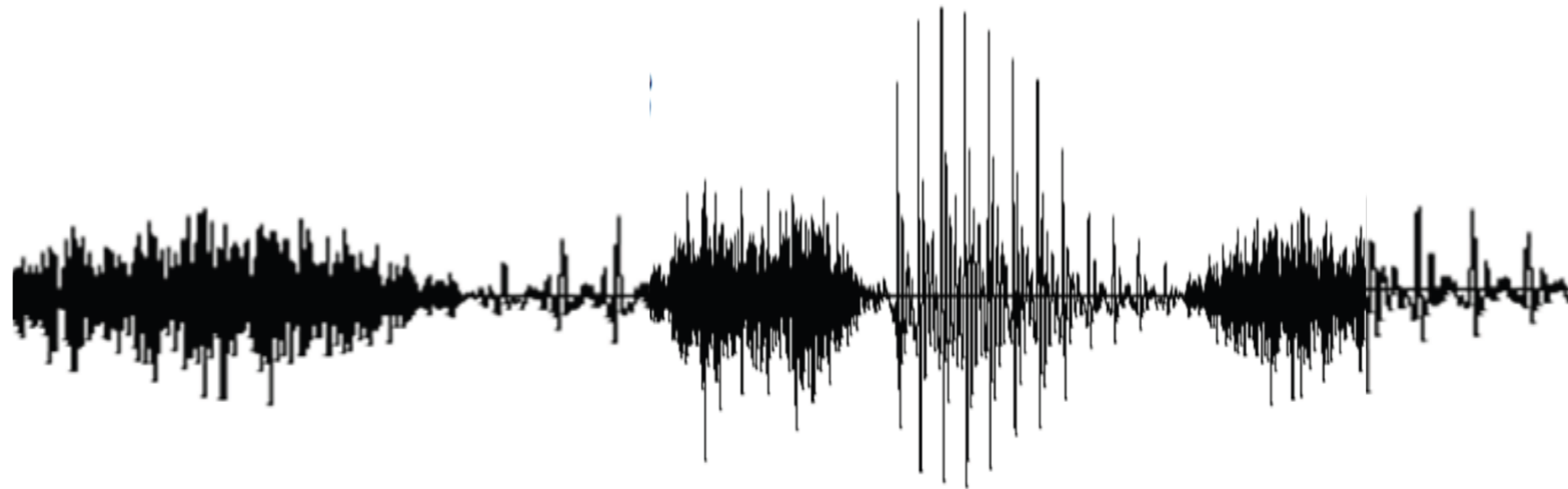
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weight
vector
 $\mathbf{w} \in \mathbb{R}^n$

feature
map

Feature map I

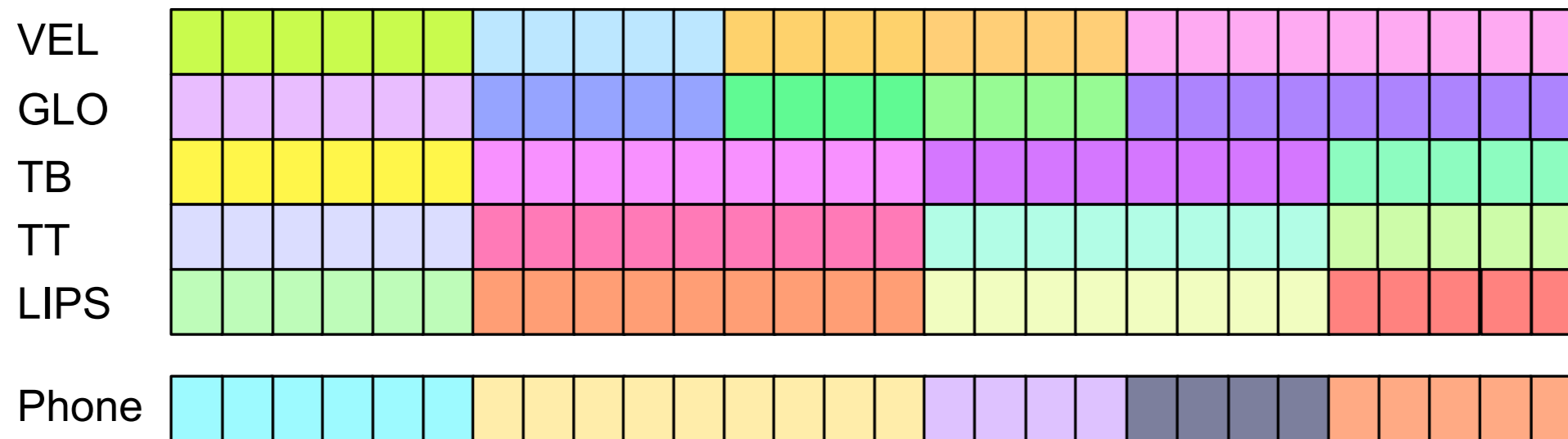
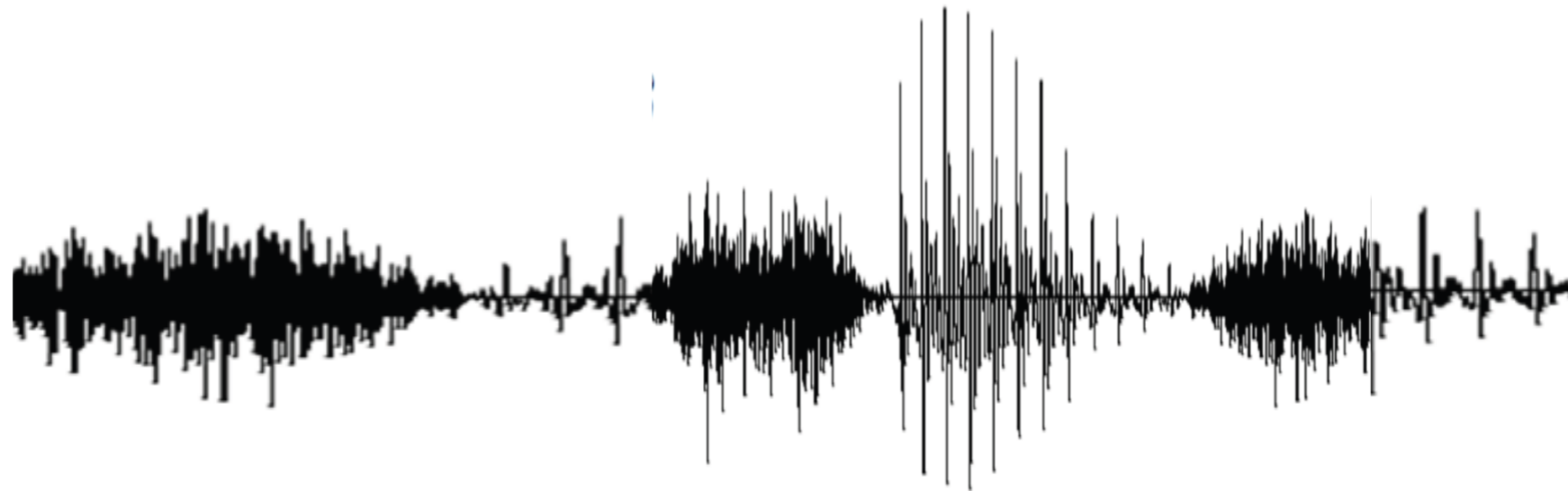
How likely is current frame to correspond to each of the AFs given segmentation?



MLP classifiers of phones and articulatory features

Feature map I

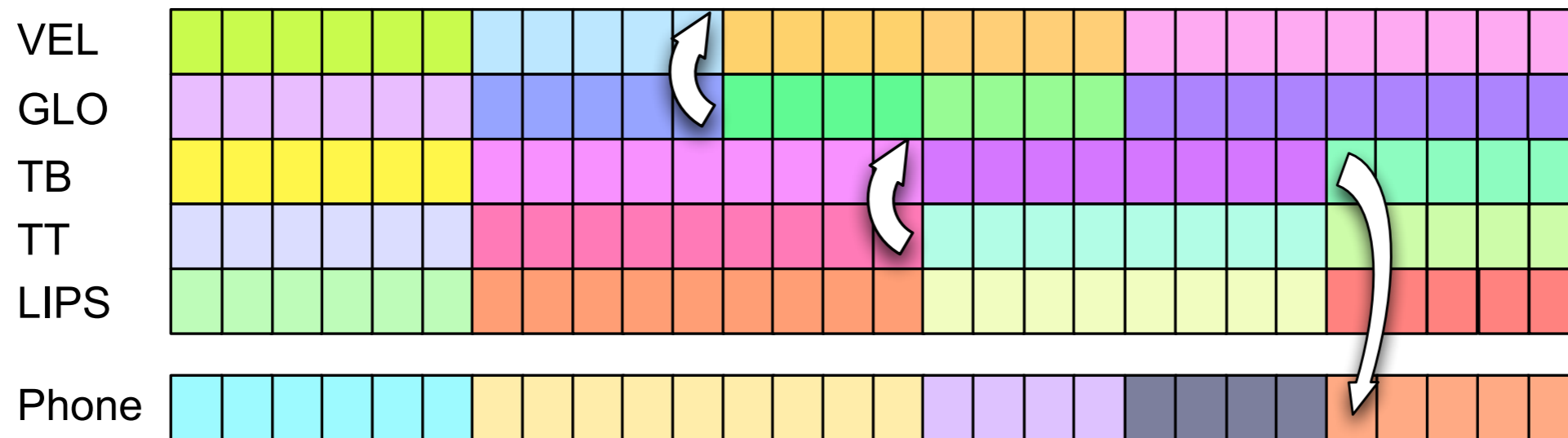
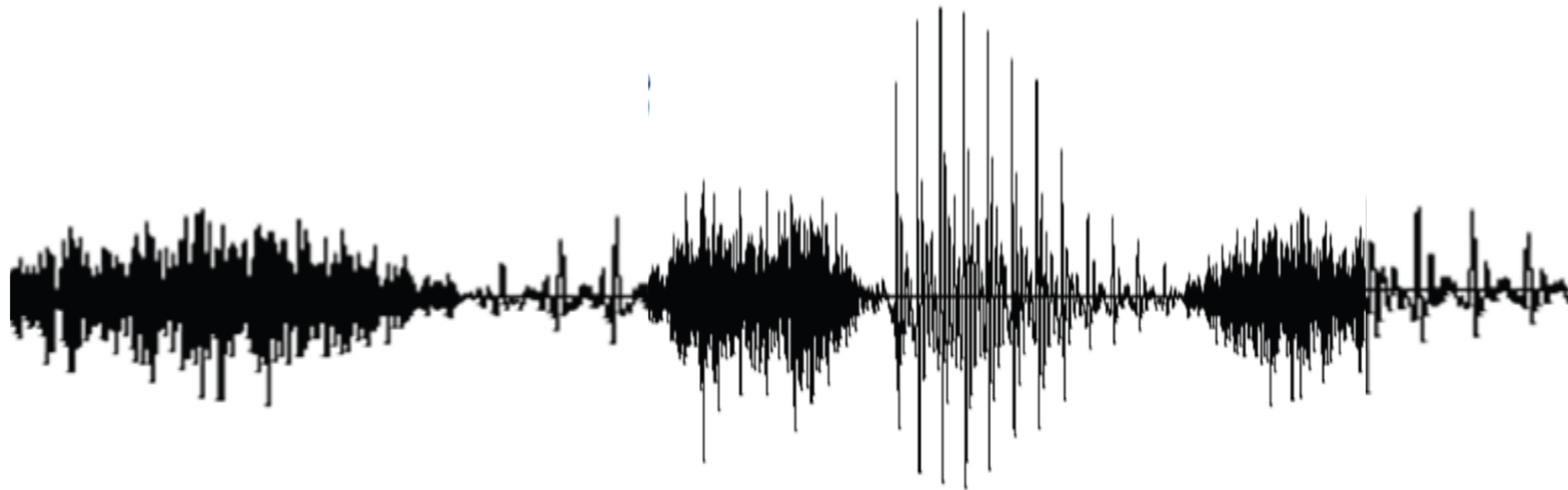
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MLP classifiers of phones and articulatory features

Feature map II

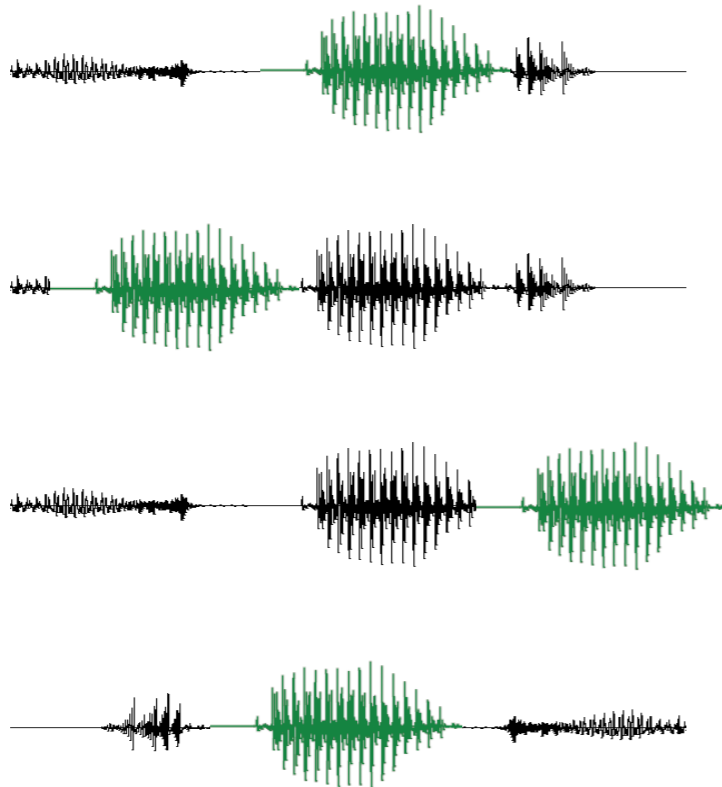
How likely is AF in stream i at previous frame corresponds to AF stream j at current frame



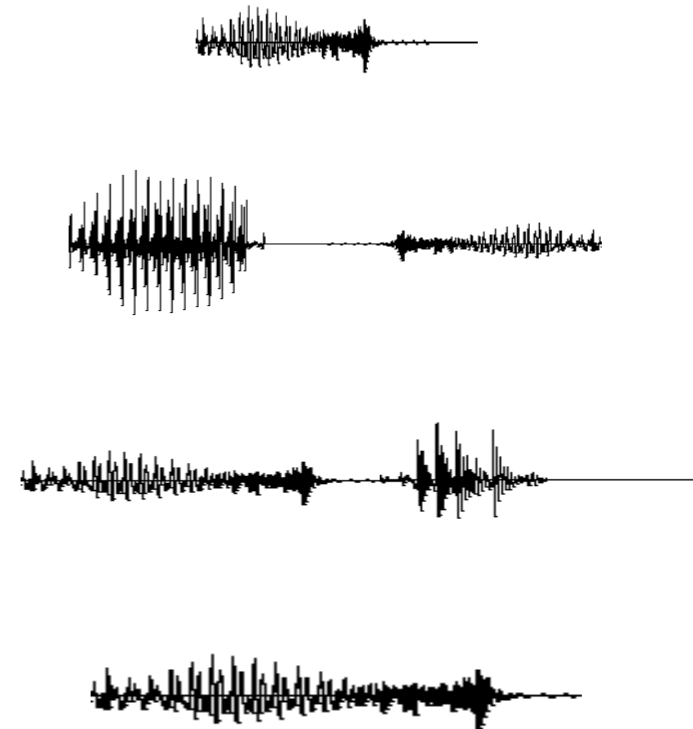
Maximizing area under ROC (AUC)

For every event (keyword) k define two sets of input signals (speech utterances):

\mathcal{X}_k^+



\mathcal{X}_k^-



Maximizing area under ROC (AUC)

By definition of the area under the ROC:

$$A = \mathbb{P} \left[\max_{\bar{t}} f_{\mathbf{w}}(\bar{\mathbf{x}}^+, k, \bar{t}) > \max_{\bar{t}} f_{\mathbf{w}}(\bar{\mathbf{x}}^-, k, \bar{t}) \right]$$

Maximizing area under ROC (AUC)

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$$\mathbf{w}^* = \arg \min_{\mathbf{w}} \frac{1}{m} \sum_{i=1}^m \left[1 - \max_{\bar{t}} f_{\mathbf{w}}(\bar{\mathbf{x}}_i^+, k_i, \bar{t}) + \max_{\bar{t}} f_{\mathbf{w}}(\bar{\mathbf{x}}_i^-, k_i, \bar{t}) \right]_+ + \frac{\lambda}{2} \|\mathbf{w}\|^2$$

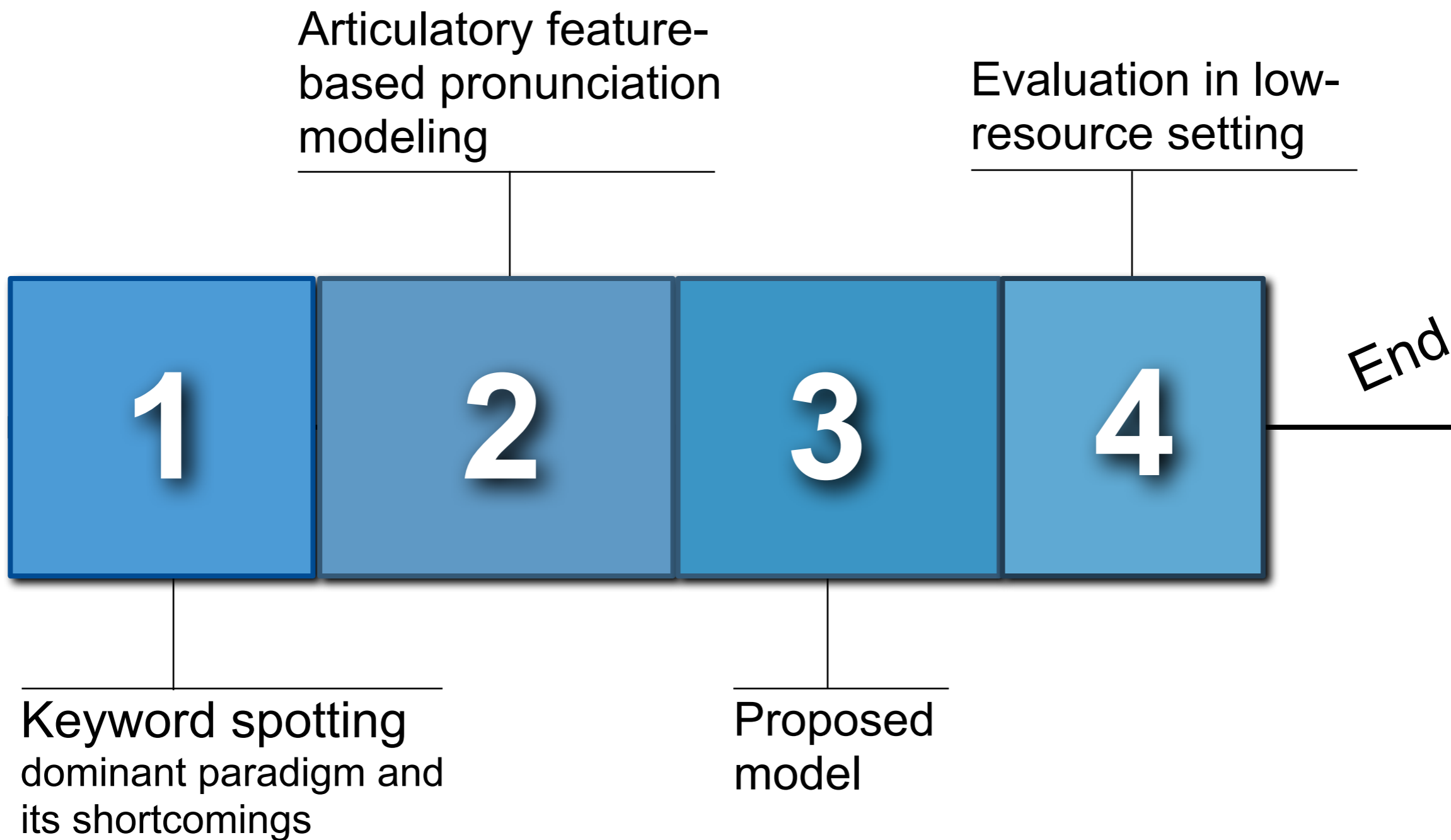
(Keshet, Grangier and Bengio, 2009)

Implementation

- Iterative algorithm to solve the optimization problem efficiently on huge data (millions of examples)
- Theorems support the maximization of AUC

(**Keshet**, Grangier and Bengio, 2009; Prabhavalkar, **Keshet**, Livescu and Fosler-Lussier, 2012)

Outline



Experiments

- Constructed four corpora containing 500-5000 utterances respectively by randomly selecting utterances from Switchboard
- Development set (40 terms) and Test set (60 terms)
 - 20 positive and negative sentences each

Utterances	500	1000	2500	5000
Training Data	0.8 hrs	1.5 hrs	3.7 hrs	7.4 hours

Experiments

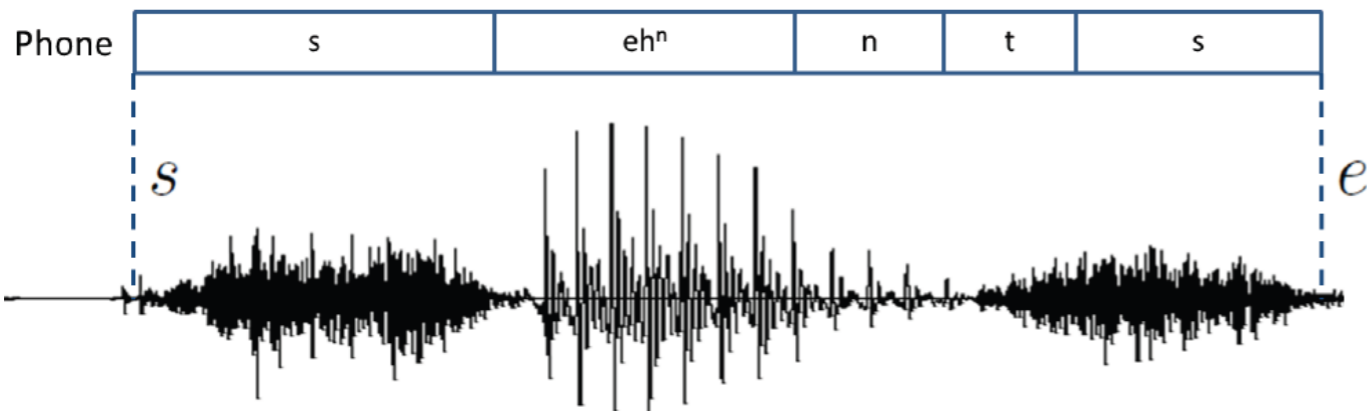
- Creation of “positive” and “negative” examples from training data
 - Each word with at least 5 phonemes in pronunciation chosen as “positive example”
 - Randomly selected utterance not containing word from training data as corresponding “negative example”

Utterances	500	1000	2500	5000
Positive Examples	1538	2876	7245	14570

Experiments

VEL	non-nasal (σ_1^1)	non-nasal (σ_2^1)	nasal (σ_3^1)	non-nasal (σ_4^1)
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GLO	wide	critical	critical	wide
TB	uvular/medium	palatal/medium	uvular/medium	uvular/medium
TT	alveolar/critical	alveolar/medium	alveolar/closed	alveolar/critical
LIPS	wide/labial	wide/labial	wide/labial	wide/labial



Articulatory Stream	State Space Size
Lips (L)	8
Tongue (T)	25
Glottis/Velum (G)	5

- Enforce synchrony for Lip features (L); Tongue features (T); combination of Glottis and velum (G)
- Allow at most one state of asynchrony between streams

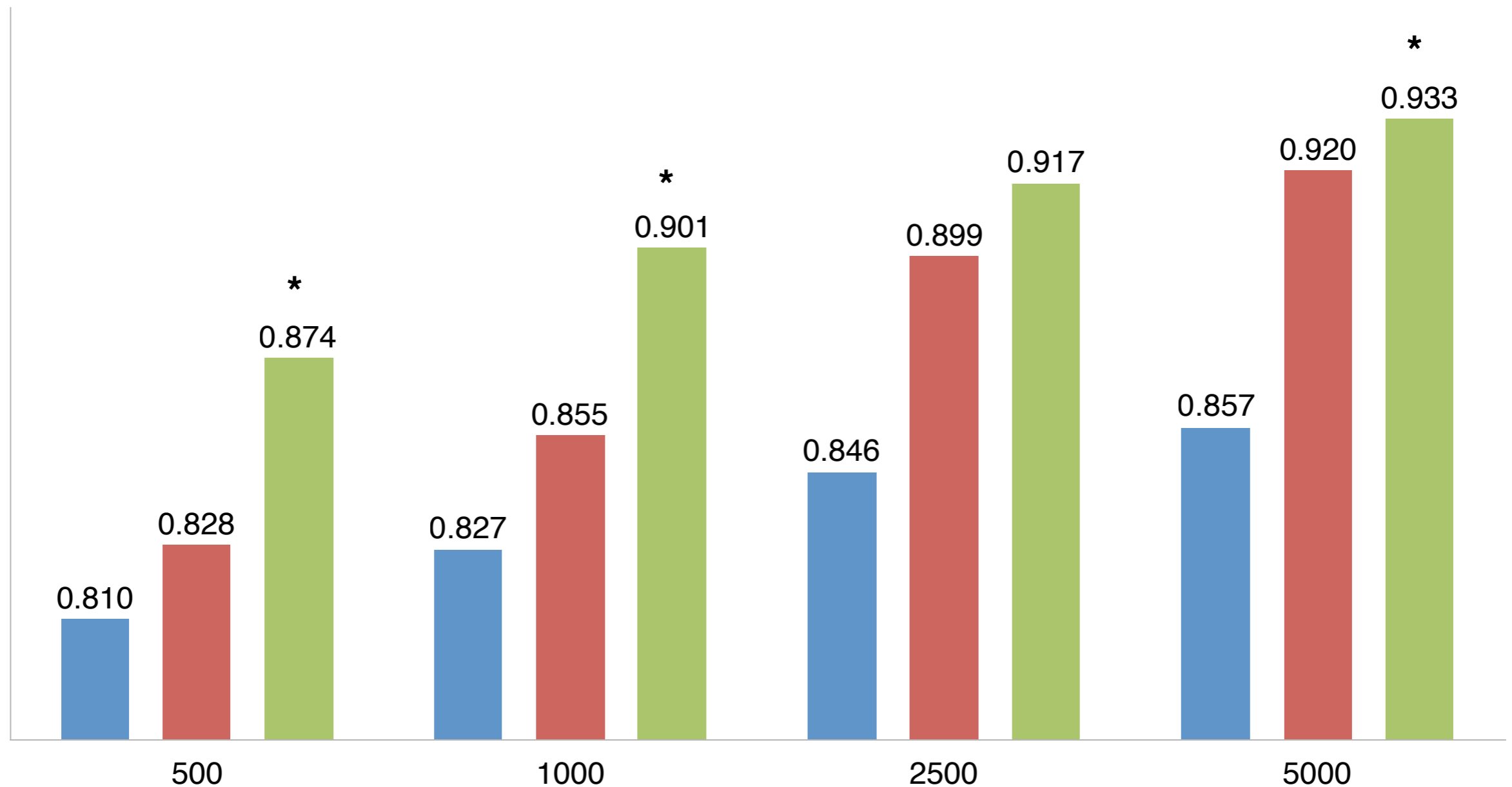
Experiments

- MLPs trained on Switchboard Transcription Project (STP) (Greenberg et al. 96) data to predict phones and L, T, G labels
- “Tandem” feature extraction: projected computed phone and L, T, G log posteriors on to top 39 principal components using PCA
 - “Tandem” features used as acoustic features in baseline monophone/triphone GMM-HMM keyword-filler and discriminative systems

Results: HMM, Disc-Phone

- HMM-mono
- HMM-tri
- Disc-Phone

AUC performance

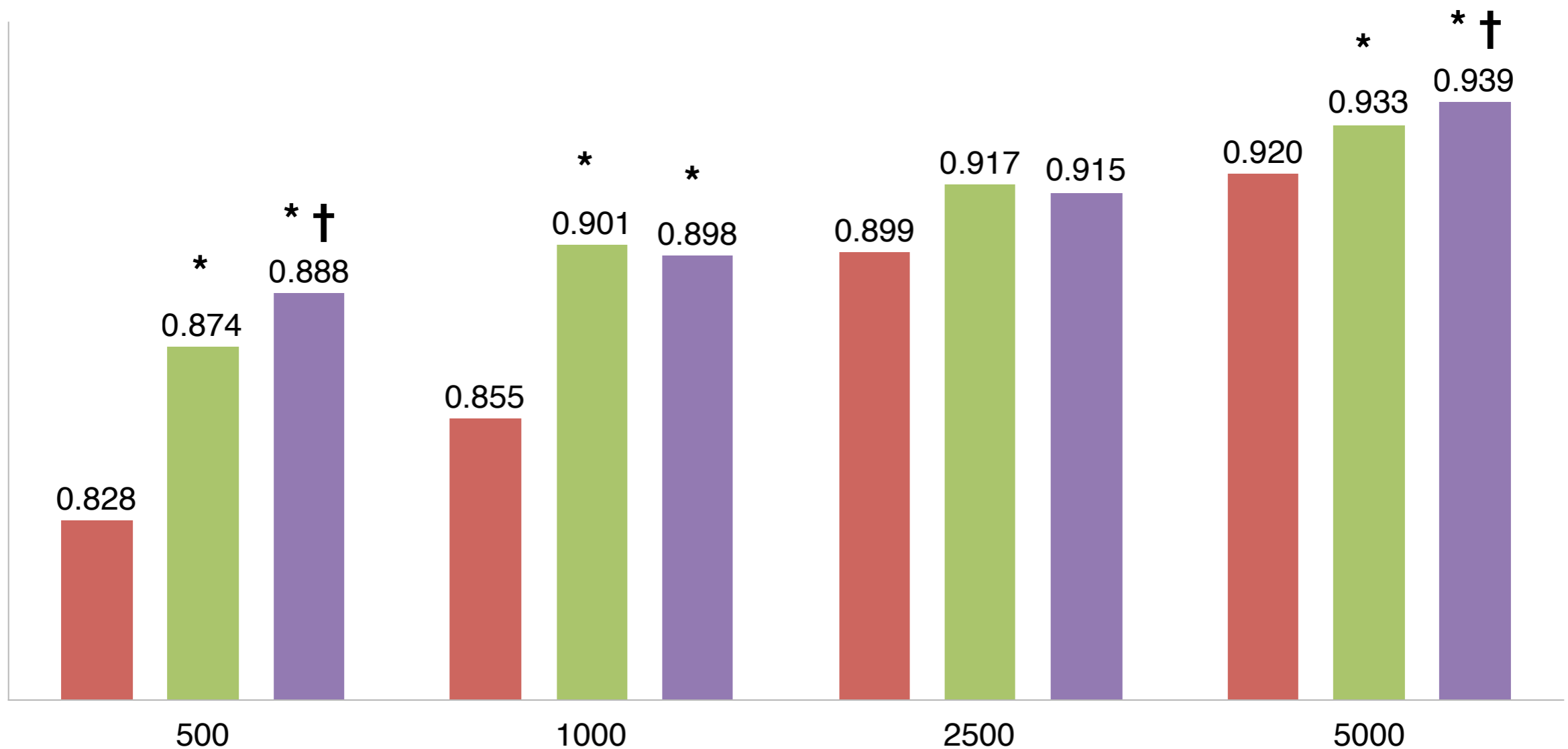


* : significant ($p \leq 0.05$) difference over HMM-tri

Results: HMM, Disc-Phone, Disc-AF

■ HMM-tri
■ Disc-Phone
■ Disc-AF(1)

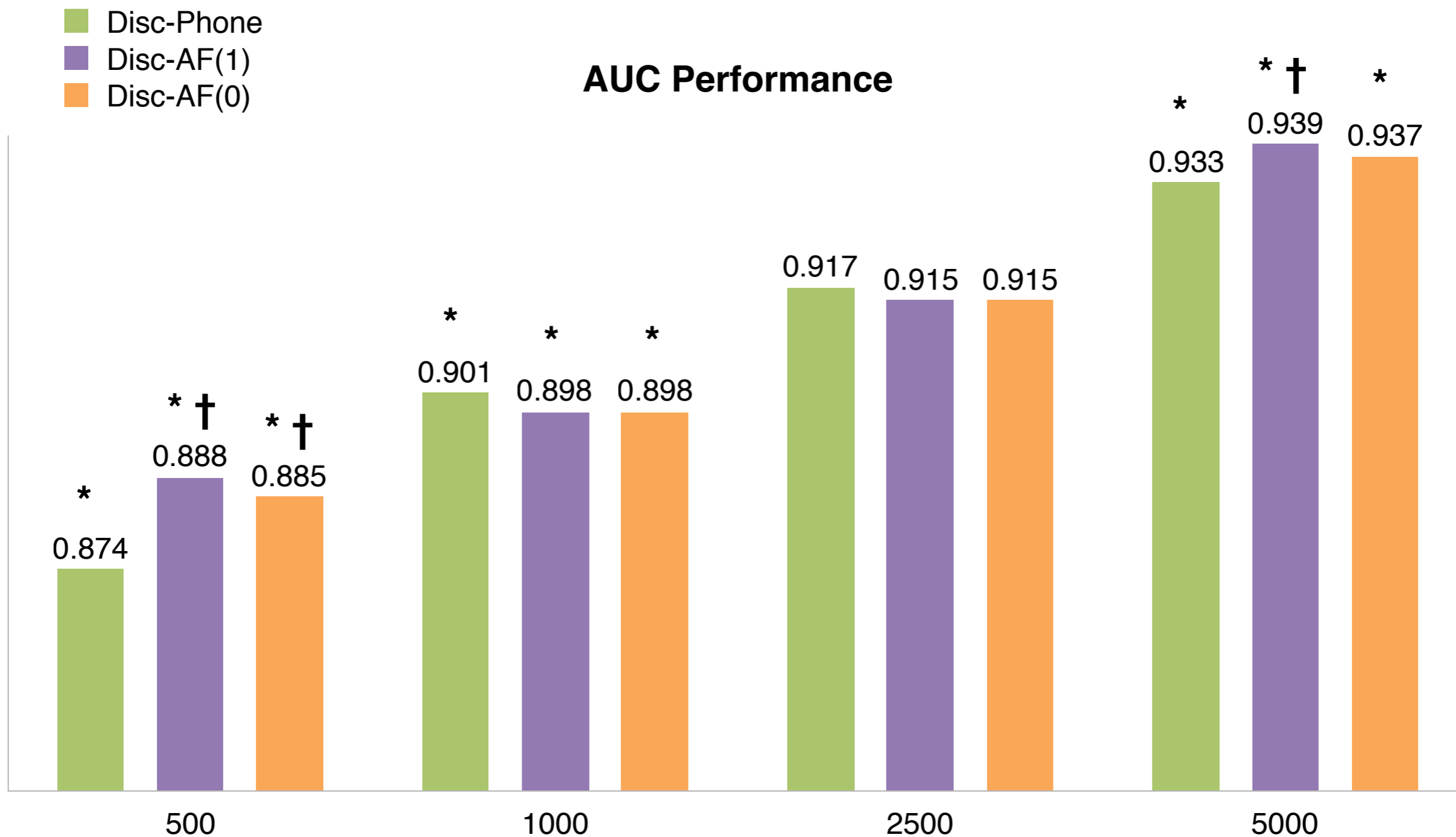
AUC performance



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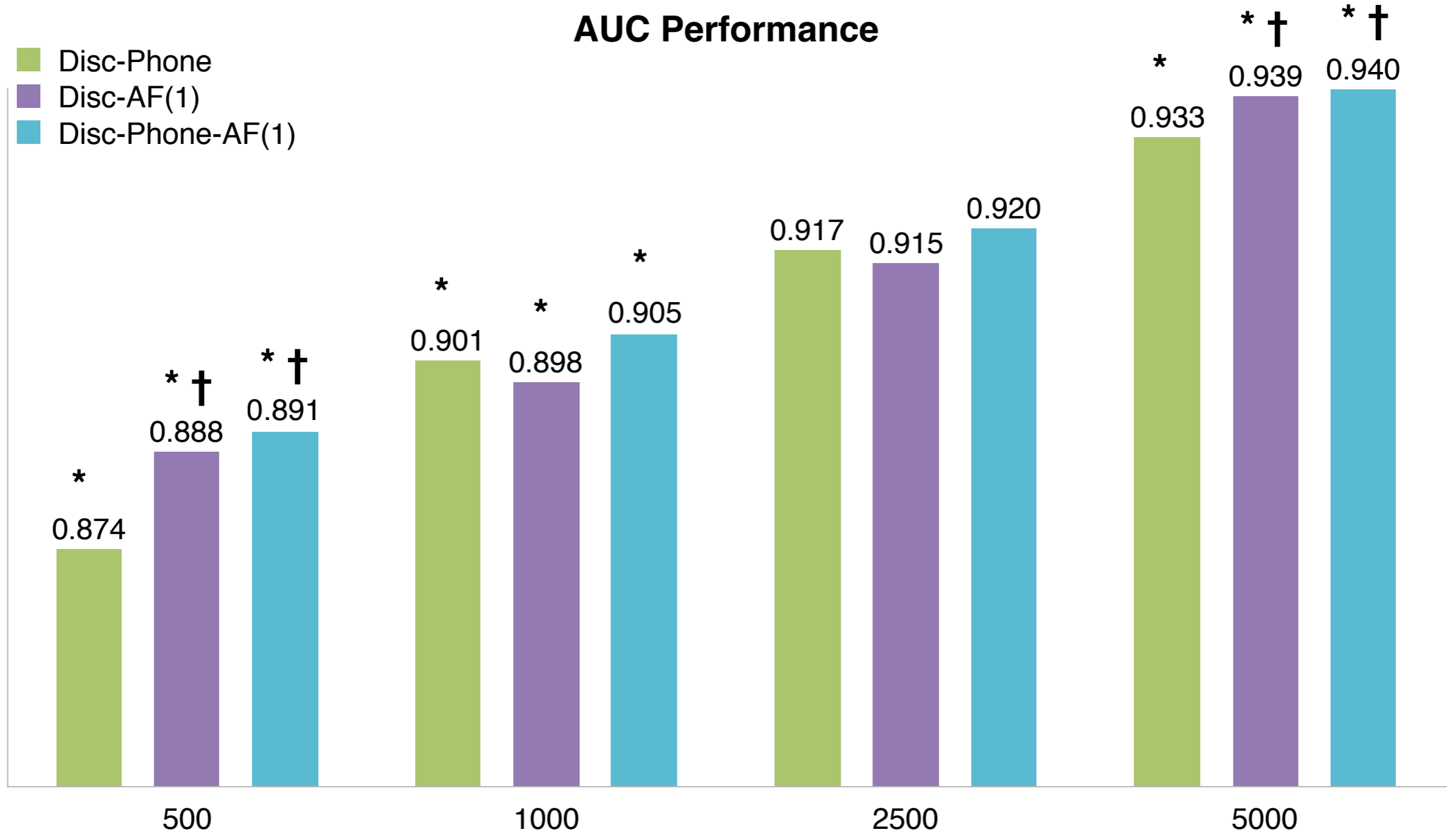
Effect of Asynchrony



* : significant ($p \leq 0.05$) difference over HMM-tri

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Combining Phone, AF Models



* : significant ($p \leq 0.05$) difference over HMM-tri

† : significant ($p \leq 0.05$) difference over Disc-

Phone

Conclusions

- Discriminative systems outperform the HMM systems by large margins
- AF-based system outperform phone-based systems in very-low-resource conditions
 - System appears to hypothesize greater asynchrony for words with pronunciation variation
- In current work, we are exploring techniques for optimizing ATWV instead

Acknowledgement



articulatory
phonology
modeling



discriminative
keyword
spotting

Thanks!