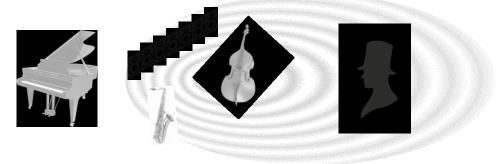
How the distance and radius of two circular loudspeaker arrays affect sound field reproduction and directivity controls

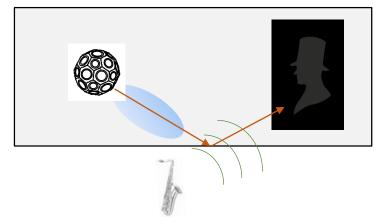
Yi Ren and Yoichi Haneda
The University of Electro-Communications
1-5-1 Chofugaoka, Chofu, Tokyo 182-8585, Japan
ren.yi@uec.ac.jp

Introduction

Sound field reproduction (SFR)



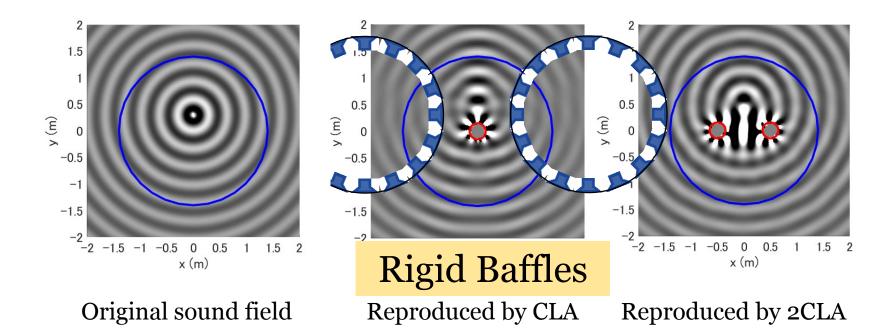
• Directivity control (DC)



- · 2-D Sound Field
 - Circle -> Infinite Cylinder

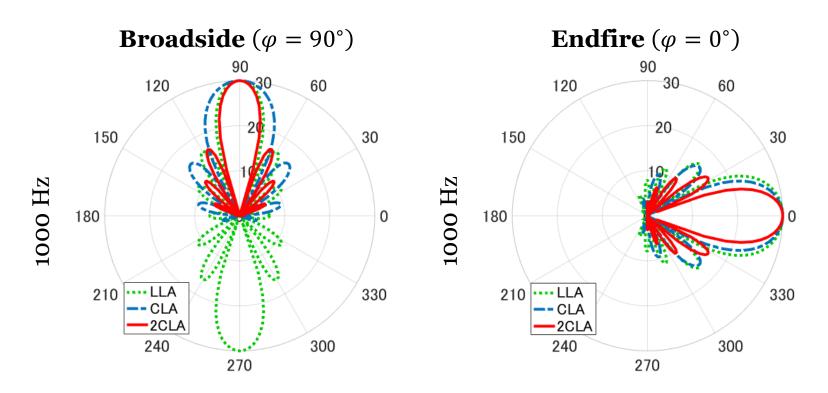
Previous Works

• SFR using a Two Circular Loudspeaker Array (2CLA) model w/ comparison with a single Circular Loudspeaker Array (CLA) (Ren & Haneda, 2018)



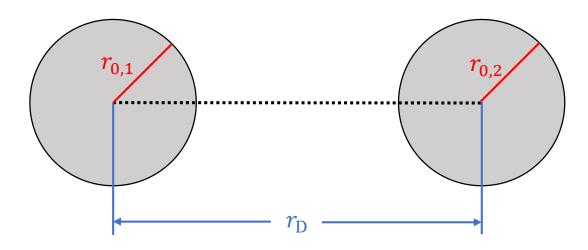
Previous Works

• DC using 2CLA w/ comparison with CLA and Linear Loudspeaker Array (LLA) (Ren & Haneda, 2019)



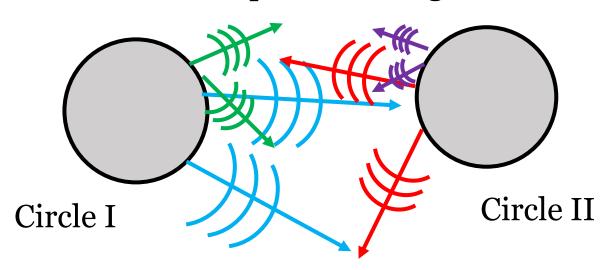
Properties

- Properties of a CLA
 - Radius r_0
 - Absorption coefficient
 - Sampling the circle
- Properties of a 2CLA
 - Radius of circle 1 $r_{0,1}$ and circle 2 $r_{0,2}$ \longrightarrow $r_0 = r_{0,1} = r_{0,2}$
 - Distance between the centers of the two circles $r_{\rm D}$



Transfer function of the 2CLA

Multiple Scattering



$$G_{(\zeta)}(\boldsymbol{r}|\boldsymbol{r}') = \left(\boldsymbol{\psi}_{(\zeta)}^{\mathrm{T}} + \boldsymbol{\psi}_{(\overline{\zeta})}^{\mathrm{T}} \mathbf{T}_{(\overline{\zeta})} + \boldsymbol{\psi}_{(\zeta)}^{\mathrm{T}} \mathbf{T}_{(\zeta)} \mathbf{T}_{(\overline{\zeta})} + \cdots\right) \boldsymbol{\gamma}_{(\zeta)} \quad (1)$$
direct sound
1st reflection

Transfer function of the 2CLA

$$G_{(\zeta)}(\boldsymbol{r}|\boldsymbol{r}') = \left(\boldsymbol{\psi}_{(\zeta)}^{\mathrm{T}} + \boldsymbol{\psi}_{(\bar{\zeta})}^{\mathrm{T}} \mathbf{T}_{(\bar{\zeta})} + \boldsymbol{\psi}_{(\zeta)}^{\mathrm{T}} \mathbf{T}_{(\zeta)} \mathbf{T}_{(\bar{\zeta})} + \cdots\right) \boldsymbol{\gamma}_{(\zeta)} \quad (1)$$

 $\boldsymbol{\gamma}_{(\zeta)}, \boldsymbol{\psi}_{(\zeta)}$ is $(2N+1) \times 1$ vectors of $\gamma_{\nu,(\zeta)}$ and $\psi_{\nu,(\zeta)}$

$$\gamma_{\nu,(\zeta)} = -\frac{e^{-j\nu\phi'_{(\zeta)}}}{2\pi k r_{0,(\zeta)} H_{\nu}^{(2)'}(k r_{0,(\zeta)})}, \quad \psi_{\nu,(\zeta)} = H_{\nu}^{(2)}(k r_{(\zeta)}) e^{j\nu\phi_{(\zeta)}},$$

 $\mathbf{T}_{(\zeta)}$ is a $(2N+1)\times(2N+1)$ matrix for multiple scattering

$$T_{\nu,\mu,(1)} = -\frac{J'_{\mu}(kr_{0,(1)})}{H_{\mu}^{(2)'}(kr_{0,(1)})} H_{\mu-\nu}(kr_{D})$$

$$J'_{\nu}(kr_{0,(2)})$$

$$T_{\nu,\mu,(2)} = -\frac{J'_{\mu}(kr_{0,(2)})}{H_{\mu}^{(2)'}(kr_{0,(2)})} H_{\nu-\mu}(kr_{D})$$

k: wavenumber, $J_{\nu}(z)$: Bessel function, $H_{\nu}^{(2)}(z)$: Hankel function of the 2nd kind

How the r_0 and r_D matter?

I. Sound field reproduction

II. Directivity control

I. Sound Field Reproduction

Pressure-matching method (PMM)

$$\mathbf{d} = \frac{\mathbf{G}^{\mathrm{H}}\mathbf{P}}{\mathbf{G}^{\mathrm{H}}\mathbf{G} + \lambda \mathbf{I}}$$

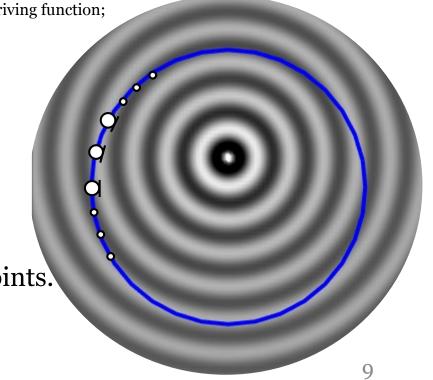
P: sound pressure at control points; G: transfer function; d: driving function;

 λ : regularization parameter.

 Original sound field: monopole at (0, 0.25 m).

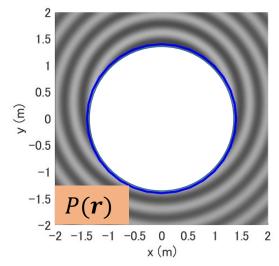
• Frequency domain design.

Control area: outside the control points.

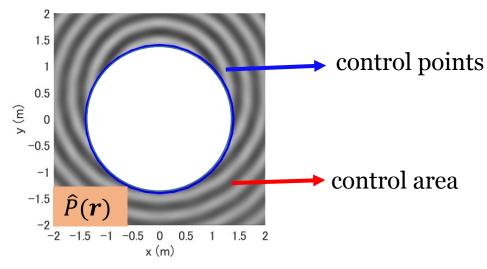


I. Sound Field Reproduction

Evaluation



Original sound field



Reproduced sound field

• Signal-to-Distortion Ratio (SDR)
$$SDR = 10 \log_{10} \frac{\int |P(\mathbf{r})|^2 d\mathbf{r}}{\int |P(\mathbf{r}) - \hat{P}(\mathbf{r})|^2 d\mathbf{r}}$$

Simulation conditions

Fixed distance r_D :

$$r_0 = \begin{cases} 0.075 \text{ m} \\ 0.15 \text{ m} , r_D = 1 \text{ m} \\ 0.3 \text{ m} \end{cases}$$

Fixed radius r_0 :

$$r_0 = 0.15 \text{ m}, r_D = \begin{cases} 0.5 \text{ m} \\ 1 \text{ m} \\ 1.5 \text{ m} \end{cases}$$

Frequencies:

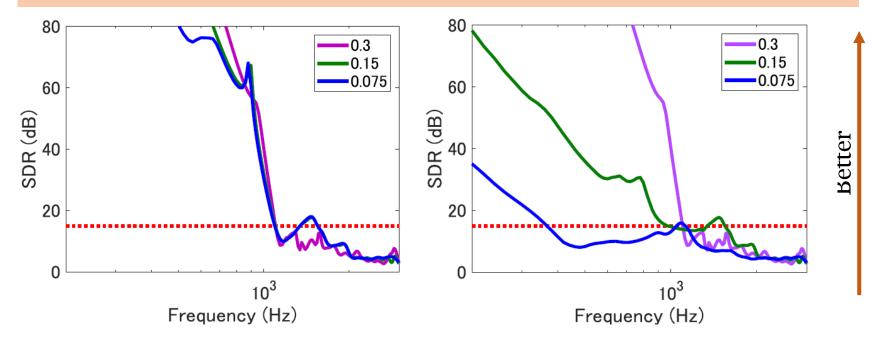
$$200 - 3000 \text{ Hz}$$

Regularization parameter λ :

$$\lambda = 0$$
: Theoretical $\lambda = {\lambda | W_0 = 0 \text{ dB}}$: Pratical Filter Gain $W_0 = 10 \log_{10} ||\mathbf{w}||_2^2$

I. Sound Field Reproduction

 r_0 : Doesn't matter for $\lambda = 0$. Do affect the filter gain. Better performance with larger r_0 when filter gain is restricted.



I. Sound Field Reproduction

 $r_{\rm D}$: Better performance with smaller $r_{\rm D}$. 80 80 **1**.5 60 60 SDR (dB) SDR (dB) 20 20 10³ 10³ Frequency (Hz) Frequency (Hz)

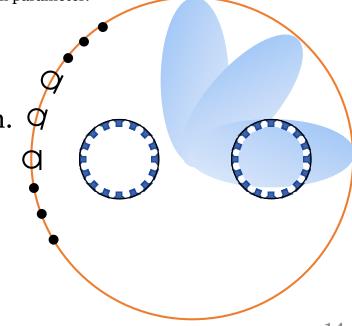
Minimum Variance Distortionless Response (MVDR)

$$\mathbf{w} = \frac{(\mathbf{R}^{-1} + \lambda \mathbf{I})\mathbf{C}^{H}}{\mathbf{C}(\mathbf{R}^{-1} + \lambda \mathbf{I})\mathbf{C}^{H}}\mathbf{f}$$

w:Beamforming filter; f: sound pressure at constraint points; C: transfer function for constraint points;

G:transfer function for suppress points; $\mathbf{R} \coloneqq \mathbf{G}^H \mathbf{G}$; λ : regularization parameter.

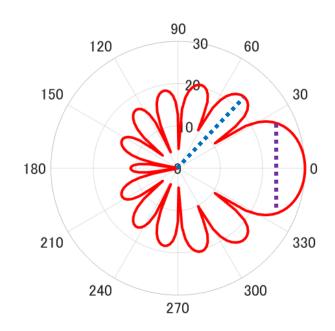
- Constraint point: one at look direction.
- Frequency domain design; f = 1.
- Constrained filter gain $W_0 = 0$ dB.
- Suppress point: other directions.



- Evaluation
 - DI(directivity index):

$$DI = 10 \log_{10} \frac{2\pi \|P_{\varphi}\|^{2}}{\int_{0}^{2\pi} \|P_{\varphi}\|^{2} d\varphi}$$

- Power of the look direction
- BW(beam width):
 - Half power beam width of the main lobe
 - The narrowness of the main beam
- SLL(side lobe level):
 - The maximum level of the side lobe
 - Relative to the main lobe



Simulation conditions

Fixed distance r_D :

$$r_0 = \begin{cases} 0.075 \text{ m} \\ 0.15 \text{ m} \\ 0.3 \text{ m} \end{cases}, r_D = 1 \text{ m}$$

Fixed radius r_0 :

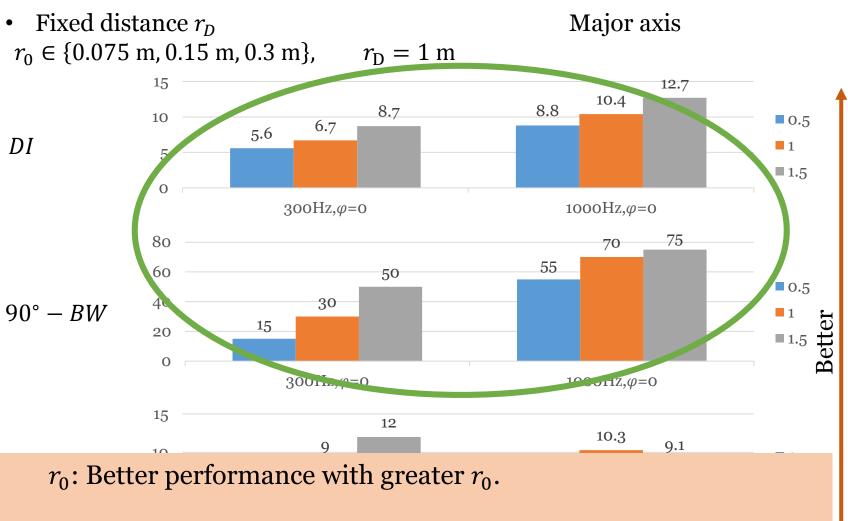
$$r_0 = 0.15 \text{ m}, r_D = \begin{cases} 0.5 \text{ m} \\ 1 \text{ m} \\ 1.5 \text{ m} \end{cases}$$

Look direction:

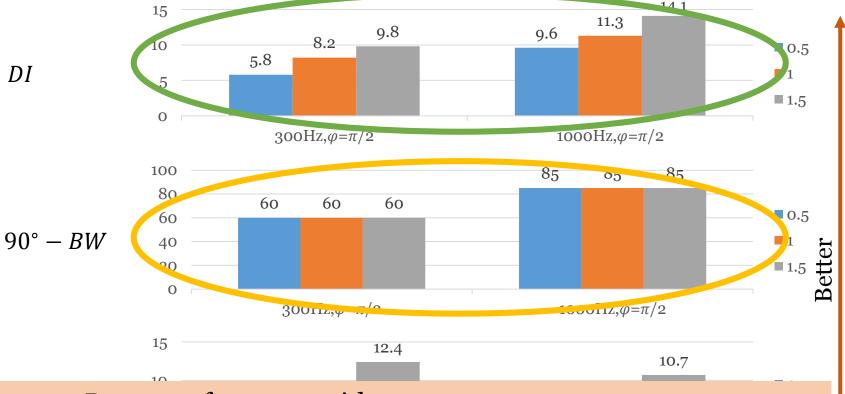
On major axis: $\varphi = 0$; on minor axis: $\varphi = \pi/2$

Frequencies:

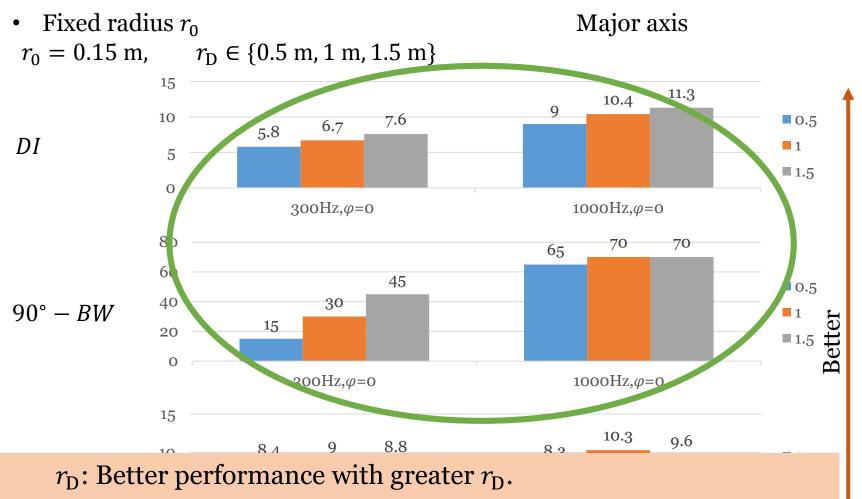
300 Hz, 1000 Hz



• Fixed distance r_D Minor axis $r_0 \in \{0.075 \text{ m}, 0.15 \text{ m}, 0.3 \text{ m}\}, r_D = 1 \text{ m}$

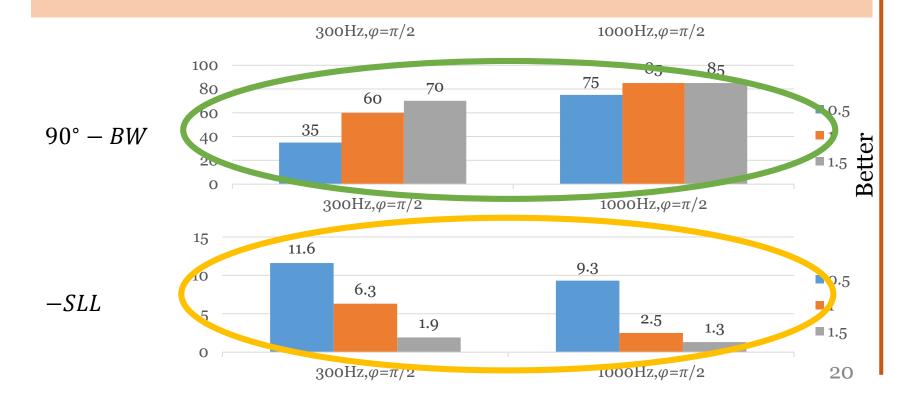


 r_0 : Better performance with greater r_0 . Doesn't affect beam width.



• Fixed radius r_0 $r_0 = 0.15 \text{ m}, r_D \in \{0.5 \text{ m}, 1 \text{ m}, 1.5 \text{ m}\}$ Minor axis

 $r_{\rm D}$: Better BW but worse SLL with greater $r_{\rm D}$. Greater $r_{\rm D}$ may lead to aliasings.



Conclusion

- r_0 and r_D do affect the performance of the 2CLA
 - Greater r_0 is supposed to get better performance when filter gain constrained
 - Greater r_D is better for controlling the sound on the major axis side while may lead to aliasing for the minor axis side

- Further works
 - Different $r_{0,(1)}$ and $r_{0,(2)}$, other methods, etc.
 - Reasons