# Phase recovery with Bregman divergences for audio source separation

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Time-frequency separation = acts on the short-time Fourier transform (STFT).







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- **2.** Structured model, e.g., nonnegative matrix factorization, deep neural networks.







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**3.** Nonnegative masking and synthesis:  $\tilde{\mathbf{s}}_j = \mathsf{STFT}^{-1}(\mathbf{M}_j \odot \mathbf{X}).$ 





Nonnegative masking:  $\angle \mathbf{S}_j = \angle \mathbf{X}$ .

**X** Issues in sound quality when sources overlap.

 $\checkmark$  Inconsistency:  $\hat{\mathbf{S}}_j \notin \mathsf{STFT}(\mathbb{R}^N)$ .



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Multiple Input Spectrogram Inversion (MISI) [Gunawan, 2010]

▷ Extends the Griffin-Lim algorithm to multiple signals by solving:

$$\min_{\mathbf{s}_j} \sum_{j=1}^J \|\mathbf{V}_j - |\mathsf{STFT}(\mathbf{s}_j)|\|^2 \text{ s.t. } \sum_{j=1}^J \mathbf{s}_j = \mathbf{x}.$$

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#### Goal

Extend MISI to non-quadratic losses for source separation.

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### **Proposed method**

#### **Problem setting**

#### Bregman divergences

$$\mathcal{D}_{\psi}(\mathbf{P} \mid \mathbf{Q}) = \sum_{f,t} \psi(p_{f,t}) - \psi(q_{f,t}) - \psi'(q_{f,t})(p_{f,t} - q_{f,t})$$

- $\triangleright\,$  The generating function  $\psi$  determines the divergence.
- $\triangleright$  Encompass the  $\beta$ -divergences, with particular cases: Euclidean ( $\beta = 2$ ), Kullback-Leibler ( $\beta = 1$ ) and Itakura-Saito ( $\beta = 0$ ) [Hennequin, 2011]

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Problem formulation: 
$$\min_{\mathbf{s}_j} \sum_{j=1}^J \mathcal{C}_j(\mathbf{s}_j)$$
 s.t.  $\sum_{j=1}^J \mathbf{s}_j = \mathbf{x}$ 

▷ Accounting for the non-symmetry of Bregman divergences:

$$\mathcal{C}_{j}(\mathbf{s}_{j}) = \underbrace{\mathcal{D}_{\psi}(\mathbf{V}_{j} \mid |\mathsf{STFT}(\mathbf{s}_{j})|^{d})}_{\text{"right" problem}} \quad \text{or} \quad \underbrace{\mathcal{D}_{\psi}(|\mathsf{STFT}(\mathbf{s}_{j})|^{d} \mid \mathbf{V}_{j})}_{\text{"left" problem}}$$

 $\triangleright d = 1$  (V<sub>j</sub> are magnitudes) or d = 2 (V<sub>j</sub> are power spectrograms).

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#### Projected gradient descent



- > The set defined by the mixing constraint is convex.
- ▷ The data fitting terms are independent from each other.
- Projected gradient descent:

$$\begin{aligned} \mathbf{y}_{j} \leftarrow \mathbf{s}_{j} - \mu \nabla \mathcal{C}_{j}(\mathbf{s}_{j}) \\ \mathbf{s}_{j} \leftarrow \mathbf{y}_{j} + \frac{1}{J} \left( \mathbf{x} - \sum_{i=1}^{J} \mathbf{y}_{i} \right) \end{aligned}$$

 $\triangleright$  Compute the gradient  $\nabla C_j$  using the chain rule [Vial, 2021].

Vial et al., "Phase retrieval with Bregman divergences and application to audio signal recovery", IEEE JSTSP, 2021.

Initialization: Wiener-like mask:  $\mathbf{s}_j = \mathsf{STFT}^{-1}(\mathbf{V}_j^{1/d} \odot e^{\mathrm{i} \angle \mathbf{X}})$ 

#### Algorithm overview

Initialization: Wiener-like mask:  $s_j = STFT^{-1}(V_j^{1/d} \odot e^{i \angle X})$ Update rules

$$\mathsf{STFT} \qquad \qquad \mathbf{S}_j = \mathsf{STFT}(\mathbf{s}_j)$$

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> STFT Compute the gradient

$$\begin{split} \mathbf{S}_{j} &= \mathsf{STFT}(\mathbf{s}_{j}) \\ \mathbf{G}_{j} &= \psi''(|\mathbf{S}_{j}|^{d}) \odot (|\mathbf{S}_{j}|^{d} - \mathbf{V}_{j}) \text{ (right)} \\ \mathbf{G}_{j} &= \psi'(|\mathbf{S}_{j}|^{d}) - \psi'(\mathbf{V}_{j}) \text{ (left)} \end{split}$$

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Gradient descent

$$\begin{split} \mathbf{S}_{j} &= \mathsf{STFT}(\mathbf{s}_{j}) \\ \mathbf{G}_{j} &= \psi''(|\mathbf{S}_{j}|^{d}) \odot (|\mathbf{S}_{j}|^{d} - \mathbf{V}_{j}) \text{ (right)} \\ \mathbf{G}_{j} &= \psi'(|\mathbf{S}_{j}|^{d}) - \psi'(\mathbf{V}_{j}) \text{ (left)} \\ \mathbf{Y}_{j} &= \mathbf{S}_{j} - \mu d \times \mathbf{S}_{j} \odot |\mathbf{S}_{j}|^{d-2} \odot \mathbf{G}_{j} \end{split}$$

> STFT Compute the gradient

Gradient descent Inverse STFT

$$\begin{split} \mathbf{S}_{j} &= \mathsf{STFT}(\mathbf{s}_{j}) \\ \mathbf{G}_{j} &= \psi''(|\mathbf{S}_{j}|^{d}) \odot (|\mathbf{S}_{j}|^{d} - \mathbf{V}_{j}) \text{ (right)} \\ \mathbf{G}_{j} &= \psi'(|\mathbf{S}_{j}|^{d}) - \psi'(\mathbf{V}_{j}) \text{ (left)} \\ \mathbf{Y}_{j} &= \mathbf{S}_{j} - \mu d \times \mathbf{S}_{j} \odot |\mathbf{S}_{j}|^{d-2} \odot \mathbf{G}_{j} \\ \mathbf{y}_{j} &= \mathsf{STFT}^{-1}(\mathbf{Y}_{j}) \end{split}$$

 $\begin{array}{ll} \mathsf{STFT} & \mathbf{S}_{j} = \mathsf{STFT}(\mathbf{s}_{j}) \\ \mathsf{Compute the gradient} & \mathbf{G}_{j} = \psi''(|\mathbf{S}_{j}|^{d}) \odot (|\mathbf{S}_{j}|^{d} - \mathbf{V}_{j}) \text{ (right)} \\ & \mathbf{G}_{j} = \psi'(|\mathbf{S}_{j}|^{d}) - \psi'(\mathbf{V}_{j}) \text{ (left)} \\ \mathsf{Gradient descent} & \mathbf{Y}_{j} = \mathbf{S}_{j} - \mu d \times \mathbf{S}_{j} \odot |\mathbf{S}_{j}|^{d-2} \odot \mathbf{G}_{j} \\ \mathsf{Inverse STFT} & \mathbf{y}_{j} = \mathsf{STFT}^{-1}(\mathbf{Y}_{j}) \\ \mathsf{Mixing} & \mathbf{s}_{j} = \mathbf{y}_{j} + \frac{1}{J} \left( \mathbf{x} - \sum_{i=1}^{J} \mathbf{y}_{i} \right) \end{array}$ 

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MISI is a particular case (quadratic loss, d = 1, and  $\mu = 1$ ):

$$\mathbf{Y}_j = \mathbf{V}_j \odot rac{\mathbf{S}_j}{|\mathbf{S}_j|}$$

## Experiments

#### Protocol

- Task: speech enhancement (J = 2), 100 mixtures:
  - ▷ Clean speech from the VoiceBank dataset.
  - ▷ Real-life noises from the DEMAND dataset (living room, bus, and public square noises).
  - $\triangleright$  Mixtures at various input SNR (-10, 0, and 10 dB).

#### Magnitude estimation

- ▷ Open-Unmix (a freely available pretrained Bi-LSTM network).
- > The network is trained on different speakers and noises.

#### Split

- $\triangleright~50$  mixtures for validation (tuning the step size  $\mu$ ).
- $\triangleright$  50 mixtures for testing (MISI and the proposed algorithm, 5 iterations).

#### Results

Signal-to-distortion ratio (improvement over the baseline amplitude mask):



 $\triangleright$  The proposed method outperforms MISI when d = 2:

- $\triangleright$  At high/moderate input SNR when  $\beta > 1$ .
- $\triangleright$  At low input SNR for all  $\beta$  and the "left" problem.
- $\triangleright$  Performance peak around  $\beta = 1.25$ , close to Kullback-Leibler ( $\beta = 1$ ).
- ▷ Results depend on the type of noise.

# Alternative divergences have some potential for phase retrieval in audio source separation from highly corrupted spectrograms

Perspectives

- ▷ Alternative optimization schemes (majorization-minimization, ADMM).
- ▷ Inclusion within deep learning (e.g., with deep unfolding) for end-to-end separation.

