

Rotation-robust beamforming based on sound field interpolation with regularly circular microphone array

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Introduction

Background

- Array signal processing assumes a time-invariant ATS^{*}.
- ATS's variation makes re-estimation of spatial information (e.g., spatial filter and DOA⁺) necessary.
 - \Rightarrow makes online processing difficult.
- Motivation

*: acoustic transfer system
 +: direction of arrival
 **: circular microphone array

- We want to follow the ATS's variation caused by CMA^{**} rotation.
- We want to apply interpolation to existing beamformings.



Prior work for ATS's variation

- Case of source movement
 - Blockwise spatial filter estimation using DOA information [Nikunen+, 2018], [Naqvi+, 2011]
 - Sequential covariance estimation every time-frequency bins
 - with a Bayesian tracker [Taseska+, 2018]





- Case of sensor movement
 - Motion compensation method [Tourbabin+, 2015]
 - Formulating circular harmonic domain rotation matrix

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Applying to DOA estimation using a CMA

Idea of sound field interpolation

Continuous sound field on circle's circumference, z(θ)
 z(θ) is a periodic function with 2π

Discretizing $z(\theta)$ with $2\pi/M$ intervals

= Observing the sound field using an M-ch CMA

$$z_m = z \left(2\pi \frac{m}{M}\right), \quad m = 0, \dots, M-1.$$

z(θ) can be reconstructed by the discrete signal z_m if the sampling theorem is satisfied^{*}, resulting in **sound field interpolation** possible.



Sound field interpolation using noninteger sample shift

Relationship b/w CMA rotation and sample shift

- Sound field observed by a CMA in the reference position, z_m
- Sound field observed by the CMA rotated Δ (= $2\pi\delta/M$) [rad]

= δ -sample shifted z_m in the spatial axis

$$z_{m+\delta} = z \left(2\pi \frac{m}{M} + \Delta \right)$$

The above equation enables estimating $z_{m+\delta}$ from z_m , m=0,..., M-1.



Formulation of linear interpolation

Linear representation of δ -sample shifted sound field using the sample shift theorem in the DFT

$$z_{m+\delta} = \mathscr{F}_{\mathrm{D}}^{-1} \left[\mathscr{F}_{\mathrm{D}} \left[z_m \right] e^{j\Delta k} \right]$$
$$= \frac{1}{M} \sum_{k=-M/2+1}^{M/2} \left(Z_k e^{j\Delta k} \right) e^{-j2\pi \frac{mk}{M}} \stackrel{\mathrm{def}}{=} \sum_{n=0}^{M-1} z_n u_{m,n,\delta}$$

 \mathscr{F}_D : DFT operation

Sound field interpolation using sinc function (It is derived from the equation above^{*}.)

*: The detailed derivation is appended on the last page.

$$u_{m,n,\delta} = \begin{cases} \frac{1+(-1)^{n-m}e^{j\delta\pi}}{M} + \frac{\operatorname{sinc}\left(L/2\right)}{\operatorname{sinc}\left(L/M\right)} \cdot \cos\left(\frac{M+2}{2M}L\pi\right), & M \text{ is even,} \\ \frac{1}{M} + \frac{M-1}{M} \frac{\operatorname{sinc}\left(L(M-1)/2M\right)}{\operatorname{sinc}\left(L/M\right)} \cdot \cos\left(\frac{M+1}{2M}L\pi\right), & M \text{ is odd.} \\ L = n - m - \delta \end{cases}$$
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Remark of formulation

How to handle Nyquist frequency (NyqF) component

$$z_{m+\delta} = \sum_{n=0}^{M-1} z_n u_{m,n,\delta} \quad (*)$$

$$u_{m,n,\delta} = \begin{cases} \frac{1+(-1)^{n-m}}{M} + \frac{\operatorname{sinc}(L/2)}{\operatorname{sinc}(L/M)} \cdot \cos\left(\frac{M+2}{2M}L\pi\right), & M \text{ is even,} \\ \frac{1}{M} + \frac{M-1}{M} \frac{\operatorname{sinc}(L(M-1)/2M)}{\operatorname{sinc}(L/M)} \cdot \cos\left(\frac{M+1}{2M}L\pi\right), & M \text{ is odd.} \end{cases}$$

When M is even, a noninteger sample shift results in a complexvalued term even if the sign of the NyqF is positive or negative, and which causes some contradiction;

e.g., sample shift of a real-valued even point signal by (*) translates to a complex-valued signal.

In this study, we neglect the NyqF component by setting δ =0.

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Matrix representation

Formulation of the relationship b/w the observation by a Δ -rotated CMA, **x**, and by the CMA w/o rotation, **x**₀

 $\boldsymbol{x} = \begin{bmatrix} x_1 & \cdots & x_M \end{bmatrix}^{\mathrm{T}} \\ = \begin{bmatrix} z \left(2\pi (0+\delta)/M \right) & \cdots & z \left(2\pi (M-1+\delta)/M \right) \end{bmatrix}^{\mathrm{T}} \\ = \begin{bmatrix} u_{0,0,\delta} & u_{0,1,\delta} & \cdots & u_{0,M-1,\delta} \\ u_{1,0,\delta} & u_{1,1,\delta} & \cdots & u_{1,M-1,\delta} \\ \vdots & \vdots & \ddots & \vdots \\ u_{M-1,0,\delta} & u_{M-1,1,\delta} & \cdots & u_{M-1,M-1,\delta} \end{bmatrix} \begin{bmatrix} z_0 \\ z_1 \\ \vdots \\ z_{M-1} \end{bmatrix} \\ = \boldsymbol{U}_{\Delta} \boldsymbol{x}_0.$

Note: \mathbf{U}_{Δ} is a cyclic matrix & does not depend on frequencies

[cf.] Our method has a tight relationship to the circular harmonics domain rotation matrix used by the motion compensation method [Tourbabin+, 2015].

Applying interpolation to beamforming

Situation: A user or humanoid robot

- wears a CMA on the head.
- rotates the head to listen to ambient conversations attentively.



Problem:

- \Box The rotation angle Δ is given.
- We estimate the observation before CMA rotation.
- We do beamforming using interpolated M-ch signals.

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Evaluation condition

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- Mixture: 2 sources from SiSEC [Araki+, 2012], fs=16 kHz, 12 envs.
- RIR : Simulator [Habets, 2008], RT60 = 100 ms
- STFT: 64 ms Hamming window with 1/8 shifts
- Exp. 1: interpolation performance
 □ Array rot. △: 10, 20, and 30 deg
 □ Measure: SER (signal-to-error ratio)

$$\operatorname{SER}_{m,k} = 10 \log_{10} \left(\frac{\sum_{t} |x_{m,t,k}|^2}{\sum_{t} |\hat{x}_{m,t,k} - x_{m,t,k}|^2} \right) \\ x_{m,t,k} \in \mathbb{C}$$

- Exp. 2: signal enhancement
 - ☐ Array rot. △: 10, 20, and 36 deg
 - Beamformer: MPDR* + RTF** [Doclo+, 2015]
 - 2 methods of applying beamformer.
 - Measure : SDR, SIR [Vincent+, 2006]



*minimum power distortionless response **relative transfer function

How to apply to beamforming

Method 1: Constantly use of pre-estimated spatial filter
 Pre-estimation of spatial filter when the CMA does not rotate

$$\boldsymbol{w} = \mathrm{MPDR}(\boldsymbol{V}, \boldsymbol{a}), \ \ \boldsymbol{V} = \mathbb{E}\left[\boldsymbol{x}_{0}\boldsymbol{x}_{0}^{\mathrm{H}}\right]$$

Interpolation & beamforming $y = \boldsymbol{w}^{\mathrm{H}} \hat{\boldsymbol{x}}_0 = \boldsymbol{w}^{\mathrm{H}} \left(\boldsymbol{U}_{-\Delta} \boldsymbol{x} \right)$

interpolatior

 Δ : rotation angle

a : transfer function

 $oldsymbol{x}_0$: observation in the reference position

Method 2: Re-estimation of spatial filter

 x_{c}

covariance

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Exp. 1: Interpolation performance 1/2

Examples of SER when 2 sources are active

Lower band interpolation worked well, but not higher one.



Exp. 1: Interpolation performance 2/2

- Averaged SER improvement (up to 3 kHz)
 - More the number of microphones improves performance.
 - **D** NyqF components decrease performance when M=4, $6\downarrow$

e.g., when M=4, only 3 components contribute to interpolation.



Exp. 2: Signal enhancement

SDR & SIR, M=5

□ In No-Int case, bigger rotation decreases SDR & SIR. Int & Int+Re-est come close to the performance of No-Rot. rotation angle rotation angle Good 20 deg 20 deg 10 deg 36 deg 10 dea 36 deg 25F 16 20 12 SDR [dB] [gp] XIS Bad 10° ROT INT INT EST INT INT EST INT INT EST NOPROC ROT INT INT EST INT INTERT INT INT EST INT INTERT INTERT INT INTERT INTERT INT INTERT INT INTERT INTERT INT INTERT INTER No-Proc: unprocessed, No-Rot: no rotation, No-Int: no interpolation, Int: interpolation, Int + Re-est: interpolation + filter re-estimation IEEE ICASSP 2021 Full virtual lune 2021

Conclusion

Summary

- Sound field interpolation method for rotation-robust beamforming using CMA.
- Lower band interpolation accuracy was high when M is odd.
- Higher band interpolation accuracy was low, but applying to beamforming worked well.

Future work

- Applying to different array processing, e.g., source separation
- Clarifying relation to circular harmonics domain beamforming

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Derivation (1/2)

Derivation (2/2)

$$\begin{aligned} \text{Set} \quad L &= n - m - \delta \\ u_{m,n,\delta} &= \frac{1}{M} \left\{ 1 + \frac{W^{L\frac{M}{2}}}{1} + \sum_{k=1}^{M/2-1} W^{Lk} + \sum_{k=1}^{M/2-1} W^{-Lk} \right\} \\ &= \frac{1}{M} \left\{ 1 + W^{L\frac{M}{2}} + W^{L} \frac{1 - W^{L\frac{M}{2}}}{1 - W^{L}} + W^{-L} \frac{1 - W^{-L\frac{M}{2}}}{1 - W^{-L}} \right\} \\ &= \frac{1}{M} \left\{ 1 + W^{L\frac{M}{2}} + W^{\frac{L}{4}(M+2)} \frac{W^{-L\frac{M}{4}} - W^{L\frac{M}{4}}}{W^{-\frac{L}{2}} - W^{\frac{L}{2}}} + W^{-\frac{L}{4}(M+2)} \frac{W^{L\frac{M}{4}} - W^{-L\frac{M}{4}}}{W^{\frac{L}{2}} - W^{-\frac{L}{2}}} \right\} \\ &= \frac{1}{M} \left(1 + W^{L\frac{M}{2}} + W^{\frac{L}{4}(M+2)} \frac{\sin(\pi L/2)}{\sin(\pi L/M)} + W^{-\frac{L}{4}(M+2)} \frac{\sin(\pi L/2)}{\sin(\pi L/M)} \right) \\ &= \frac{1}{M} \left(1 + W^{L\frac{M}{2}} \right) + \frac{1}{2} \cdot \frac{\operatorname{sinc}(L/2)}{\operatorname{sinc}(L/M)} \left(W^{\frac{L}{4}(M+2)} + W^{-\frac{L}{4}(M+2)} \right) \\ &= \frac{1}{M} \left(1 + e^{-jL\pi} \right) + \frac{\operatorname{sinc}(L/2)}{\operatorname{sinc}(L/M)} \cdot \cos\left(\frac{M+2}{2M}L\pi\right). \end{aligned}$$

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