

## Introduction

- **Harmonic/percussive source separation (HPSS)**, is a useful pre-processing tool for many audio applications.
- **Conventional:** The distinctive source-specific structures of amplitude spectrograms has been used.
- **Problem:** Phase reconstruction is required because of their amplitude-only treatment.
- **Proposal:** A **optimization-based HPSS method** simultaneously **treats the amplitude and phase**.
- **Results:** The numerical experiment validated the effectiveness of the proposed method in terms of SDR.

## HPSS based on anisotropic smoothness

- HPSS based on anisotropic smoothness assumes the power spectrograms of harmonic and percussive components have the following relation:

$$H_{\omega,\tau} \approx H_{\omega,\tau \pm 1}, \quad P_{\omega,\tau} \approx P_{\omega \pm 1,\tau}$$

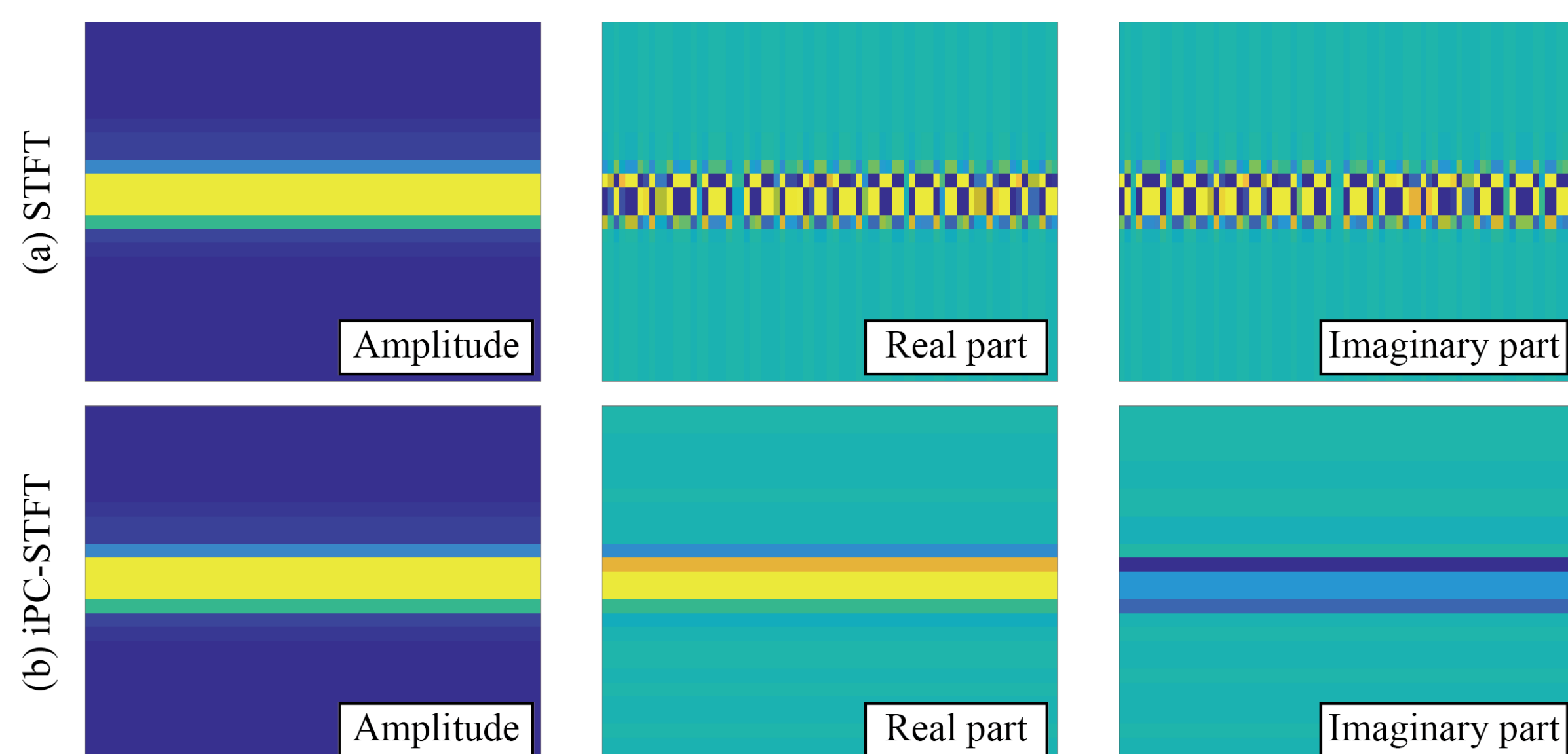
- Based on the above relation, a HPSS method is formulated as the following optimization problem.

$$\begin{aligned} \min_{\mathbf{H}, \mathbf{P}} \quad & \frac{1}{2\sigma_h^2} \|D_\tau(\mathbf{H})\|_{\text{Fro}}^2 + \frac{1}{2\sigma_p^2} \|D_\omega(\mathbf{P})\|_{\text{Fro}}^2 \\ \text{s.t.} \quad & H_{\omega,\tau} + P_{\omega,\tau} = |X_{\omega,\tau}|^{2\gamma}, \quad H_{\omega,\tau} \geq 0, \quad P_{\omega,\tau} \geq 0 \end{aligned}$$

- This approach considers **only amplitude spectrograms**, and thus **the phase information is ignored**.

## Property of complex-valued spectrograms

- Although the amplitude spectrogram of a **sinusoid** is constant in the time direction, its **complex-valued spectrogram periodically fluctuates**.



## Proposed phase-aware HPSS method

### Instantaneous phase corrected total variation

- The complex-valued spectrogram of a sinusoid has the following relation.

$$\mathcal{F}(\mathbf{x})_{\omega,\tau} = \mathcal{F}(\mathbf{x})_{\omega,\tau-1} e^{2\pi j f a / L}$$

- The complex-valued spectrogram becomes constant in each sub-band if its phase evolution is eliminated.
- We use **the instantaneous phase corrected STFT (iPC-STFT)** proposed in our previous study.

$$\mathcal{F}_{\text{iPC}}(\mathbf{x})_{\omega,\tau} = \prod_{\eta=0}^{\tau-1} e^{-2\pi j v_{\omega,\eta} a / L} \mathcal{F}(\mathbf{x})_{\omega,\tau}$$

- **The iPC-STFT spectrogram of a sinusoid is smooth in each sub-band** when the instantaneous frequency is accurately estimated.
- The instantaneous frequency can be estimated from the observed signal.

$$v_{\omega,\tau} = b\omega - \text{Im}[\tilde{\mathcal{F}}(\mathbf{x})_{\omega,\tau} / \mathcal{F}(\mathbf{x})_{\omega,\tau}]$$

## Proposed algorithm

### Primal-dual splitting algorithm

- Primal-dual splitting (PDS) algorithm is one of the proximal splitting algorithms which can solve the following convex optimization problem.

$$\min_{\mathbf{x}} \Theta(\mathbf{x}) + \Upsilon_1(\mathcal{L}_1(\mathbf{x})) + \Upsilon_2(\mathcal{L}_2(\mathbf{x}))$$

- To apply the PDS algorithm to the proposed HPSS problem, we reformulate it as follows.

$$\min_{\mathbf{x}_h, \mathbf{x}_p} \iota_{\mathbf{x}}(\mathbf{x}_h, \mathbf{x}_p) + \frac{1}{2} \|\mathcal{L}_h(\mathbf{x}_h)\|_{\text{Fro}}^2 + \lambda \|\mathcal{F}(\mathbf{x}_p)\|_{2,1}$$

#### Algorithm 1 Proposed HPSS algorithm

**Input:**  $\mathbf{x}, \mathbf{x}_h^{[0]}, \mathbf{x}_p^{[0]}, \mathbf{Y}_h^{[0]}, \mathbf{Y}_p^{[0]}, \lambda, \mu_1, \mu_2, \alpha$

**Output:**  $\mathbf{x}_h^{[n+1]}, \mathbf{x}_p^{[n+1]}$

**for**  $n = 1, 2, \dots$  **do**

$$(\tilde{\mathbf{x}}_h, \tilde{\mathbf{x}}_p) = P_{\mathbf{x}}(\mathbf{x}_h^{[n]} - \mu_1 \mathcal{L}_h^*(\mathbf{Y}_h^{[n]}), \mathbf{x}_p^{[n]} - \mu_1 \mathcal{F}^*(\mathbf{Y}_p^{[n]}))$$

$$\mathbf{z}_h = \mathbf{y}_h^{[n]} + \mathcal{L}_h(2\tilde{\mathbf{x}}_h - \mathbf{x}_h^{[n]})$$

$$\mathbf{z}_p = \mathbf{y}_p^{[n]} + \mathcal{F}(2\tilde{\mathbf{x}}_p - \mathbf{x}_p^{[n]})$$

$$\tilde{\mathbf{y}}_h = \mathbf{z}_h - \mu_2 \text{prox}_{(1/\mu_2)\|\cdot\|_{\text{Fro}}}(\mathbf{z}_h/\mu_2)$$

$$\tilde{\mathbf{y}}_p = \mathbf{z}_p - \lambda \mu_2 \text{prox}_{(1/\lambda\mu_2)\|\cdot\|_{2,1}}(\mathbf{z}_p/\lambda\mu_2)$$

$$(\mathbf{x}_{h,p}^{[n+1]}, \mathbf{y}_{h,p}^{[n+1]}) = \alpha(\tilde{\mathbf{x}}_{h,p}, \tilde{\mathbf{y}}_{h,p}) + (1 - \alpha)(\mathbf{x}_{h,p}^{[n]}, \mathbf{y}_{h,p}^{[n]})$$

**end for**

### Proposed optimization-based HPSS method

- Using iPC-STFT, we propose a **phase-aware HPSS method** through the following **convex optimization**.

$$\begin{aligned} \min_{\mathbf{x}_h, \mathbf{x}_p} \quad & \frac{1}{2} \|\mathbf{W} \odot D_\tau(\mathbf{X}_h)\|_{\text{Fro}}^2 + \lambda \|\mathbf{X}_p\|_{2,1} \\ \text{s.t.} \quad & \mathbf{x} = \mathbf{x}_h + \mathbf{x}_p, \quad \mathbf{X}_h = \mathcal{F}_{\text{iPC}}(\mathbf{x}_h), \quad \mathbf{X}_p = \mathcal{F}(\mathbf{x}_p) \end{aligned}$$

$\mathbf{x}$  : the time domain observed signal       $\lambda$  : a regularization parameter  
 $\mathbf{X}_h$ : the time domain signal of harmonic components       $\mathbf{W}$  : a data-dependent weight  
 $\mathbf{X}_p$ : the time domain signal of percussive components

- Main points of the proposed method:
  - **The phase-aware smoothness** in the time direction is assumed for **harmonic components**.
  - **The time-frame-wise sparsity** is considered for **percussive components** instead of the frequency-directional smoothness.
  - It treats **variables in the time domain**, and the perfect reconstruction constraint is introduced.
  - It is formulated as a **convex optimization** problem.

## Numerical experiment

- The proposed method was compared with the anisotropic smoothness based method (Ono's), median-filtering (MF), kernel additive model (KAM), and phase-aware time-frequency masking (PM).
- **Prop-mix** (using instantaneous frequency estimated from the noisy signal) **outperformed the conventional methods in terms of SDR**, and Prop-ora achieved the highest SDR.

Experimental condition			Ono's	MF	KAM	PM	Prop-mix	Prop-ora
Sampling rate	44100 Hz	Har.	SDR 5.8	8.6	4.9	-8.6	<b>9.3</b>	10.3
			SIR 11.2	15.1	<b>23.1</b>	6.1	12.3	13.8
			SAR 7.6	10.2	5.1	-7.5	<b>15.4</b>	15.4
Window	Hann	Per.	SDR -8.1	-4.2	-4.7	-12.1	<b>-3.8</b>	-2.7
			SIR -2.8	-1.3	-2.3	-3.2	<b>1.7</b>	2.8
			SAR -1.9	3.5	<b>4.2</b>	-6.7	1.6	2.6
Window length	93 ms	Ave.	SDR -1.1	2.0	0.1	-10.4	<b>2.8</b>	3.8
			SIR 4.2	6.9	<b>10.4</b>	1.5	7.0	8.3
			SAR 2.8	6.9	4.6	-7.1	<b>8.5</b>	9.0
Shift length	23 ms							
$\lambda$	0.5							

## Conclusion

- We proposed a phase-aware HPSS method through convex optimization which treats both amplitude and phase simultaneously.
- The experimental results indicated the accurate estimation of the instantaneous frequency can improve the performance of the proposed method, which is included in our future works.