

WASEDA U	niversity
早稲田大学	,



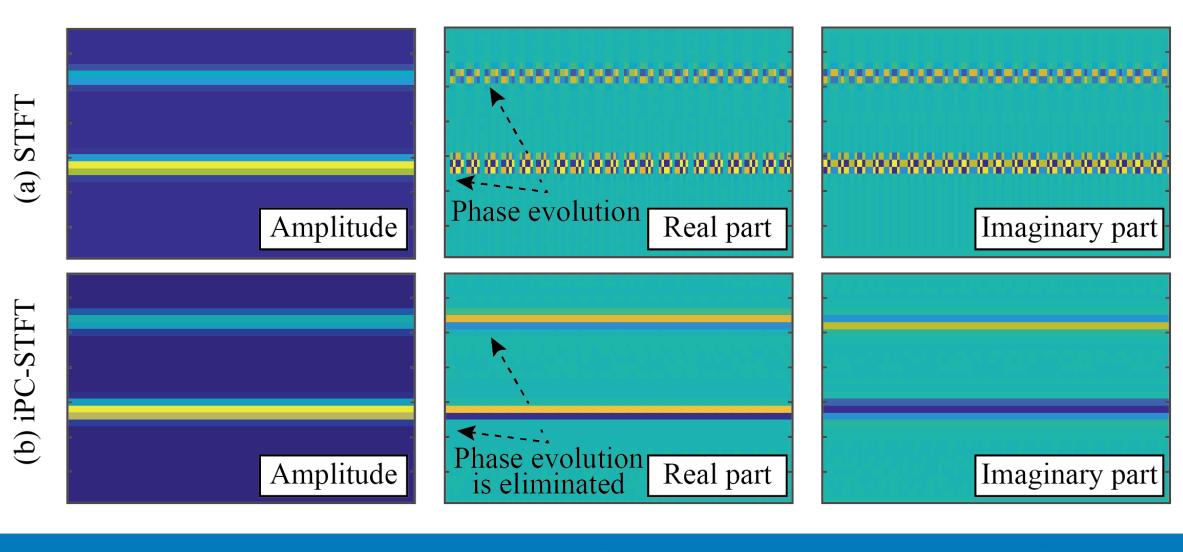
# Introduction

- Low-rankness of amplitude spectrograms has been effectively used in audio signal processing methods including NMF.
- **Problem**: Such methods suffer from the problem of phase reconstruction because they focused on the amplitude treatment.
- **Proposal 1: A complex-valued spectrogram** is modeled by a low-rank complex-valued matrix by modifying its phase.
- **Proposal 2**: A prior emphasizing harmonic signals is presented based on the proposed low-rank modeling.
- **Results**: In audio denoising, an optimization-based method using the proposed prior outperformed NMF and its variants.

# **Property of complex-valued spectrogram**

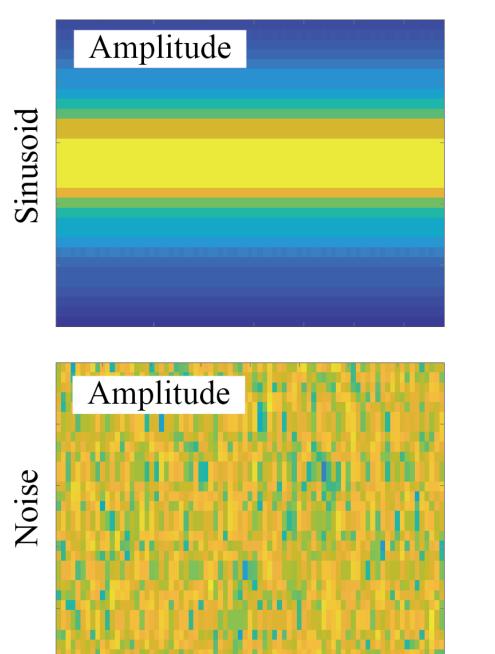
#### **Rank of complex-valued spectrograms**

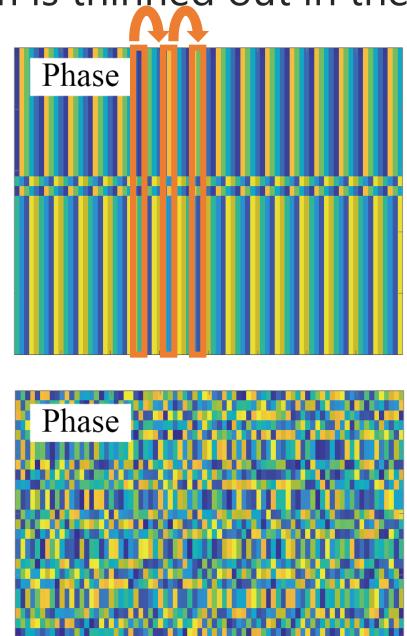
- Previous work proved that the complex-valued spectrogram of a sum of *r* complex exponentials becomes rank-*r* in some conditions.
- A complex-valued spectrogram of a sum of sinusoids is not lowrank when the number of sinusoids increases.
- Direct low-rank modeling of a complex-valued spectrogram can not be considered for audio signals.



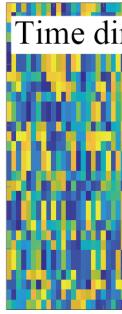
# **Property of time-difference of phase**

• The time directional phase difference of a complex exponential is constant, and thus the phase in a future frame is predictable even if the spectrogram is thinned out in the time direction.





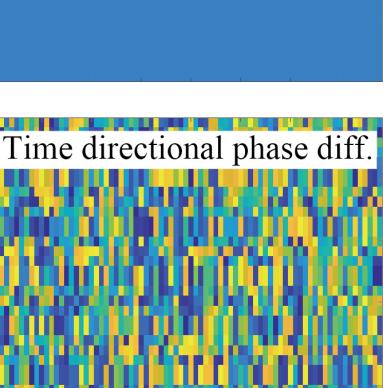




# Low-rankness of complex-valued spectrogram and its application to phase-aware audio processing

Yoshiki Masuyama\*, Kohei Yatabe, Yasuhiro Oikawa (Waseda University, Japan)

Time directional phase diff.



# Proposed low-rank modeling method of complex-valued spectrograms

#### Instantaneous phase corrected STFT (iPC-STFT)

- If the phase evolution of each sinusoid is eliminated, the complex-valued spectrograms of a sum of sinusoids can be low-rank.
- Assuming each sinusoid is sufficiently separated, its phase spectrogram has the following relation.

 $\phi[\xi, \tau + 1] = \phi[\xi, \tau] + 2\pi a v[\xi, \tau]/L$ 

• To cancel the phase evolution, we use the instantaneous phase corrected STFT (iPC-STFT).

$$\mathcal{G}_{iPC}^{\mathbf{w}}(\mathbf{x})[\xi,\tau] = \prod_{\eta=0}^{\tau-1} e^{-2\pi j a v[\xi,\eta]/2}$$

 The instantaneous frequency can be estimated from the observed signal.

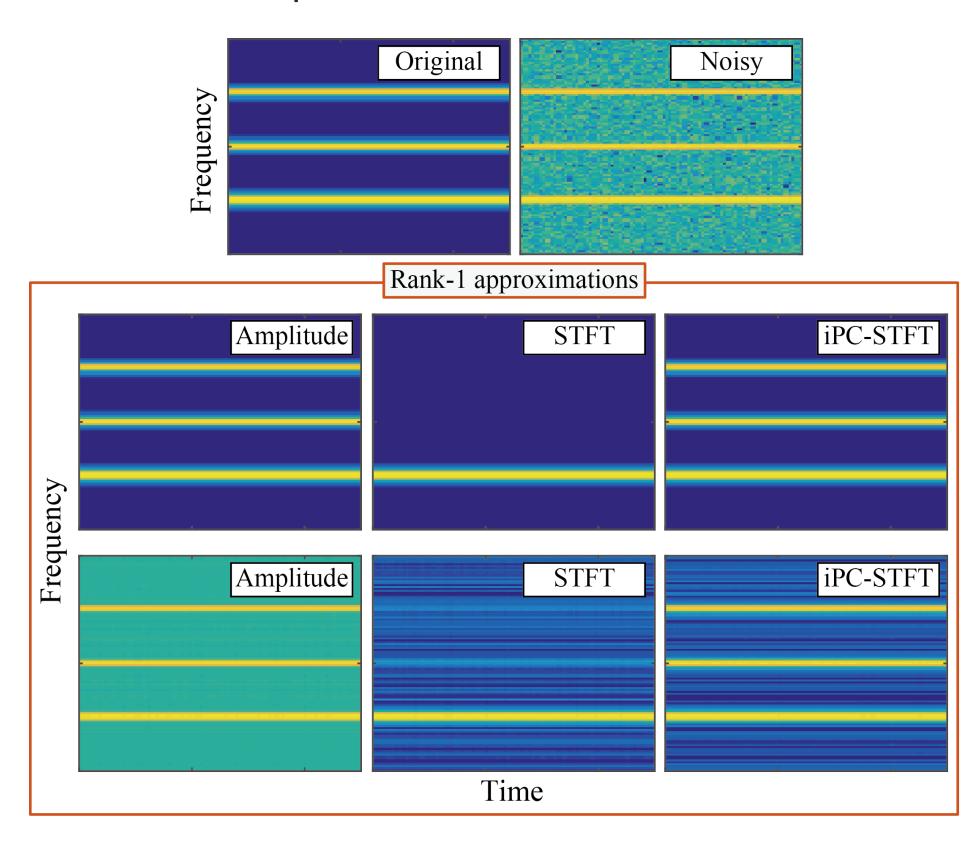
 $v[\xi, \tau] = b\xi - \operatorname{Im}[\mathcal{G}^{\mathbf{w}'}(\mathbf{x})[\xi, \tau] / \mathcal{G}^{\mathbf{w}}(\mathbf{x})[\xi, \tau]]$ 

- Once the instantaneous frequency is estimated, iPC-STFT is defined as an invertible linear transform
- iPC-STFT also eliminated the phase evolution of a locally stational signal whose phase spectrogram satisfies the above relation.

## Toy example

#### **Rank-1 approximations of clean/noisy sinusoids**

- In both cases, the rank-1 approximation of iPC-STFT spectrogram represented all sinusoids, while that of STFT represented only one sinusoid.
- In the noisy case, the rank-1 approximation of iPC-STFT spectrogram effectively reduced the noise, while that of amplitude did not.



\*Email: mas-03151102@akane.waseda.jp

 $\mathcal{C}^{L}\mathcal{G}^{\mathbf{w}}(\mathbf{x})[\xi,\tau]$ 

### Low-rankness of iPC-STFT spectrogram

$$\begin{aligned} \mathcal{G}_{iPC}^{\mathbf{w}}(\mathbf{s})[\xi,\tau+1] &= \prod_{\eta=0}^{\tau} e^{-\tau} \\ &= \prod_{\eta=0}^{\tau-1} e^{-\tau} \\ &= \int_{\eta=0}^{\tau} e^{-\tau} \end{aligned}$$

### Harmonic signal prior based on proposed low-rank modeling

with the instantaneous frequency estimated in advance.

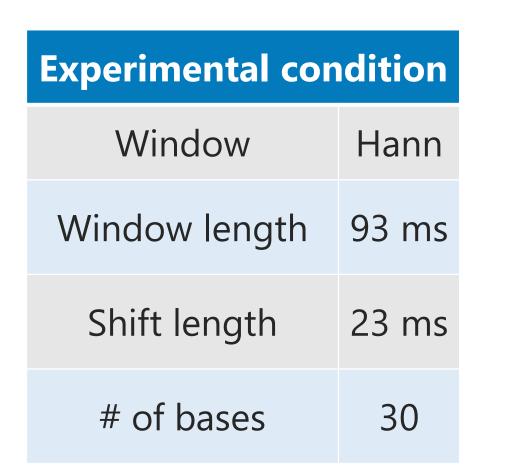
 $\mathcal{P}_{iPCLR}(\mathbf{x}) = \left\| \mathcal{G}_{iPC}^{\mathbf{w}}(\mathbf{x}) \right\|_{*}$ 

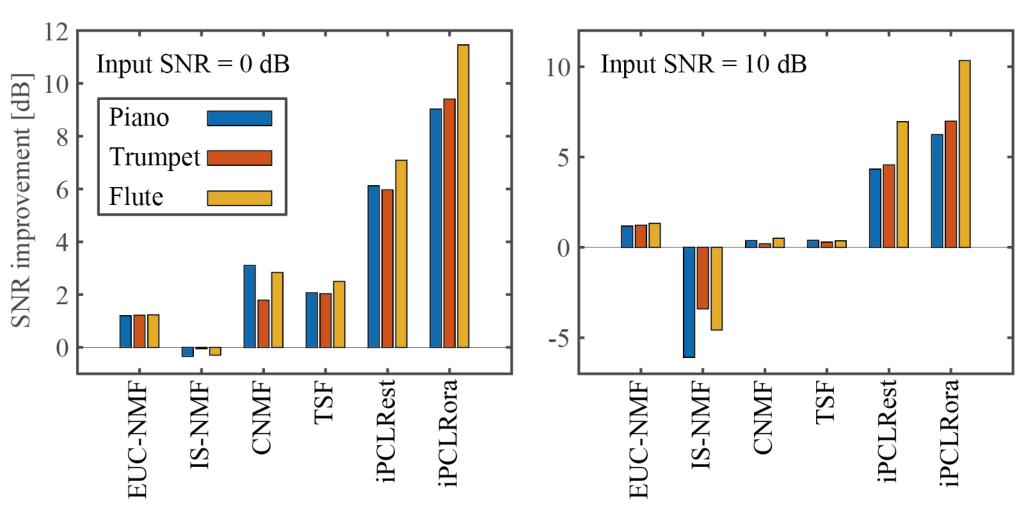
considered for audio denoising.

$$\mathbf{x}^{\star} = \arg\min_{\mathbf{x}} \frac{1}{2} \|\mathbf{x} - \mathbf{x}\|_{\mathbf{x}}$$

# **Application to audio denoising**

- sions using three melodies played by different musical instruments.





# Conclusion

- amplitude by applying iPC under mild assumptions.
- Seeking further applications of iPCLR remains as future works.

• Based on the property of iPC-STFT, the following relation is obtained for the iPC-STFT spectrogram of a sum of sinusoids, and thus it becomes rank-1.

 $\mathcal{L}^{-2\pi jv[\xi,\eta]a/L}\mathcal{G}^{\mathbf{w}}(\mathbf{s})[\xi,\tau+1]$ 

 $\mathcal{L}^{-2\pi jv[\xi,\eta]a/L}\mathcal{G}^{\mathbf{w}}(\mathbf{s})[\xi,\tau]$ 

 $= \mathcal{G}^{\mathbf{w}}(\mathbf{s})[\xi, 0] \quad (= \mathcal{G}^{\mathbf{w}}_{iPC}(\mathbf{s})[\xi, 0])$ 

• For applying the proposed low-rank modeling to audio signal processing, we propose a prior of harmonic signals, named iPC low-rankness (iPCLR)

• Using the proposed iPCLR, the following convex optimization problem is

# $\| - \mathbf{d} \|_{2}^{2} + \lambda \mathcal{P}_{iPCLR}(\mathbf{x}) \|$

• The proposed method was compared with NMF and its phase-aware exten-

• **iPCLRest** (using instantaneous frequency estimated from the noisy signal) outperformed the conventional methods, and iPCLRora (using instantaneous frequency estimated from the clean signal) achieved the highest SNR.

• We showed that the rank of a complex-valued spectrogram can be as low as its

Based on the low-rankness, we further proposed a prior of harmonic signals.