

## Introduction

### Speech separation (Cocktail-party problem)

- ◆ **Goal:**  
Segregating each stream of sound from mixed speech of many speakers.
- ◆ **Application:**
  - Robust speech recognition: preprocessing noisy or multi-speaker speech data
  - Improve speech quality: boosting signal noise ratio for targeted speech

## Current approaches

- ◆ Non-negative matrix factorization (NMF)
  - Model non-negative data using parts-based, additive representations
  - Exploit speaker-specific parts to separate mixed speech
- ◆ Sparse Non-Negative Matrix Factorization (SNMF)
  - Extend NMF by sparsely combining parts
  - Estimate over-complete dictionaries
- ◆ **Limitations**
  - Learn parts **independently**
  - Does not **adapt** to other speakers' interference

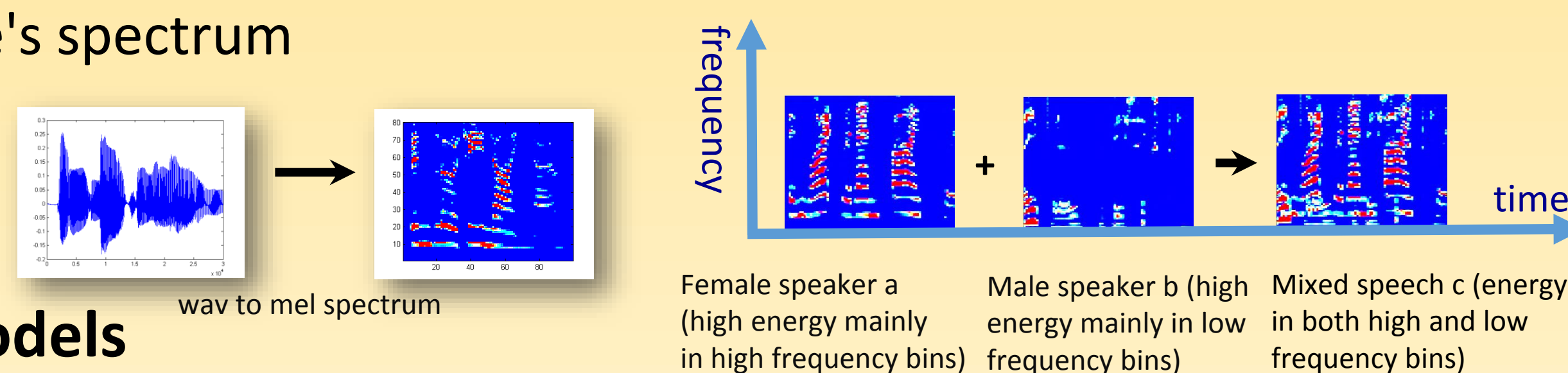
## Our approaches

- **Main Idea**
  - ◆ Discriminative non-negative matrix factorization (DNMF)
    - Learn parts **jointly** for all speakers
    - Optimize parts to be maximally effective in segregating from other speakers
  - ◆ Pairwise DNMF
    - Extend DNMF by distinguishing only pairwise speakers
    - Reduce computational cost

## Nonnegative matrix factorization

### ◆ Intuitions

- Represent speech signals with nonnegative magnitudes of their mel spectrum
- Model mixed signal's spectrum as additive sum of each individual source's spectrum



### ◆ Models

- Let  $D_{i1}, D_{i2}, \dots, D_{iK}$  denote speaker  $i$ 's speech prototypes (e.g., one for each phoneme's spectrum),  $S_i$  denotes the input signal's spectrum of speaker  $i$
- Minimize the difference between input signals and linear combinations of those prototypes for each speaker

$$F(D, H) = \sum_i KL(S_i \parallel D_i H_i)$$

### ◆ Learning **How to learn prototypes D without knowing h?**

- Iteratively learn D and H for each speaker
- Update rules\*

$$H_i \leftarrow H_i \cdot \frac{D_i^T S_i / D_i H_i}{\sum D_i} \quad D_i \leftarrow D_i \cdot \frac{S_i H_i^T / D_i H_i}{\sum H_i}$$

## Discriminative NMF

### ◆ Intuitions

- Reconstructed speech from clean conditions should also be optimal under other interfering conditions
- Learning jointly all prototypes and consider the sparsity of H

### ◆ Models

- Let  $S_{ij}$  denotes the mixed signal's spectrum of speaker  $i$  and  $j$
- Let  $\hat{S}_{ij}$  denotes the reconstruction of the mixed signal's spectrum

$$F(D, H) = \sum_i KL(S_i \parallel D_i H_i) + \sum_{i,j} KL(S_{ij} \parallel \hat{S}_{ij}) + \lambda \sum H_i \quad \text{where } \hat{S}_{ij} = [D_i \ D_j] \times \begin{bmatrix} H_i \\ H_j \end{bmatrix}$$

### ◆ Pairwise speakers

- Limit the speakers involved during training
- Easily adapt to multiple speakers

### ◆ Optimization algorithm

- Optimize each speaker's prototypes alternatively

$$H_i \leftarrow H_i \cdot \frac{(D_i)^T \sum_j S_{ij} / \hat{S}_{ij}}{\sum D_i + \lambda}$$

$$D_i \leftarrow D_i \cdot \frac{\sum_j S_{ij} H_i^T + U(VH_i^T \cdot D_i) \cdot D_i}{U\left(\sum_j \frac{S_{ij} H_i^T}{\hat{S}_{ij}} \cdot D_i\right) \cdot D_i + VH_i^T}$$

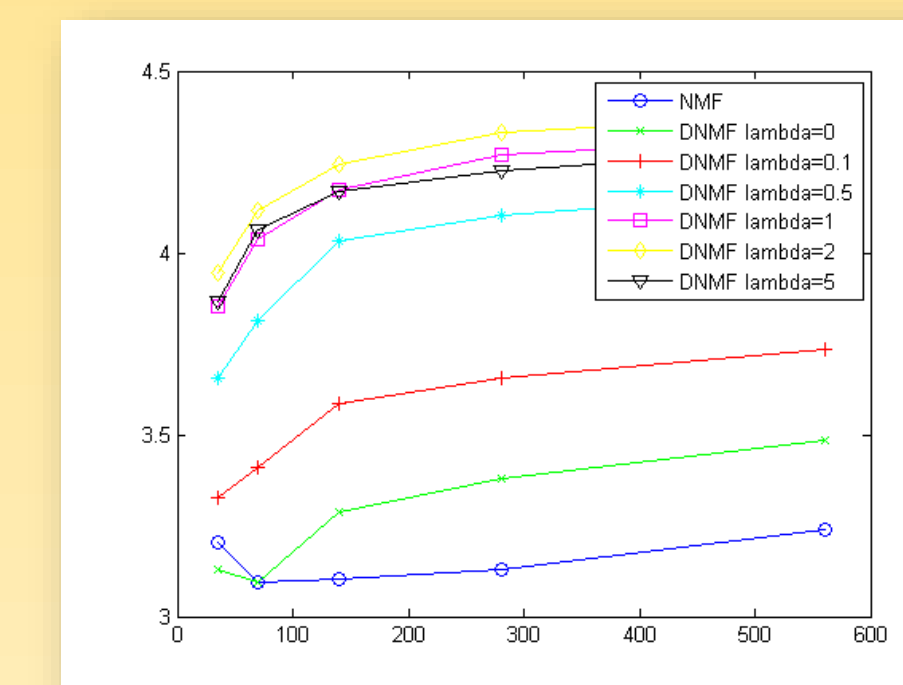
## Experiment setup

- ◆ The Grid Corpus
  - 34 speakers and 1000 sentences per speaker
  - half of the 1000 sentences for each speaker are used for training and the other half for evaluation
- ◆ Evaluation
  - tune parameters and validate on development set(half of the evaluation set)
  - evaluate the performance of the model on test set
  - Signal noise ration (SNR): the ratio of signal power from reconstructed speech to the residual signals after subtracting reconstructed speech
- ◆ Analyze the prototypes and reconstruction coefficients to gain further insight

## Results

### ◆ DNMF vs. NMF

component	$\lambda$	DNMF						
		0	0	0.1	0.5	1	2	5
35		3.20	3.13	3.33	3.66	3.85	3.95	3.87
70		3.10	3.10	3.41	3.81	4.04	4.12	4.06
140		3.10	3.29	3.59	4.03	4.17	4.24	4.17
280		3.13	3.38	3.66	4.10	4.27	4.33	4.23
560		3.24	3.48	3.74	4.17	4.32	4.38	4.29



- Outperform NMF in improving SNR

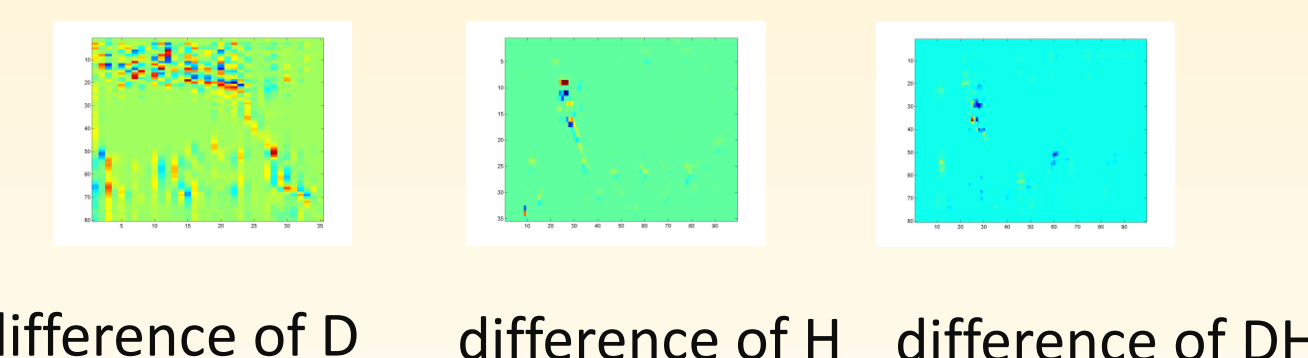
### ◆ DNMF vs. SNMF

- No significant improvement

### ◆ Why similar results?

- the dictionary D from training and the activity H from testing are different
- the reconstruction, DH, are similar.

component	DNMF				SNMF		
	best lambda	best beta	snr	best lambda	best beta	snr	
35	2	1	3.94	2	1	4.03	
70	2	1	4.11	2	1	4.13	
140	2	1	4.25	5	2	4.26	
280	2	1	4.33	5	2	4.31	
560	2	1	4.37	2	1	4.37	



- ◆ Pairwise DNMF vs. DNMF

- Compare one pair of speaker (different gender)
- Slightly better than DNMF, need further investigation

component	35		140		560	
	pair	dnmf	pair	dnmf	pair	dnmf
SNR	5.57	5.42	5.72	5.70	5.96	5.97

## Conclusion

- ◆ We have developed a new method for speech separation. The key idea is to learn speaker-specific parts discriminatively.
- ◆ Our method yields promising results, improving the popular approach NMF.
- ◆ Our method is applicable to other problems where NMF is used.

## Selected References

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- [3] Eggert J, Korner E. Sparse coding and NMF[C]//Neural Networks, 2004. Proceedings. 2004 IEEE International Joint Conference on. IEEE, 2004, 4: 2529-2533.
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