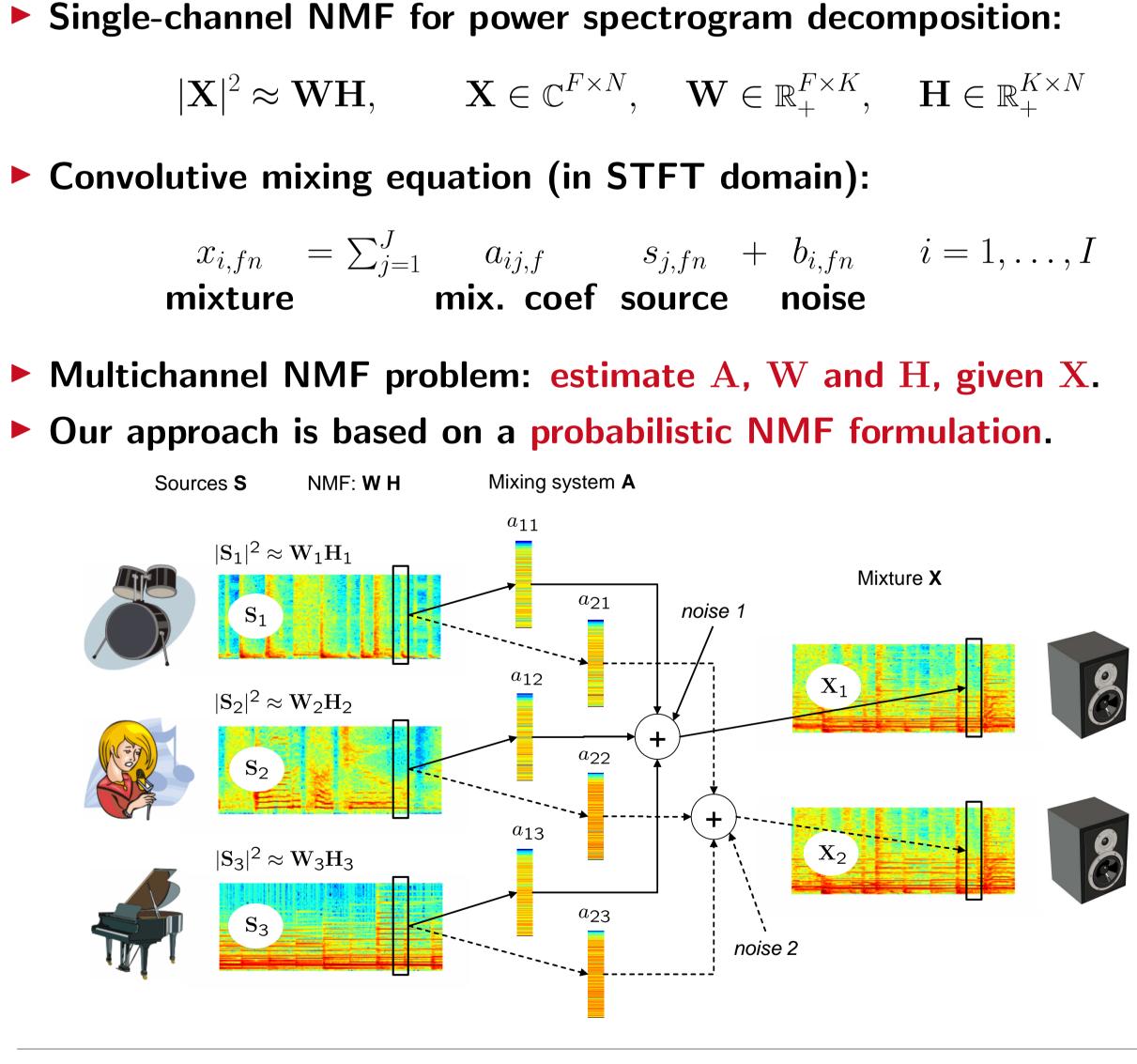


²CNRS LTCI, Télécom ParisTech, Paris, France ¹Institut Télécom, Télécom ParisTech, CNRS LTCI, Paris, France e-mail: {ozerov,fevotte}@telecom-paristech.fr web: http://perso.telecom-paristech.fr/~ozerov/

- Nonnegative Matrix Factorization (NMF) is usually used for singlechannel audio signal power spectrogram decomposition.
- We propose a multichannel NMF framework in a general case of convolutive mixtures of sources.
- Possible applications: source separation (blind or supervised) and information retrieval from audio (e.g., music transcription).
- Here we apply multichannel NMF to different stereo audio source separation tasks, and obtain very promising results.

Introduction



Probabilistic NMF Model

Source STFT is modeled as a sum of latent Gaussian components:

 $s_{j,fn} = \sum_{k=1}^{K_j} c_{j,k,fn}$ with $c_{j,k,fn} \sim \mathcal{N}_c(0, w_{j,fk}h_{j,kn})$

• Maximum Likelihood (ML) estimation of $\mathbf{W}_i = [w_{i,fk}]_{f,k}$ and $\mathbf{H}_i = [w_{i,fk}]_{f,k}$ $[h_{j,kn}]_{k,n}$ given source STFT $S_j = [s_{j,fn}]_{f,n}$ is equivalent to NMF decomposition $|\mathbf{S}_i|^2 \approx \mathbf{W}_i \mathbf{H}_i$ with Itakura-Saito (IS) divergence [1].

Multichannel Nonnegative Matrix Factorization in Convolutive Mixtures With Application to Blind Audio Source Separation

Alexey Ozerov¹ and Cédric Févotte²

Abstract



Proposed Multichannel NMF Methods

Exact Likelihood Maximization with EM Algorithm

- Criterion ($\theta = \{A, W, H\}$, $x_{fn} = [x_{1, fn}, \dots, x_{I, fn}]^T$): $C_1(\boldsymbol{\theta}) = -\log p(\mathbf{X}|\boldsymbol{\theta}) = -\sum_{fn} \log p(\mathbf{x}_{fn}|\boldsymbol{\theta}).$
- **Expectation-Maximization (EM) Algorithm (** $\mathbf{C} = [c_{j,k,fn}]_{j,k,f,n}$ **)**:

E step:	$Q(\boldsymbol{\theta} \boldsymbol{\theta}^{(l)}) = \int \log$	$p(\mathbf{X}, \mathbf{C}$
M step:	$oldsymbol{ heta}^{(l+1)}$ =	arg m

- $\blacktriangleright \mathbf{A}^{(l+1)} \approx \mathbf{A}^{(l)}$ for small noise, thus a "simulated annealing" strategy is used.
- Related to other model-based methods (e.g., GMM-based [2]).

Individual Likelihoods Maximization with MU rules

• Criterion (IS divergence $d_{IS}(x|y) = \frac{x}{u} - \log \frac{x}{u} - 1$):

 $C_2(\boldsymbol{\theta}) = -\sum_{i=1}^{I} \log p(\mathbf{X}_i | \boldsymbol{\theta}) = -\sum_{i, fn} \log p(x_{i, fn} | \boldsymbol{\theta})$ $(\neq \log p(\mathbf{X}|\boldsymbol{\theta}))$ $= \sum_{i,fn} d_{IS} \left(|x_{i,fn}|^2 \left| \sum_{j} |a_{ij,f}|^2 \sum_{k=1}^{K_j} w_{j,fk} h_{j,kn} \right).$

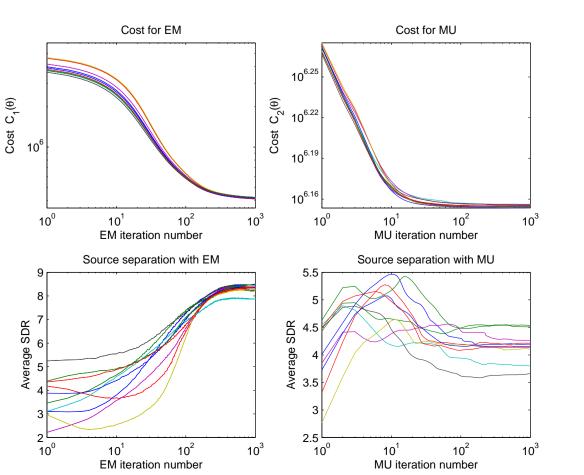
Multiplicative Update (MU) Rules:

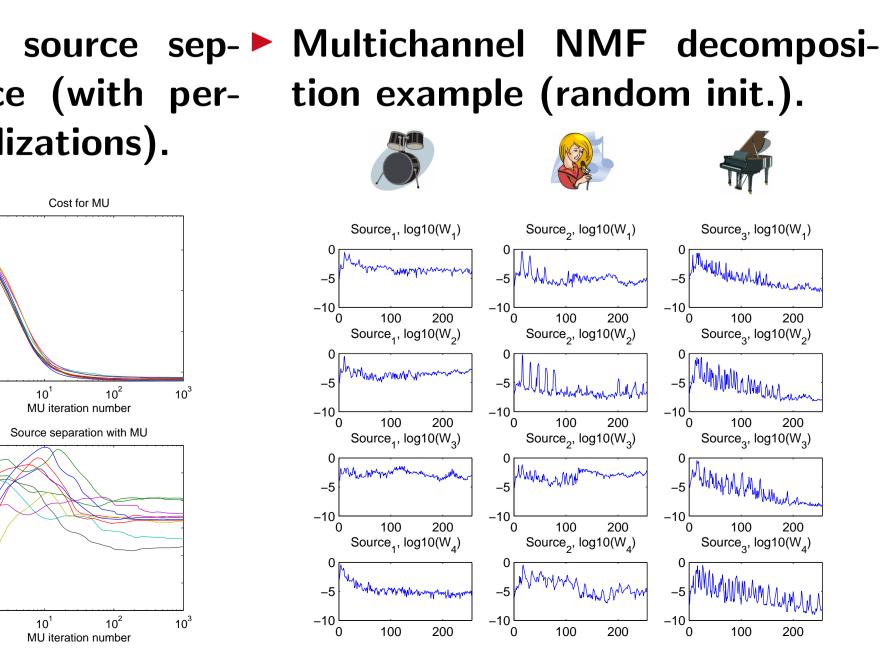
 $\theta_r \leftarrow \theta_r [\nabla_{\theta_r} C_2(\boldsymbol{\theta})]_- / [\nabla_{\theta_r} C_2(\boldsymbol{\theta})]_+,$

- $\nabla_{\theta_r} C_2(\boldsymbol{\theta}) = [\nabla_{\theta_r} C_2(\boldsymbol{\theta})]_+ [\nabla_{\theta_r} C_2(\boldsymbol{\theta})]_- \text{ and } [\nabla_{\theta_r} C_2(\boldsymbol{\theta})]_+, [\nabla_{\theta_r} C_2(\boldsymbol{\theta})]_- \geq 0.$ Related to Nonnegative Tensor Factorization (NTF).

Convergence and Decomposition

Convergence vs. aration performance (with perturbed oracle initializations).





 $\mathbf{C}|\boldsymbol{\theta}) \, p(\mathbf{C}|\mathbf{X}, \boldsymbol{\theta}^{(l)}) \, d\mathbf{C},$ $\max_{\boldsymbol{\theta}} Q(\boldsymbol{\theta}|\boldsymbol{\theta}^{(l)}).$

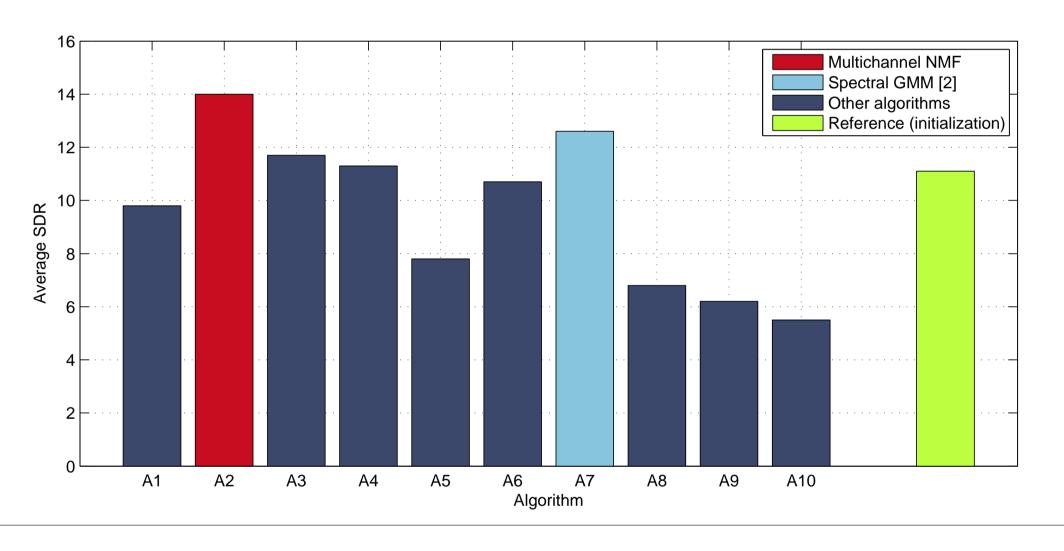
Source Separation Results with EM

Mixture type

Linear instantaneous Synthetic convolutive Live-recorded (convolutive)

Table 1: Separation in terms of average Signal to Distortion Ratio (SDR) (dB).





- Strong points of the proposed approach:

- Weak point: sensitive to the parameters initialization.
- separation," in ICA'09, 2009.



Results for different mixture types (initialization = baseline).

Music sources (3)		Speech sources (4)	
Baseline	Proposed	Baseline	Proposed
11.6	15.2	7.6	9.3
-0.8	-0.6	3.5	4.5
3.3	4.2	3.6	4.3

Signal Separation Evaluation Campaign (SiSEC 2008) / ICA 2009. "Under-determined speech and music (instantaneous) mixtures":

Conclusions

▷ use of both spectral and spatial diversities for source separation, joint and blind estimation of source and mixing models,

covers both underdetermined and (over)determined noisy cases, source model frees us from convolutive BSS permutation ambiguity, > computational load growing linearly with number of components.

References

[1] C. Févotte, N. Bertin, and J.-L. Durrieu, "Nonnegative matrix factorization with the Itakura-Saito divergence. With application to music analysis," Neural Computation, vol. 21, pp. 793–830, 2009. [2] S. Arberet, A. Ozerov, R. Gribonval, and F. Bimbot, "Blind spectral-GMM estimation for underdetermined instantaneous audio source