

Spectral Clustering for one mic Audio Blind Separation

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December 18, 2006

Problem

Audio Blind Separation:

- Original mixed audio out \longrightarrow Audio signals s_i
- Restrictions s_i :
 - 1 $\sum_i^n s_i$ perceived similarly to out
 - 2 s_i $i = 1..n$ should mean something to a human
(examples: tracks, instruments, auditory streams, physical sources, notes, chords, noises...)

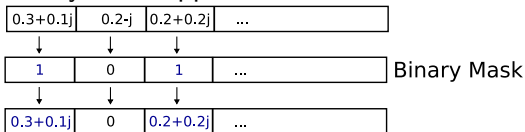
Extraction of the audio signals

Time Frequency Masking

1 Signal splitted into overlapped frames of fixed size in time.

2 FFT

3 Binary mask applied



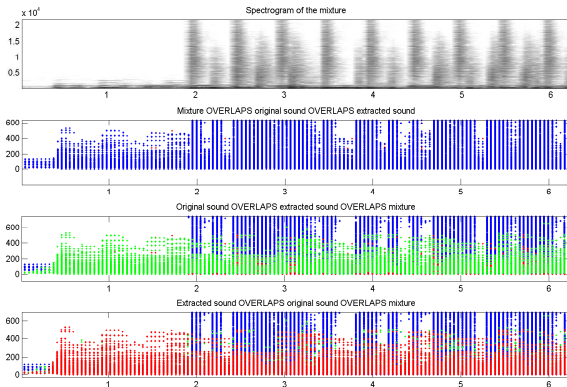
4 IFFT

5 Overlap-and-add process.

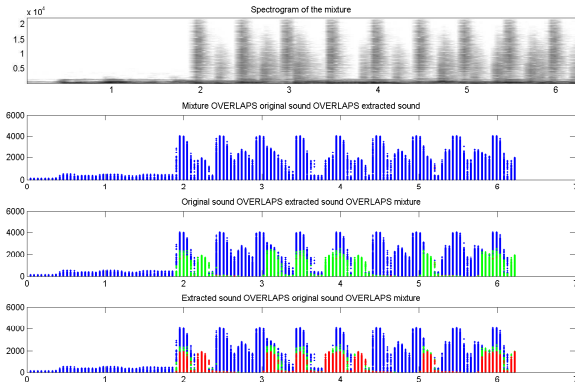
Data

- Mixture and sound track waveforms available.
'mix.wav' = 'guitar.wav' + 'kick.wav' + 'snare.wav' + 'hh.wav'
- We know that it's possible to extract each of them.
We know how to generate ideal binary masks if the target sound is available.

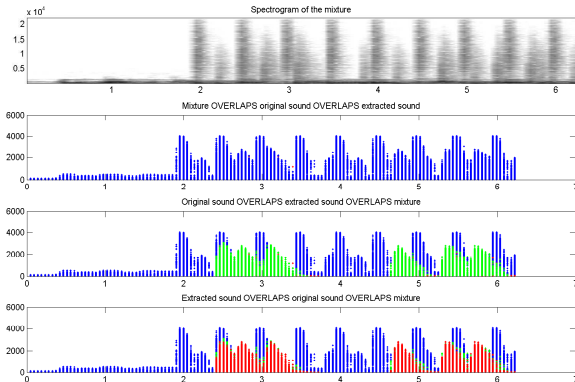
Example: ideal binary mask to extract 'guitar.wav'



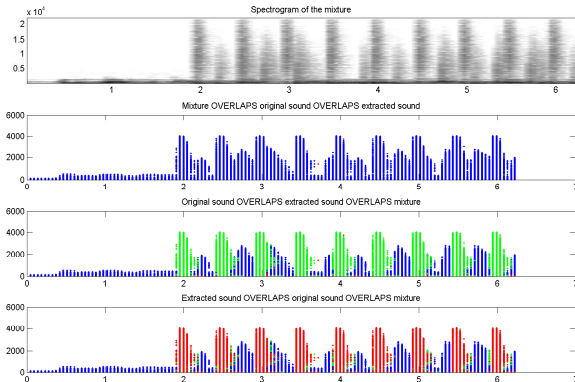
Example: ideal binary mask to extract 'kick.wav'



Example: ideal binary mask to extract 'snare.wav'



Example: ideal binary mask to extract 'hh.wav'



Machine learning to cluster the time-frequency points

Learning the binary mask...

- Clusters are not disjoint. We focus on extracting one single audio signal each time.
- SVM or Spectral Clustering? Spectral Clustering seem to be more appropriate when there are intersections.

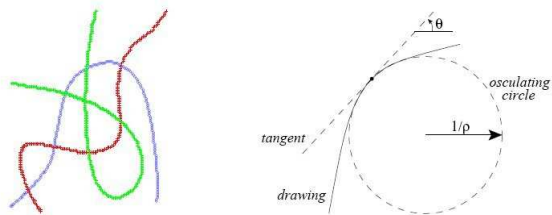


Figure: Labeled hand drawings by spectral clustering. Francis R.Bach, Michael I.Jordan 06.

Spectral Clustering

- Let $A = (A_r)_{r \in 1 \dots R}$ be the R disjoint clusters of the points such that $\bigcup_r A_r = \{p_1, p_2, \dots, p_N\} = V$ which the algorithm should output.

Let $W(A, B) = \sum_{i \in A} \sum_{j \in B} W_{ij}$ the total weight between the sets of points A and B.

Let a similarity matrix W .

Finally let D be a diagonal matrix whose i -th diagonal element is the sum of the elements in the i -th row of W .

- We want to minimize the R -way normalized cut:

$$C((A_r)_{r \in (1 \dots R)}, W) = \sum_{r=1}^R \frac{W(A_r, V \setminus A_r)}{W(A_r, V)}$$

- Algorithm that solves it by computing the eigenvectors of $D^{-1/2} W D^{-1/2}$ and performing a weighted Kmeans clustering of them.

Spectral Clustering applied to audio

- W is huge! Solutions:
 - Analyze the audio in short frames.
 - Approximate W by a sparse matrix. "low-band rank decomposition" suggested by Francis R.Bach, Michael I.Jordan 06. Numerical methods that take advantage of it to find the eigenvectors of $D^{-1/2}WD^{-1/2}$.
- How we compute the distance between two points?
 - Use features that are related to how we group sounds. "Auditory Scene Analysis" by Bregman.
 - Automatically learn the weight of each feature. Francis R.Bach, Michael I.Jordan 06.

Simulations

Simplified implementation:

- We adapt spectral clustering used for image processing. L. Zelnik-Manor and P. Perona 04.
- We use a sparse W similarity matrix which sets a neighbourhood of 7×7 nonzero time-frequency points.
- We analyse a very limited amount frames.

Poor results:

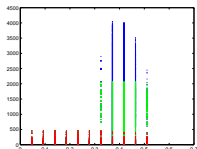


Figure: Output of our algorithm: spectral clustering of the time-frequency points (green). Blue points are the mixture points, and red points are guitar

Conclusion

Bad results but there's still room for improvement:

- More emphasis on finding a good similarity matrix, by introducing psychoacoustic features like pitch, common fate (onset, offset, frequency comodulation).
- Learn automatically their weight to fit the training data.

Main references

- Title: Learning Spectral Clustering, With Application to Speech Separation
Authors: Francis R. Bach, Michael I. Jordan
Year: 2006
- Title: Self-Tuning Spectral Clustering
Authors: L. Zelnik-Manor and P. Perona
Year: 2004