ICASSP 2018

AASP-L2: Multi-microphone Speech Enhancement and Source Separation Tuesday, April 17, 16:00 - 16:20

#### Joint Separation and Dereverberation of Reverberant Mixtures with Determined Multichannel Non-negative Matrix Factorization

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# Problem setting

Aim: Blind source separation (BSS) under highly reverberant environments

Assumptions:

•# of sources = # of mics

Sources do not move



## Frequency-wise instantaneous mixture

Anechoic mixture can be approximated as frequency-wise instantaneous mixture



 BSS problem involves frequency-wise source separation and permutation alignment across frequencies

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 BSS problem involves frequency-wise source separation and permutation alignment across frequencies

#### Independent Vector Analysis (IVA) [Kim+2006, Hiroe2006]

 Simultaneously solves frequency-wise source separation and permutation alignment



Finds separation matrices such that

- the independence of separated signals is maximized, and
- the power of each separated signal varies coherently across frequencies

#### Multichannel non-negative matrix factorization (MNMF)

[Ozerov+2010, Sawada+2012]

Multichannel extension of non-negative matrix factorization
 The power spectrogram of each source is modeled as a product of two non-negative matrices



#### Independent Low-Rank Matrix Analysis (ILRMA)

[Kameoka+2010, Kitamura+2016]

Idea combining IVA and MNMF

MNMF framework specialized for determined systems



#### Motivation of this work

- All BSS systems using frequency-wise instantaneous mixture model are weak against long reverberation
- To make ILRMA robust against long reverberation, we employ frequency-wise deconvolution system [Nakatani+2008, Yoshioka+2011, Kameoka+2010, ...] as the mixing model

Instantaneous: 
$$\mathbf{W}^{\mathsf{H}}(f)\mathbf{x}(f,n) = \mathbf{s}(f,n)$$

Local Gaussian source model

```
s_j(f,n) \sim \mathcal{N}_{\mathbb{C}}(s_j(f,n)|0, v_j(f,n)) \qquad (j=1,\ldots,J)
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Local Gaussian source model

$$s_j(f,n) \sim \mathcal{N}_{\mathbb{C}}(s_j(f,n)|0, \underline{v_j(f,n)}) \qquad (j=1,\ldots,J)$$

$$\underline{v_j(f,n)} = \sum_k h_k(f)u_k(n) \longrightarrow \mathsf{NMF} \mathsf{model}$$
  
Low-rank matrix



Local Gaussian source model

$$s_j(f,n) \sim \mathcal{N}_{\mathbb{C}}(s_j(f,n)|0, \underline{v_j(f,n)}) \qquad (j = 1, \dots, J)$$
$$\underline{v_j(f,n)} = \sum_k h_k(f)u_k(n) \longrightarrow \mathsf{NMF} \mathsf{model}$$

• Mixing model  $\mathbf{y}(f,n) = \mathbf{x}(f,n) - \sum_{n'=1}^{N'} \mathbf{G}^{\mathsf{H}}(f,n')\mathbf{x}(f,n-n')$   $\mathbf{s}(f,n) = \mathbf{W}^{\mathsf{H}}(f,0)\mathbf{y}(f,n)$ 

Local Gaussian source model

$$s_j(f,n) \sim \mathcal{N}_{\mathbb{C}}(s_j(f,n)|0, \underline{v_j(f,n)}) \qquad (j = 1, \dots, J)$$
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Mixing model  $\mathbf{y}(f,n) = \mathbf{x}(f,n) - \sum_{n'=1}^{\infty} \mathbf{G}^{\mathsf{H}}(f,n')\mathbf{x}(f,n-n')$  $\mathbf{s}(f,n) = \mathbf{W}^{\mathsf{H}}(f,0)\mathbf{y}(f,n)$ Log-likelihood  $L(\boldsymbol{\theta}) = 2N \sum_{f} \log |\det \mathbf{W}^{\mathsf{H}}(f, 0)| - \sum_{f, n, j} \left( \log v_j(f, n) + \frac{|s_j(f, n)|^2}{v_j(f, n)} \right)$ 

#### Optimization algorithm

Log-likelihood

$$\begin{split} L(\boldsymbol{\theta}) &= \\ 2N \sum_{f} \log \left| \det \mathbf{W}^{\mathsf{H}}(f, 0) \right| - \sum_{f, n, j} \left( \log v_{j}(f, n) + \frac{|s_{j}(f, n)|^{2}}{v_{j}(f, n)} \right) \\ \text{where} \left\{ \begin{array}{l} \mathbf{y}(f, n) &= \mathbf{x}(f, n) - \sum_{n'} \mathbf{G}^{\mathsf{H}}(f, n') \mathbf{x}(f, n - n') \\ \mathbf{s}(f, n) &= \mathbf{W}^{\mathsf{H}}(f, 0) \mathbf{y}(f, n) \end{array} \right. \end{split}$$

• Optimization process (S1)  $\theta_G \leftarrow \operatorname{argmax} L(\theta)$  : Dereverberation filter (S2)  $\theta_W \leftarrow \operatorname{argmax} L(\theta)$  : Separation matrix  $\theta_W$ (S3)  $\theta_V \leftarrow \operatorname{argmax} L(\theta)$  : NMF parameters  $\theta_V$ 

## (S1) Dereverberation filter update

• When  $\theta_W = { \mathbf{W}^{\mathsf{H}}(f, 0) }_f$  is fixed,  $L(\boldsymbol{\theta})$  becomes equal to the objective function of a multivariate linear prediction problem when seen as a function of  $\theta_G = { \mathbf{G}^{\mathsf{H}}(f, 1), \dots, \mathbf{G}^{\mathsf{H}}(f, N') }_f$ 

• Thus, the optimal  $\theta_G$  that minimizes  $L(\theta)$  can be found by solving a Yule-Walker equation

## (S2, S3) Updates of remaining parameters

- When  $\theta_G$  is fixed (and so the dereverberated signals  $\mathbf{y}(f, n)$  can be treated as observed signals),  $L(\theta)$  becomes equal to the log-likelihood of ILRMA
- Thus, we can use the same optimization scheme as ILRMA:
  (S2) Separation matrix update

with Iterative Projection (IP) [Ono2011]

- $L(\theta)$  can be maximized analytically with respect to one of the column vectors of  $\mathbf{W}^{\mathsf{H}}(f, 0)$
- We can iteratively maximize  $L(\theta)$  with respect to each column
- (S3) NMF parameter update with majorization-minimization [Kameoka+2006, Nakano+2010, Févotte2011]
  - L(θ) is equal to the objective function of Itakura-Saito divergence NMF up to constant terms when seen as a function of the NMF parameters
     14/18

#### Experimental settings

- Synthesized 10 mixtures for each gender pair of speech utterances excerpted from ATR speech database
- Used two-input four-output impulse response, which was measured in a varechoic chamber
- The reverberation time was 0.6 sec.
- Comparison :
  - Proposed (IP/FICA)
  - ILRMA, Sequential (Dereverberation +ILRMA)
- STFT : 32ms Hanning window, 8ms overlap
- Filter length N' for dereverberation

Frequency	0~0.8kHz	0.8~1.5kHz	1.5 <b>~</b> 3.0kHz	3.0kHz~
Filter length $N'$	25	20	15	10

#### Evaluation measures :

- DRR (Direct-to-reverberation ratio)
- SIR (Signal-to-Interference ratio)



### Simulation results (1/2)

#### [Direct-to-reverberation ratio]



#### Simulation results (2/2)

#### [Signal-to-Interference ratio]



#### Computational time comparison

Average computation times normalized to 1 with the reference method (ILRMA)

	Proposed (IP)	Proposed (FICA)	ILRMA
Comp. time (normalized)	2.56	2.80	1.0

**BSS** under highly reverberant environments

ILRMA + Frequency-wise deconvolution system

$$\sum_{n'=0}^{N'} \mathbf{W}^{\mathsf{H}}(f, n') \mathbf{x}(f, n - n') = \mathbf{s}(f, n)$$

 The optimization process consists of iteratively optimizing dereverberation filters, separation matrix and NMF parameters

The proposed method yielded higher separation performance and dereverberation performance