

Comparative Performance Evaluation of Near 3D Sound Field Reproduction System with Directional Loudspeakers and Wave Field Synthesis

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Outline

- Introduction
 - Ultra-realistic communication
 - Wave field synthesis system
- Diagram of proposed system
- Developed system
- Evaluation of developed system
 - Computer simulation
 - Acoustical measurement
- Conclusion

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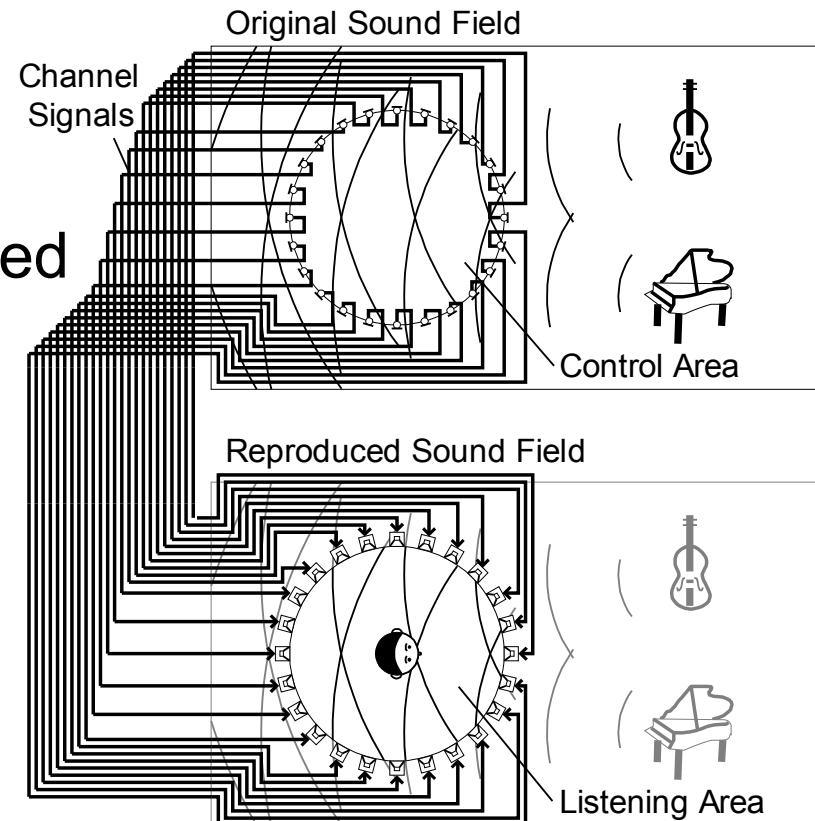
Ultra-Realistic Communication

- Future 3D television
 - 3D video and audio appear in a 3D space
 - People view an object anywhere in its vicinity
 - Without glasses
 - 3D sound field reproduction systems without headphones must be developed



Wave Field Synthesis (WFS) System

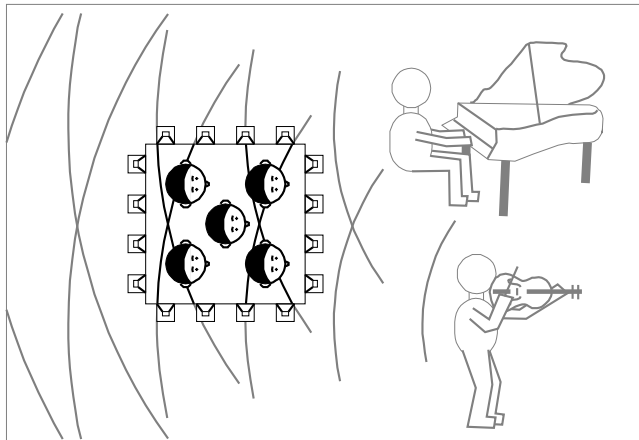
- Original sound field
 - Sound is recorded by the microphone array
- Reproduced sound field
 - Recorded sound is played by the loudspeaker array
 - 3D sound field is reproduced by Kirchhoff-Helmholtz integral equation
- Multiple listeners can listen to a sound without headphones



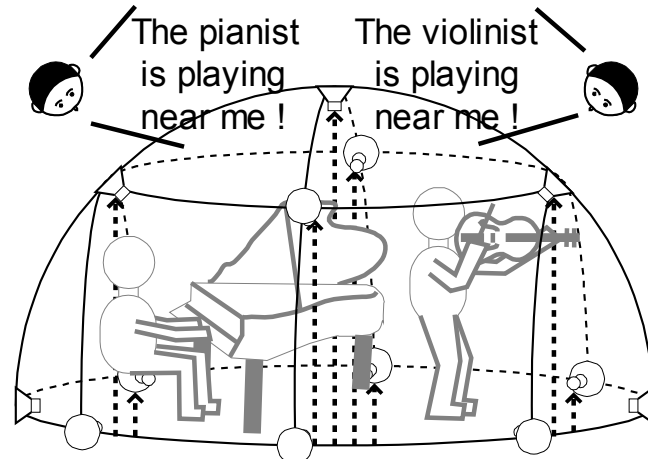
WFS System for Future 3DTV

- Conventional system
 - Loudspeakers are placed around the listeners
 - Sound scene is reproduced
- Proposed system
 - Loudspeakers are placed around sound sources
 - Radiated sound field is reproduced

Conventional System



Proposed System



Aim of Study

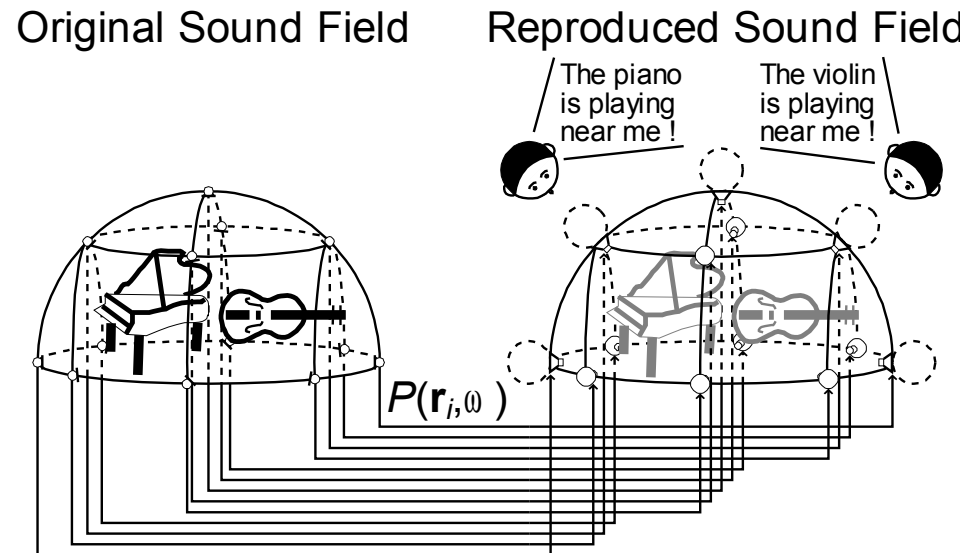
- We have proposed and developed the WFS system for future 3DTV
 - Near 3D sound field reproduction system using directional loudspeakers and wave field synthesis
- Evaluation of the performance of the sound image localization in the developed system
 - Estimation of the position of sound images
 - Computer simulation
 - Acoustical measurement

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Diagram of Proposed System

- Original sound field
 - Sound is recorded by microphones placed around sound sources
- Reproduced sound field
 - Recorded sound is played by directional loudspeakers
 - 3D sound field is reproduced
- Listeners feel that sound sources are playing in the array



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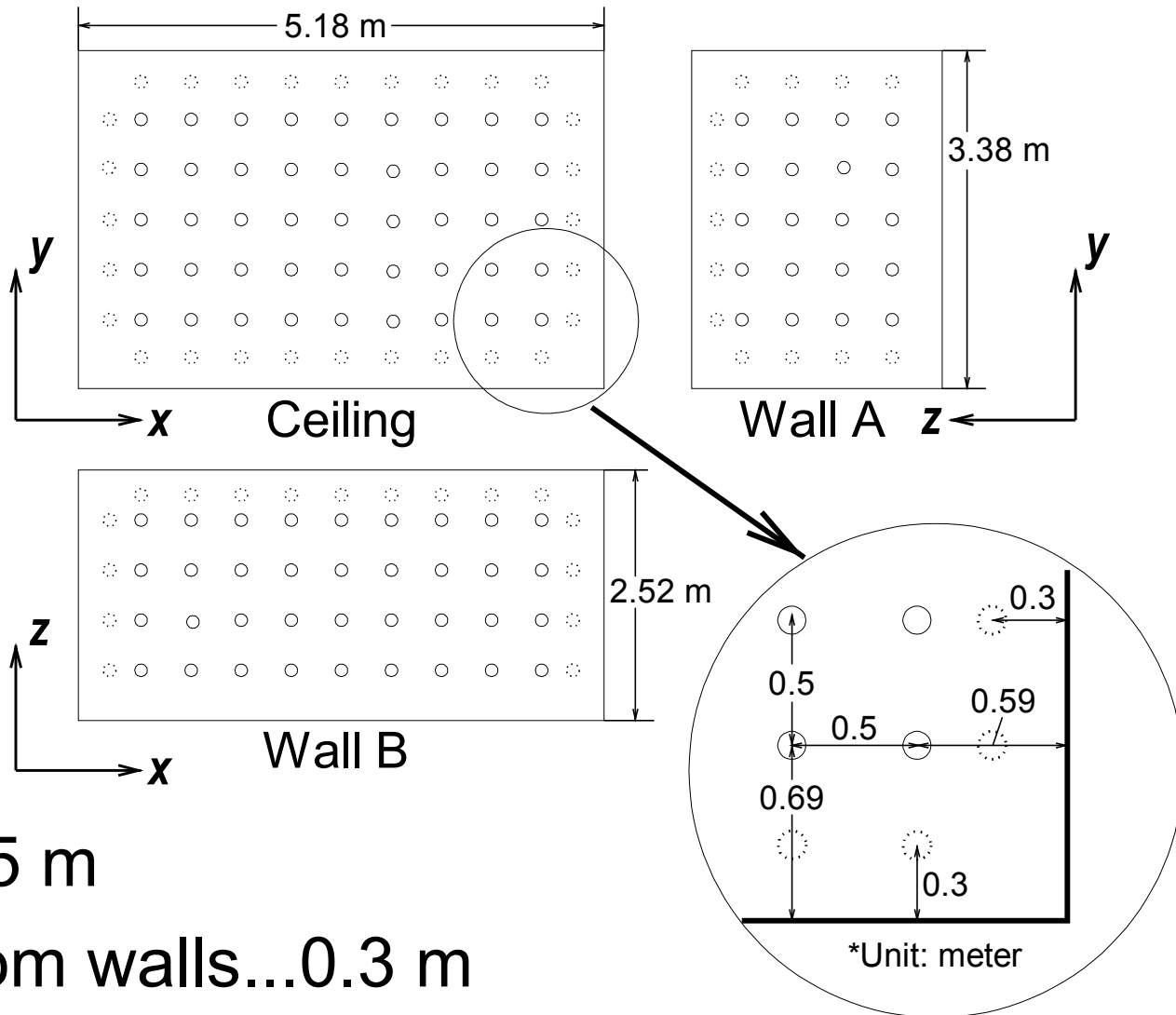
Surrounding Microphone Array

- Surrounding microphone array room
 - Equipped in RIEC, Tohoku Univ.
 - Reverberation time...about 150 ms
- 157 omnidirectional microphones
 - B&K: Type 4951
 - Placed on 5 planes
 - Mainly record the direct sound from sound sources
- 10 Amplifiers
 - B&K: Type 2694



Arrangement of Microphones

- Wall A
 - 2 planes
 - 20 (=5×4)
- Wall B
 - 2 planes
 - 36 (=9×4)
- Ceiling
 - 1 plane
 - 45 (=9×5)
- Interval...0.5 m
- Distance from walls...0.3 m



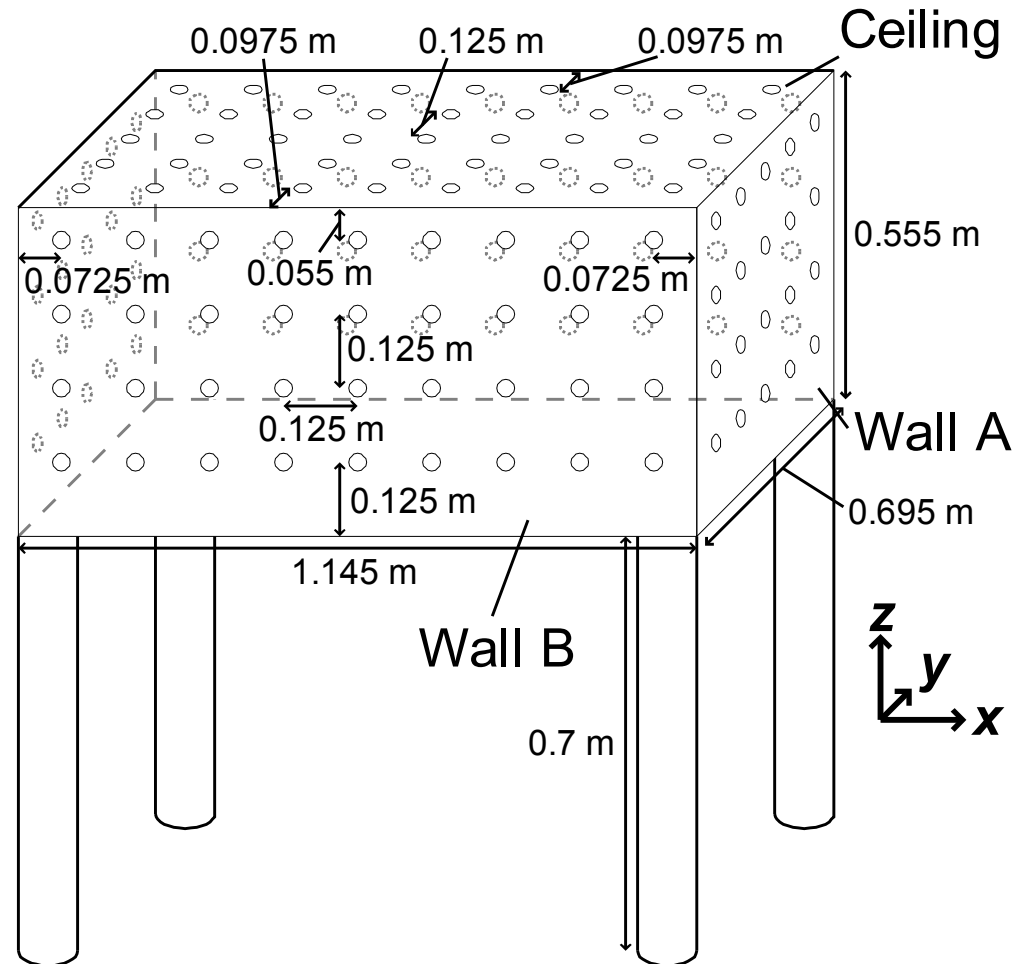
Radiated Loudspeaker Array

- Rectangular enclosure
 - Size...1/4 of surrounding microphone array
 - Material...Plywood and aluminum panels
- 157 loudspeaker units
 - AURASOUND:
 - NSW1-205-8A suitable
 - Attached to 5 planes
 - Size...1 inch
 - Directivity
 - Towards outside
- Amplifier
 - Custom-made (157ch)



Arrangement of Loudspeaker Units

- Interval...0.125 m
- Wall A
 - 2 planes
 - 20 (=5×4)
- Wall B
 - 2 planes
 - 36 (=9×4)
- Ceiling
 - 1 plane
 - 45 (=9×5)
- Elevated by 0.7 m



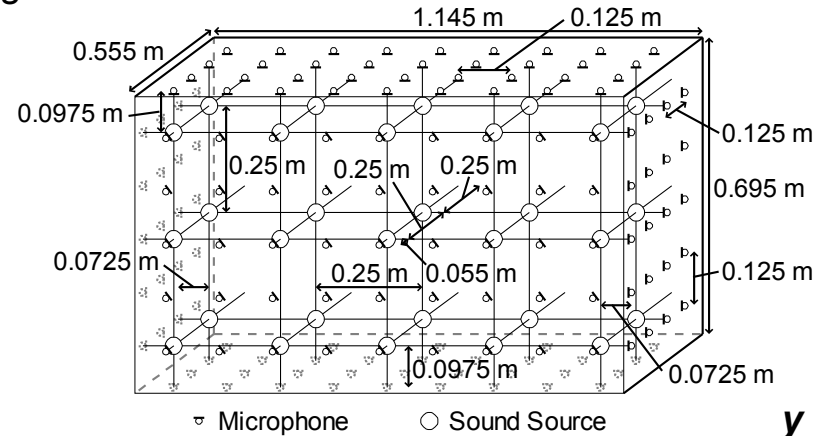
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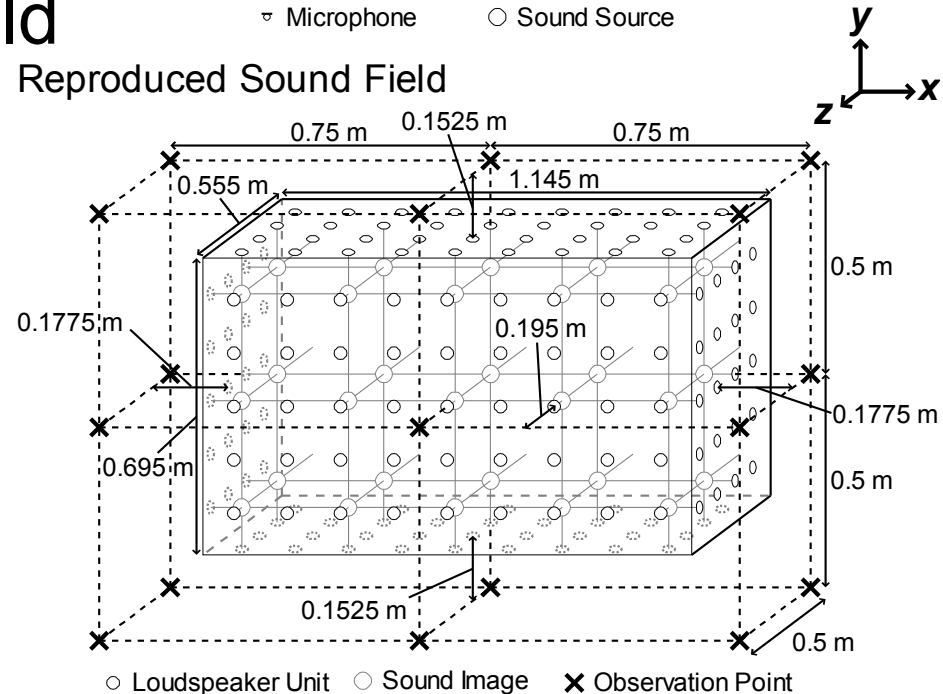
Simulation Environment

- Original sound field
 - Free space
 - 157 microphones
 - 30 sound sources
- Reproduced sound field
 - Free space
 - 157 loudspeaker units
 - 17 Observation points

Original Sound Field



Reproduced Sound Field



Synthesis of Recorded Signals

- Sound source signal $s(t)$
 - Octave-band noise (central frequency f_{cent})
- Signal of i th microphone $x_i(t)$

$$x_i(t) = g_i(t) * s(t) = \frac{1}{|\mathbf{r}_i - \mathbf{r}_0|} s\left(t - \frac{|\mathbf{r}_i - \mathbf{r}_0|}{c}\right)$$

- *: Convolution operation
- \mathbf{r}_i : Position vector of i th microphone
- \mathbf{r}_0 : Position vector of sound sources
- c : Sound velocity

Synthesis of Sound Pressure

- Sound pressure in observation points

$$p(\mathbf{R}_j, f_{\text{cent}}, t)$$

$$\begin{aligned} p(\mathbf{R}_j, f_{\text{cent}}, t) &= \sum_{i=1}^M \frac{D_{si}}{|\mathbf{R}_j - \mathbf{r}_i|} x_i \left(t - \frac{|\mathbf{R}_j - \mathbf{r}_i|}{c} \right) \\ &= \sum_{i=1}^M \frac{D_{si}}{|\mathbf{R}_j - \mathbf{r}_i| |\mathbf{r}_i - \mathbf{r}_0|} s \left(t - \frac{|\mathbf{R}_j - \mathbf{r}_i| + |\mathbf{r}_i - \mathbf{r}_0|}{c} \right) \end{aligned}$$

- \mathbf{R}_j : Position vector of j th observation point
- M : Total number of loudspeaker units
- D_{si} : Radiation directivity of i th loudspeaker unit

Calculation of Sound Intensity

- Direction of sound intensity vector
 - Corresponds to arrival direction of sound sources

$$\mathbf{I}(\mathbf{R}_j, f_{\text{cent}}) = \left\{ I_x(\mathbf{R}_j, f_{\text{cent}}), I_y(\mathbf{R}_j, f_{\text{cent}}), I_z(\mathbf{R}_j, f_{\text{cent}}) \right\}^T$$

- Calculation by cross-spectral method
 - 3 directions are calculated from sound pressures at 6 points

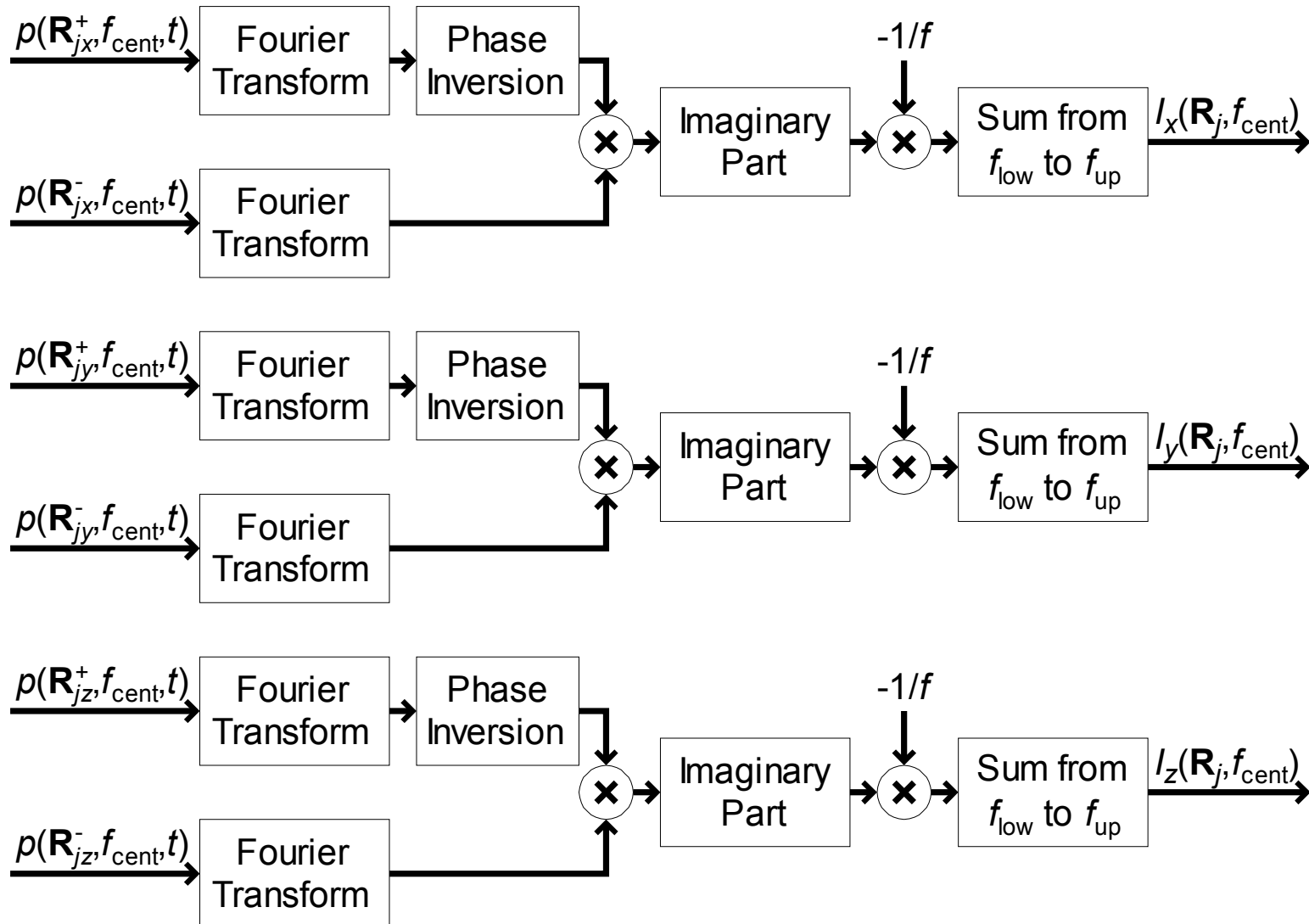
- Sound pressure: $p(\mathbf{R}_{jx}^+, f_{\text{cent}}, t)$, $p(\mathbf{R}_{jx}^-, f_{\text{cent}}, t)$, $p(\mathbf{R}_{jy}^+, f_{\text{cent}}, t)$,
 $p(\mathbf{R}_{jy}^-, f_{\text{cent}}, t)$, $p(\mathbf{R}_{jz}^+, f_{\text{cent}}, t)$, $p(\mathbf{R}_{jz}^-, f_{\text{cent}}, t)$,

$$\mathbf{R}_{jx}^{\pm} = \mathbf{R}_j \pm (\Delta, 0, 0)^T$$

$$\mathbf{R}_{jy}^{\pm} = \mathbf{R}_j \pm (0, \Delta, 0)^T \quad \Delta = 0.001 \text{ m}$$

$$\mathbf{R}_{jz}^{\pm} = \mathbf{R}_j \pm (0, 0, \Delta)^T$$

Block Diagram of SI Calculation

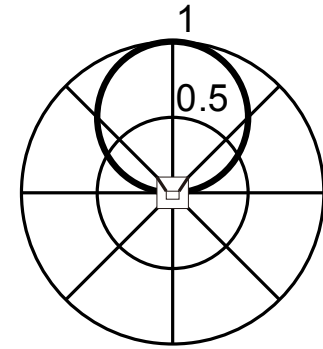
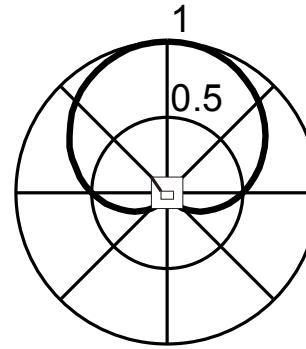
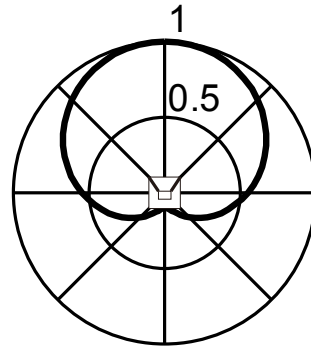
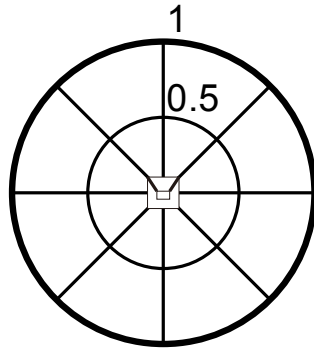


Parametric Conditions

Central frequency (f_{cent})	250, 500, 1000 Hz
Lower frequency (f_{low})	$f_{\text{cent}} \div \text{sqrt}(2)$
Upper frequency (f_{up})	$f_{\text{cent}} \times \text{sqrt}(2)$
Sound velocity(c)	340 m/s
Number of loudspeaker units (M)	157
Radiation directivity of loudspeaker units (D_{si})	Omnidirectional, Decay20dB, Unidirectional, Shotgun

Radiation Directivity of Loudspeakers

Omnidirectional Decay 20dB Unidirectional Shotgun



(Omnidirectional)

$$D_{si} = 1$$

(Decay 20dB)

$$D_{si} = 0.55 + 0.45 \cos \theta_{si}$$

(Unidirectional)

$$D_{si} = \frac{1 + \cos \theta_{si}}{2}$$

$$\cos \theta_{si} = \frac{\mathbf{n}_{si} \cdot (\mathbf{R}_j - \mathbf{r}_i)}{|\mathbf{n}_{si}| |\mathbf{R}_j - \mathbf{r}_i|}$$

(Shotgun)

$$D_{si} = \begin{cases} \cos \theta_{si} & (\theta_{si} \leq 90^\circ) \\ 0 & (\theta_{si} > 90^\circ) \end{cases}$$

Estimation of Sound Image Position

- Estimated position of sound images \mathbf{r}_E
$$\mathbf{r}_E = \frac{1}{FN} \sum_{f_{\text{cent}} \in \{250, 500, 1000\}} \sum_{j=1}^N \left\{ \mathbf{R}_j - \frac{\mathbf{I}(\mathbf{R}_j, f_{\text{cent}})}{\rho(\mathbf{R}_j, f_{\text{cent}})} \right\}$$

– \mathbf{R}_j : Position vector of j th observation point

– $\mathbf{I}(\mathbf{R}_j, f_{\text{cent}})$: Sound intensity at \mathbf{R}_j

– $\rho(\mathbf{R}_j, f_{\text{cent}})$: RMS of sound pressure at \mathbf{R}_j

$$\rho(\mathbf{R}_j, f_{\text{cent}}) = \sqrt{\frac{1}{T} \int_0^T \{ p(\mathbf{R}_j, f_{\text{cent}}, t) \}^2 dt}$$

– T : Period

– $F(=3)$: Number of octave-band noises

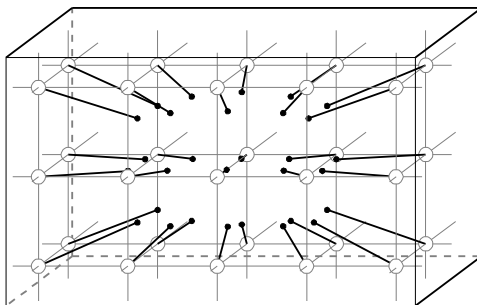
– N : Number of observation points

- All observation points, 4 observation points near source

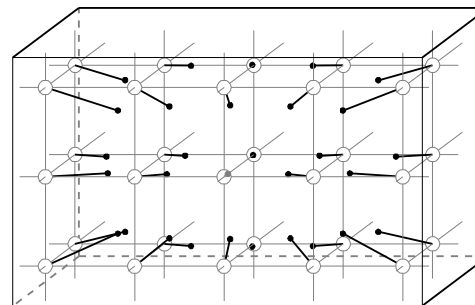
Estimated Position (All Points)

- If radiation directivity is sharpened
 - Sound images are accurately estimated
 - Listeners can accurately localize sound images at any listening position around loudspeaker array

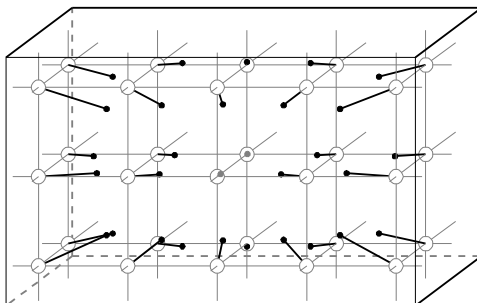
Omnidirectional



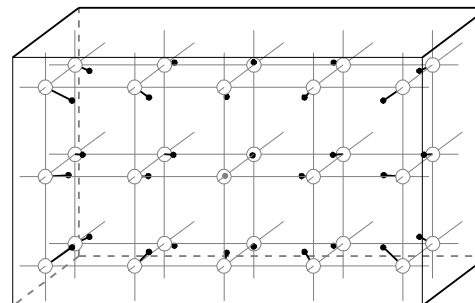
Decay 20dB



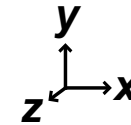
Unidirectional



Shotgun



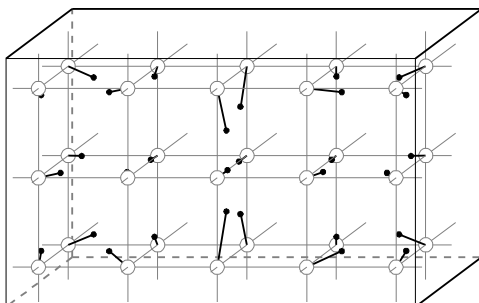
- Input Sound Image
- Estimated Sound Image



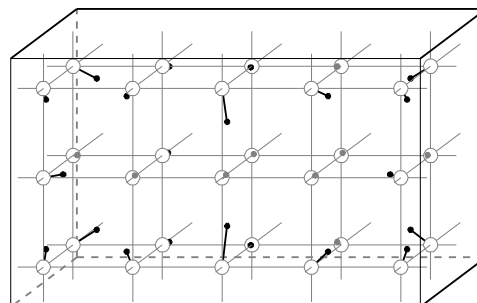
Estimated Position (4 Points Near Source)

- Even if radiation directivity is not sharpened
 - Sound images are approximately estimated
 - Listeners can accurately localize sound image close to listening position

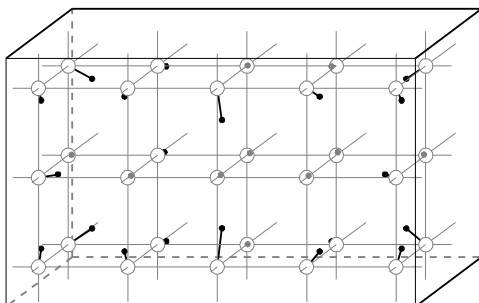
Omnidirectional



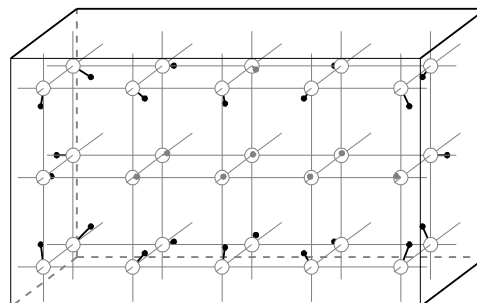
Decay 20dB



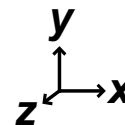
Unidirectional



Shotgun



- Input Sound Image
- Estimated Sound Image

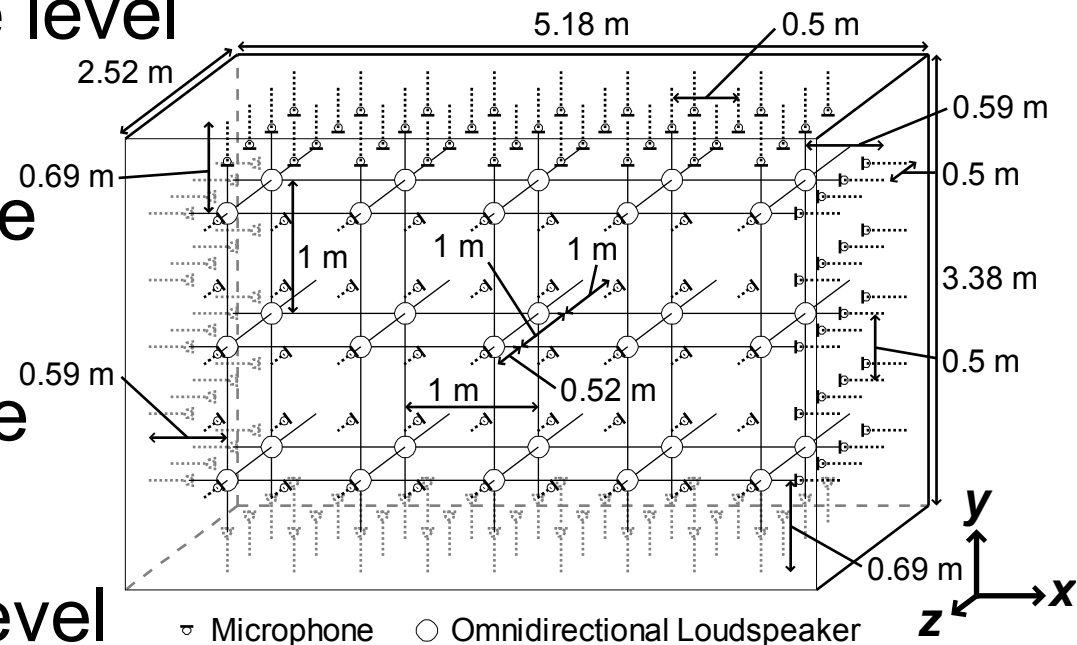


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Room Impulse Response Measurement

- Omnidirectional loudspeakers are placed at 30 positions in the surrounding microphone array room
- Background noise level
 - 18.4 dB(A)
- Reverberation time
 - 150 ms
- Room temperature
 - 20°C
- Sound pressure level
 - 85.6 dB(A) at 1m distance from loudspeakers



RIR Measurement Conditions

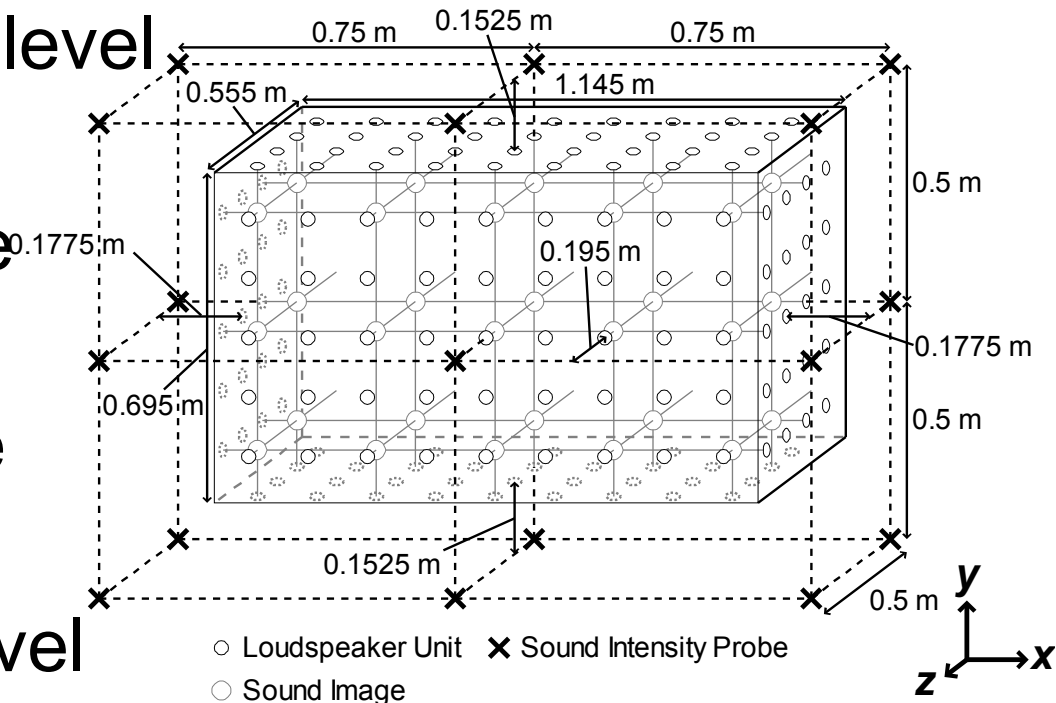
- TSP signal
 - Sampling frequency...48 kHz
 - Quantization bits...16 bits
 - Length...65536 samples
- FIR filter
 - Length...14400 taps
 - Number of synchronous repetitions...16
 - Sampling frequency is regarded as 192 kHz
 - match the size of surrounding microphone array to that of radiated loudspeaker array
 - Reverberation time...about 37.5 ms (1/4 of 150 ms)

Synthesis of 157 Channel Signals

- Convolve measured room impulse responses to sound source signals
- Sound source signal
 - Octave-band noise
 - Central frequency...250, 500, 1000 Hz
 - Sampling frequency...48 kHz
 - Quantization bits...16 bits
 - Length...10 s
 - Time of fade-in and fade-out...1 ms
- Room Impulse response
 - Transformed to 48 kHz before synthesis

Measurement of Sound Intensity

- Synthesized 157ch signals are played by the radiated loudspeaker array and sound intensity probes are placed at 17 positions
- Background noise level
 - 22 dB(A)
- Reverberation time
 - 180 ms
- Room temperature
 - 22°C
- Sound pressure level
 - 71 dB(A) at 1 m from central sound image



Estimation of Sound Image Position

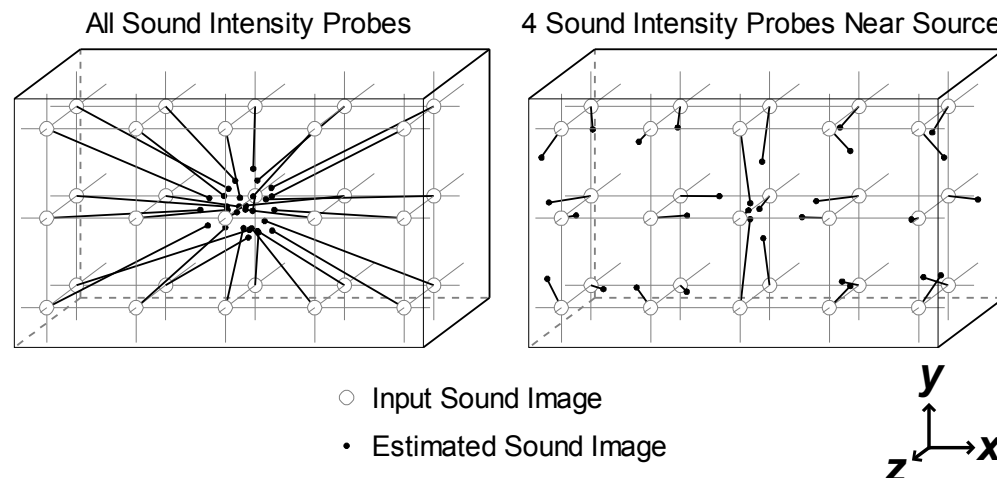
- Estimated sound image position \mathbf{r}_E

$$\mathbf{r}_E = \frac{1}{FN} \sum_{f_{\text{cent}} \in \{250, 500, 1000\}} \sum_{j=1}^N \left\{ \mathbf{R}_j - \frac{\mathbf{I}(\mathbf{R}_j, f_{\text{cent}})}{\rho(\mathbf{R}_j, f_{\text{cent}})} \right\}$$

- \mathbf{R}_j : Position vector of j th sound intensity probe
- $\mathbf{I}(\mathbf{R}_j, f_{\text{cent}})$: Sound intensity at \mathbf{R}_j
- $\rho(\mathbf{R}_j, f_{\text{cent}})$: Sound pressure at \mathbf{R}_j
- $F(=3)$: Number of octave-band noises
- N : Number of sound intensity probes
 - All sound intensity probes
 - 4 sound intensity probes near source

Estimated Sound Image Position

- All sound intensity probes
 - Estimated positions are biased to the center
- 4 sound intensity probes near source
 - Sound images are approximately estimated
- Listeners can accurately localize sound image close to listening position



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Conclusion

- Evaluation of the performance of the sound image localization in the developed system
 - The positions of sound images were estimated by computer simulation and acoustical measurement
 - If the radiation directivity is sharpened, listeners can accurately localize sound images in any listening position
 - Even if the radiation directivity is not sharpened, listeners can accurately localize the sound image close to the listening position
- Future works
 - Development of a system with better performance in a real environment by sharpening the radiation directivity of the loudspeaker units