Blind one-microphone speech separation: A spectral learning approach

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Summary

- Discriminative approach to blind one-microphone separation
- Reformulation as spectrogram segmentation
- Learning from artificially mixed data
- Machine learning algorithm for
 - segmenting
 - learning how to segment from training data

Blind one-microphone speech separation

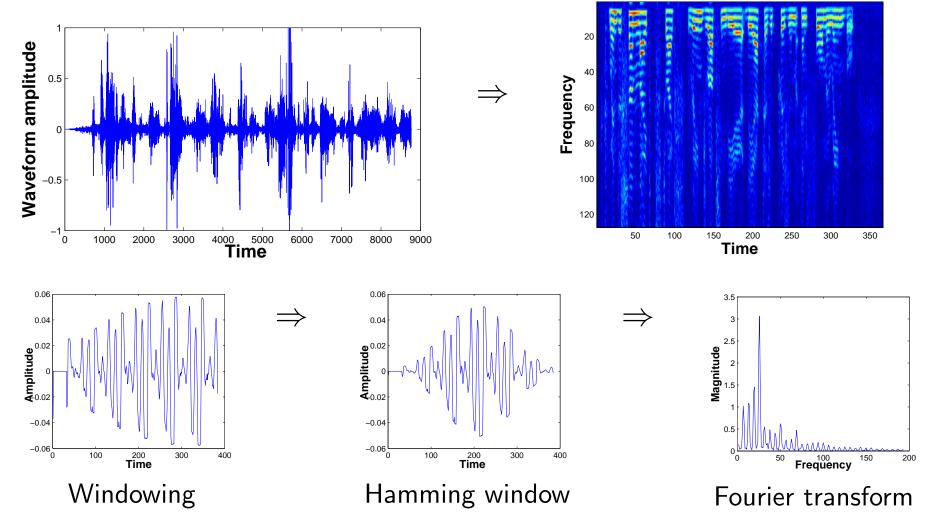
- Two or more speakers s_1, \ldots, s_m one microphone x
- Ideal acoustics $x = s_1 + s_2 + \cdots + s_m$
- Goal: recover s_1, \ldots, s_m from x
- **Blind**: without knowing the speakers in advance
- Two types of approaches

- Generative

- \ast Learn source model p(s) ... then "simply" an inference problem
- * Model too simple : does not separate
- * Model too complex : inference intractable
- * Works for non blind situations (Roweis, 2001, Lee et al., 2002)
- Discriminative: model of separation task, not of speakers

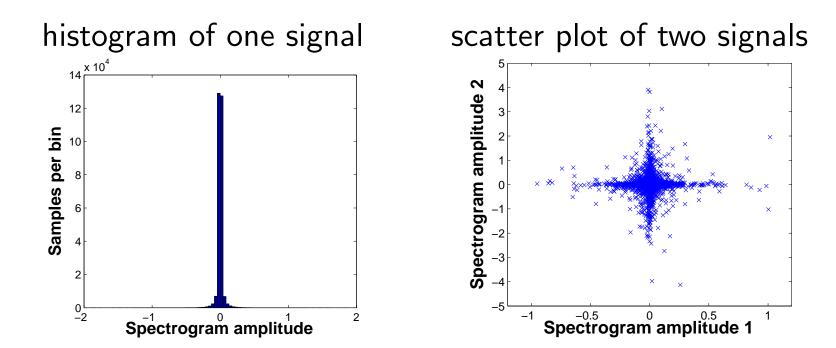
Spectrogram

- Spectrogram (a.k.a Gabor analysis, Windowed Fourier transforms)
 - cut the signals in overlapping frames
 - apply a window and compute the FFT

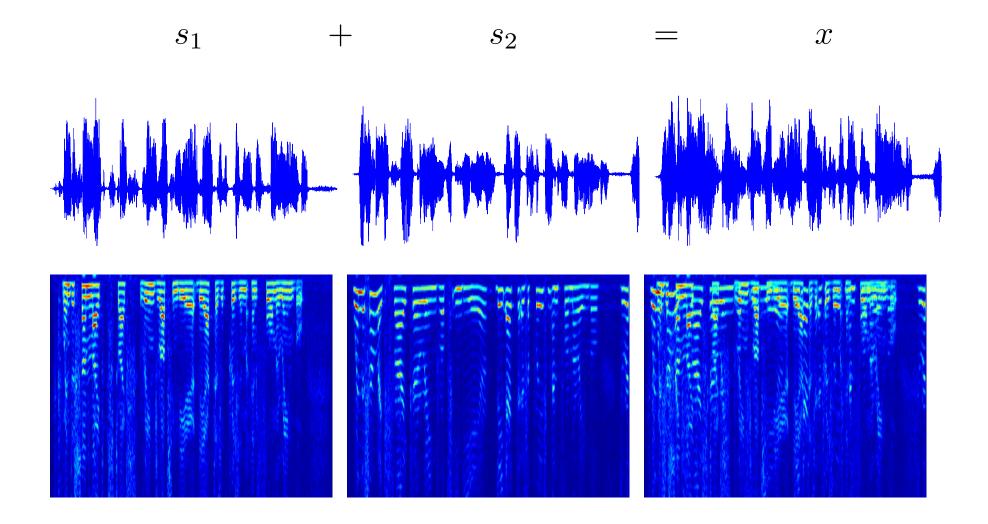


Sparsity of speech signals - spectrogram

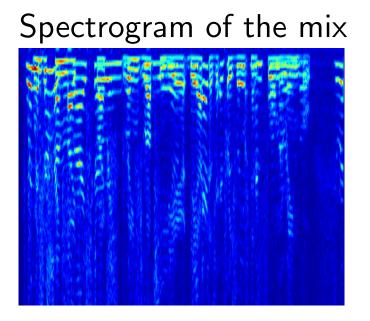
- Disjoint support of spectrograms observed by several researchers (Cooke, 1994, Roweis, 2000, Yilmaz and Rickard, 2004)
- Sparsity of the spectrogram (all pixels taken together)

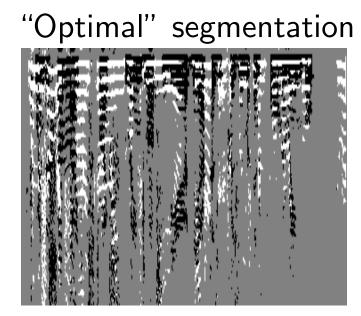


Sparsity and superposition



Building training set





- Empirical property: there exists a segmentation that leads to audibly acceptable signals (e.g., take $\arg\max(|S_1|,|S_2|)$)
- Work as possibly large training datasets
- Requires new way of segmenting images ...
- ... which can be learned from data

Summary of spectral clustering

```
Data: P elements x_p \in \mathcal{X}, p = 1, \ldots, P
        \downarrow
   Step 1: build "affinity/similarity" matrix W \in \mathbb{R}^{P \times P}
        \downarrow
   Step 2: normalize the affinity matrix: W = D^{-1/2}WD^{-1/2} where
D is diagonal with sums of rows of W
         \downarrow
   Step 3: compute the R largest eigenvectors U(W) \in \mathbb{R}^{P \times R} of \widetilde{W}
         \downarrow
   Step 4: considering U(W) as P points in \mathbb{R}^R, cluster U using
weighted K-means
```

Output: partition E

 \downarrow

Learning problem

• Input:

- spectrograms of mixed signals
- "optimal" segmentations

• Output:

- features for each spectrogram
- Parameterized similarity matrix for spectral clustering

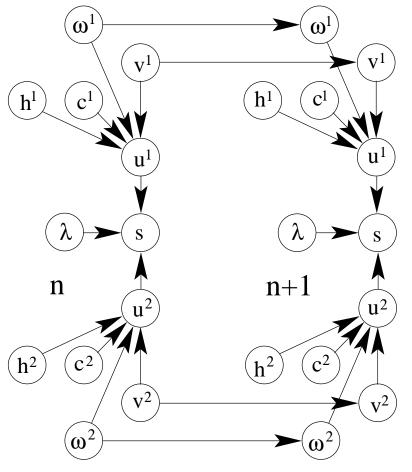
• Challenges:

- Requires complex features
- Large dimensionality of the spectrogram

Features for speech separation

- Classical cues from speech psychophysics
- Non-harmonic cues (similar to vision cues):
 - Continuity
 - Common fate cues
- Harmonic cues (requires different type of affinity matrices):
 - Pitch and potentially timbre
 - Requires multiple pitch estimation

Multiple pitch extraction

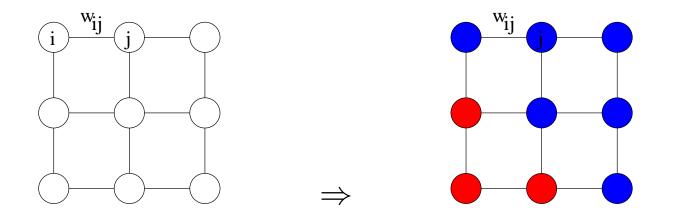


- $\omega:$ pitch frequency
- v: voicing decision
- h: spectral envelope
- c: constant unvoiced amplitude

- Additive model for the magnitude of the spectrogram
- Factorial HMM
- Smoothness prior on the spectral envelope
- Discriminative training
- Determination of number of speakers

Spectral graph partitioning

• *P* vertices of a weighted graph to partition into disjoint clusters



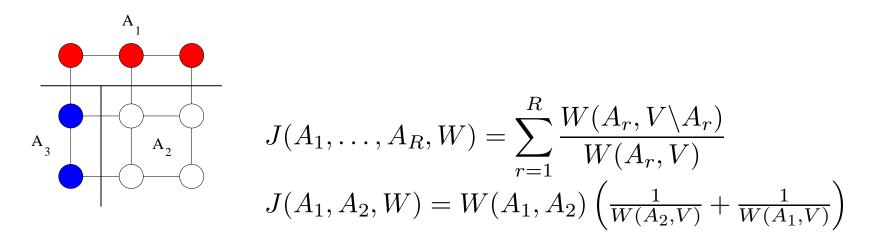
- Affinity matrix $W \in \mathbb{R}^{P \times P}$ ($W_{pp'}$ is large when points p and p' are likely to be in the same cluster)
- **Goal**: find clusters with high intra-similarity and low inter-similarity

Normalized cuts

• Weight between two sets of vertices A and B, defined as:

$$W(A,B) = \sum_{i \in A, j \in B} W_{ij}$$

(multi-way) normalized cut for partition V = A₁ ∪ · · · ∪ A_R (Shi and Malik, 2000, Zha et al, 2001):



• Goal: minimize normalized cut

Learning spectral clustering

- Learning from fully segmented images (Bach & Jordan, NIPS 2004)
- Single cost function J(W, E)
 - Minimize with respect to the partition $E \Rightarrow$ spectral clustering
 - Minimize with respect to the matrix $W \Rightarrow$ learning similarities
- Uses the power method to approximate eigenvectors
- Requires parameterized affinity matrices

Very large similarity matrices

• Three different time scales $\Rightarrow W = \alpha_1 W_1 + \alpha_2 W_2 + \alpha_3 W_3$

• Small

- Fine scale structure (continuity, harmonicity)
- very sparse approximation

Medium

- Medium scale structure (common fate cues)
- band-diagonal approximation, potentially reduced rank

• Large

- Global structure (e.g., speaker identification)
- low-rank approximation (rank is independent of duration)

Parameterized affinity matrices

• Non pitch-related features f_a , $a = 1, \ldots, P$.

$$W_{ab} = \exp(-||f_a - f_b||^\beta)$$

• Pitch related features

– feature
$$f_a$$
, $a = 1, \ldots, P$

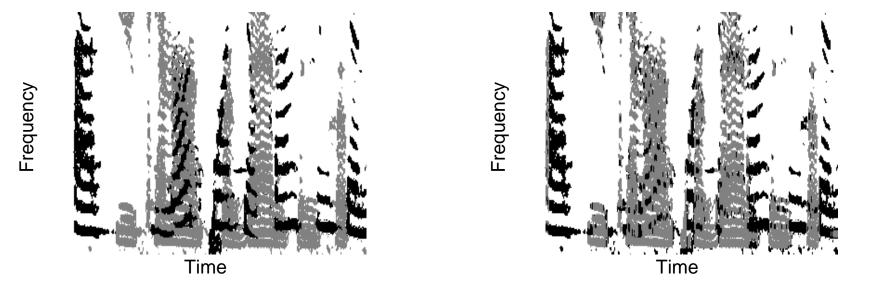
- strength of pitch y_a :

$$W_{ab} = \exp(-|g(y_a, y_b) + \beta_3|^{\beta_4} ||f_a - f_b||^{\beta_2})$$

where $g(u, v) = (ue^{\beta_5 u} + ve^{\beta_5 v})/(e^{\beta_5 u} + e^{\beta_5 v})$ ranges from the minimum of u and v for $\beta_5 = -\infty$ to their maximum for $\beta_5 = +\infty$.

Experiments

- Two datasets of speakers: one for testing, one for training
- Left: optimal segmentation right: blind segmentation



- Testing time (linear in duration of signal): currently 30 minutes for 4 seconds of speech
- Speech samples on web site

Current work

- Mixing conditions: allow some form of delay or echo
- speaker vs. speaker \Rightarrow speaker vs. non stationary noise
- Post processing of spectrogram segmentation
- Time and memory requirements