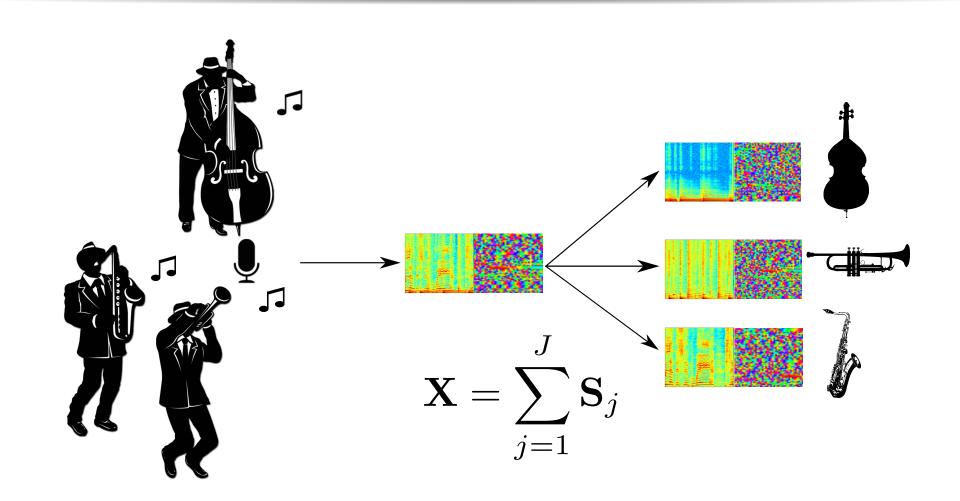
Online spectrogram inversion for low-latency audio source separation

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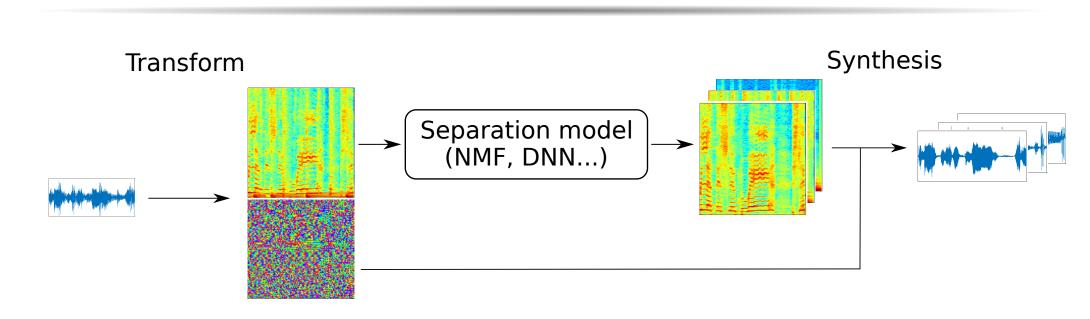
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Source separation



- Isolate individual sources from their mixture.
- Here: operate in the short-time Fourier transform (STFT) domain.

General framework



- Extract a nonnegative representation (magnitude/power spectrogram).
- Fit a structured model (nonnegative matrix factorization, deep neural network).
- Mask the mixture to retrieve isolated sources S_i .
- Synthesize time-domain signals through inverse STFT.

Phase recovery

Nonnegative masking $\rightarrow \angle S_i = \angle X$.

- The phase of the mixture is assigned to each source.
- Issues in sound quality when the sources overlap in the STFT domain.
- Inconsistent estimates: $\hat{\mathbf{S}}_i \notin \text{STFT}(\mathbb{R}^N)$.

Multiple Input Spectrogram Inversion (MISI) [1]

- Extends the Griffin-Lim algorithm to multiple signals in mixture models.
- Exhibits good phase recovery performance (as a post-processing or unfolded in a DNN).

Problems

- MISI has been introduced heuristically: no proof of convergence.
- It operates offline: non-applicable to real-time.

Deriving MISI

Problem setting

- Formulation in the time-domain: alleviates including a extra consistency constraint.
- Main objective: reduce the mismatch between the target and estimates' magnitudes.
- Add a mixing constraint: the estimates must add-up to the mixture.

Objective

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

Majorization-minimization

• Majorize the data fitting terms:

$$\|\mathbf{V}_{j} - |\operatorname{STFT}(\mathbf{s}_{j})|\|^{2} \le \|\mathbf{Y}_{j} - \operatorname{STFT}(\mathbf{s}_{j})\|^{2} \text{ with } |\mathbf{Y}_{j}| = \mathbf{V}_{j}$$

- Incorporate the mixing/magnitude constraints using Lagrange multipliers δ / Λ_i .
- New objective: find a saddle point for:

$$\|\mathbf{Y}_{j} - \operatorname{STFT}(\mathbf{s}_{j})\|^{2} + 2\Re \left(\boldsymbol{\delta}^{\mathsf{H}}(\sum_{j} \mathbf{s}_{j} - \mathbf{x})\right) + \sum_{j} \mathbf{\Lambda}_{j} \odot (|\mathbf{Y}_{j}|^{2} - \mathbf{V}_{j}^{2})$$

Update rules

Starting from initial estimates, alternate:

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

Set magnitude
$$\mathbf{Y}_j = \mathbf{V}_j \odot \frac{\mathbf{S}_j}{|\mathbf{S}_i|}$$

Inverse STFT
$$\mathbf{y}_j = iSTFT(\mathbf{Y}_j)$$

$$|\mathbf{S}_{j}|$$

$$\mathbf{s}_j = \mathbf{y}_j + \frac{1}{J} \left(\mathbf{x} - \sum_{i=1}^J \mathbf{y}_i \right)$$

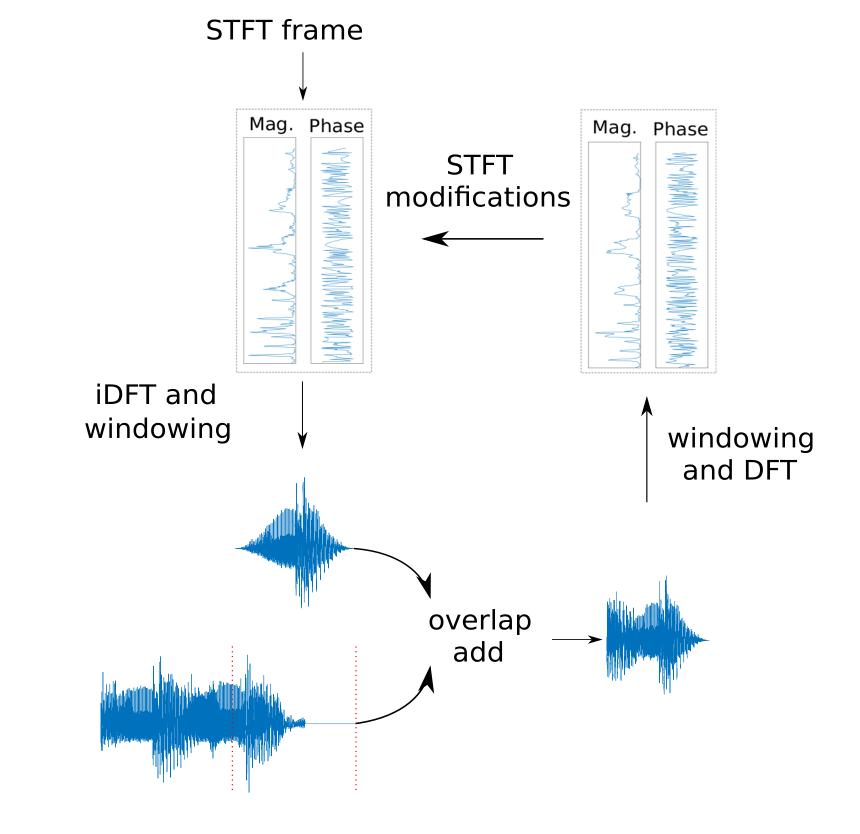
 \rightarrow MISI, but with a convergence guarantee.

Online MISI

MISI involves the inverse STFT, which does not operate online:

$$\mathbf{s}'_{j,t} = \mathrm{iDFT}(\mathbf{S}_{j,t}) \odot \mathbf{w}$$
 and $\mathbf{s}_{j}(n) = \sum_{t=0}^{j-1} \mathbf{s}'_{j,t}(n-t)$

Approach



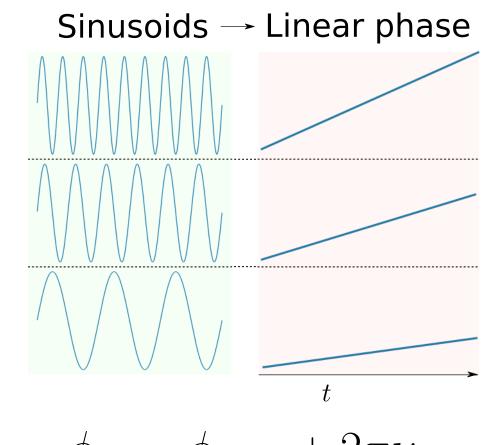
Split the overlap-add around the current frame:

$$\mathbf{s}_{j}(\mathbf{n}) = \sum_{k=0}^{t-1} \mathbf{s}_{j,k}'(\mathbf{n} - t\mathbf{l}) + \sum_{k=t}^{T-1} \mathbf{s}_{j,k}'(\mathbf{n} - t\mathbf{l})$$
past frames

past frames

Only use K future frames [2]: $\sum \mathbf{s}'_{j,k}(n-tl)$

Alternative initialization [3]



References

- [1] Gunawan and Sen, "Iterative phase estimation for the synthesis of separated sources from single-channel mixtures", IEEE Signal Processing Letters, vol. 17, no. 5, pp. 421–424, 2010.
- [2] Zhu et al., "Real-time signal estimation from modified short-time Fourier transform magnitude spectra," IEEE Transactions on Audio, Speech, and Language Processing, vol. 15, no. 5, pp. 1645–1653, 2007.
- [3] Magron et al., "Model-based STFT phase recovery for audio source separation," IEEE/ACM Transactions on Audio, Speech and Language Processing, vol. 26, no. 6, pp. 1095–1105, 2018.
- [4] Naithani et al., "Low latency sound source separation using convolutional recurrent neural networks," Proc. IEEE WASPAA, 2017.

Experimental protocol

Speech separation (J=2)

- Danish HINT dataset.
- Three speaker pairs (male+male, female+female, and male+female).

Magnitudes

• Each speaker magnitude is estimated using a lowlatency DNN [4].

Compared methods:

- Amplitude mask (AM).
- (Offline) MISI with 15 iterations.
- Online MISI with 15/(K+1) iterations, initialized with the mixture phase (oMISI-mix) or the sinusoidal phase (oMISI-sin).

Metric: Scale-invariant signal-to-distortion ratio improvement (higher is better).

Results

When the STFT uses a 50 % overlap ratio:

| | Latency | MF | MM | FF |
|-------------|--------------|-----|-----|-------------|
| AM | 16 ms | 7.5 | 5.7 | 5.1 |
| MISI | offline | 7.9 | 6.2 | 5.4 |
| oMISI - mix | 16 ms (K=0) | 7.7 | 6.1 | 5.4 |
| | 24 ms (K=1) | 7.9 | 6.2 | 5 .4 |
| | 32 ms (K=2) | 7.9 | 6.2 | 5.4 |
| oMISI - sin | 24 ms (K=1) | 7.8 | 6.2 | 5.4 |

- MISI > AM \rightarrow the relevance of phase recovery.
- oMISI with K = 1 performs as well as MISI: best trade-off between performance and latency.
- The optimal **K** depends on the overlap ratio: if there is 75 % overlap, then K = 3.
- The sinusoidal initialization does not improve the performance in this setting.
- But it does in an Oracle setting (ground truth magnitudes) for Female+Female mixtures.

Summary

- MISI is derived using majorization-minimization.
- An online implementation (with possible alternative initialization) is presented.
- oMISI reaches the same performance as MISI with a reduced latency.