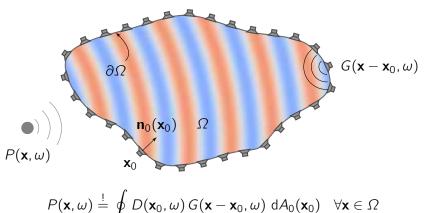


Time-Domain Realisation of Model-Based Rendering for 2.5D Local Wave Field Synthesis Using Spatial Bandwidth-Limitation

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Sound Field Synthesis



$$\underbrace{P(\mathbf{x},\omega)}_{\substack{\text{desired} \\ \text{sound field}}} \stackrel{!}{=} \oint \underbrace{D(\mathbf{x}_0,\omega)}_{\substack{\text{driving signal}}} \underbrace{G(\mathbf{x}-\mathbf{x}_0,\omega)}_{\substack{\text{sound field of} \\ \text{loudspeaker at } \mathbf{x}_0}} dA_0(\mathbf{x}_0) \quad \forall \mathbf{x} \in \Omega$$

Sound Field Synthesis

Fundamental Principles of Rendering

Model-Based

- uses mathematical models for virtual sources which are fed by (dry) source signals
- acoustic scene is typically composed of multiple virtual sources
- frequently used models are point sources and plane waves

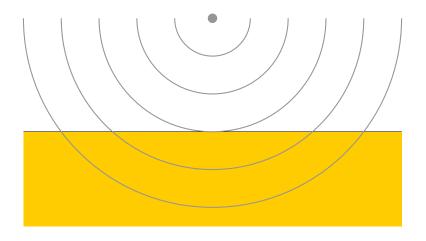
$$P_{\rm ps}(\mathbf{x},\omega) = S(\omega) \frac{\mathrm{e}^{-\mathrm{j}\frac{\omega}{\mathrm{c}}|\mathbf{x} - \mathbf{x}_{\rm ps}|}}{4\pi|\mathbf{x} - \mathbf{x}_{\rm ps}|} \qquad P_{\rm pw}(\mathbf{x},\omega) = S(\omega)\mathrm{e}^{-\mathrm{j}\frac{\omega}{\mathrm{c}}n_{\rm pw}^{\mathsf{T}}\mathbf{x}}$$

Data-Based

reproduces a scene acquired via Sound Field Analysis techniques

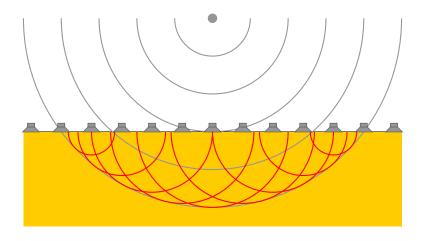
Wave Field Synthesis

Huygens-Fresnel principle



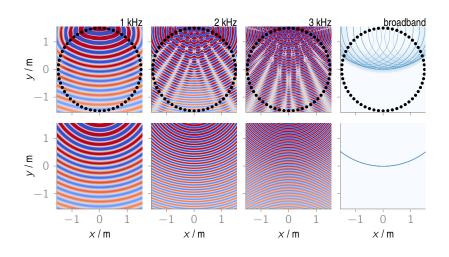
Wave Field Synthesis

Huygens-Fresnel principle

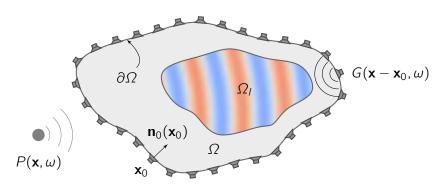


Wave Field Synthesis

Spatial Aliasing



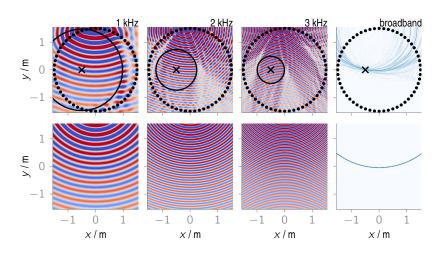
Local Sound Field Synthesis



$$\underbrace{P(\mathbf{x},\omega)}_{\text{desired}} \stackrel{!}{=} \oint\limits_{\partial \Omega} \underbrace{D(\mathbf{x}_0,\omega)}_{\text{driving signal}} \underbrace{G(\mathbf{x}-\mathbf{x}_0,\omega)}_{\text{sound field of loudspeaker at } \times_0} \mathrm{d} A_0(\mathbf{x}_0) \quad \forall \mathbf{x} \in \Omega_I$$

Local Wave Field Synthesis (LWFS) using Spatial Bandwidth Limitation (SBL)

Spatial Aliasing



LWFS-SBL

Driving Signal I

Truncated Circular Harmonics Expansion around x_c

$$P(\mathbf{x},\omega) \approx S(\omega) \sum_{\mu=-M}^{M} \check{P}_{\mu}(\mathbf{x}_{c},\omega) J_{\mu}(k\rho') \mathrm{e}^{+\mathrm{j}\mu\phi'}$$

Conversion between Representations
$$\sqrt{\bar{P}}(\phi_{\rm pw},{\bf x}_{\rm c},\omega) = \sum_{\mu=-M}^{M} {\rm j}^{\mu} \check{P}_{\mu}({\bf x}_{\rm c},\omega) {\rm e}^{+{\rm j}\mu\phi_{\rm pw}}$$

2D Plane Wave Decomposition around x_c

$$P(\mathbf{x}, \omega) \approx S(\omega) \frac{1}{2\pi} \int_{-\pi}^{\pi} \bar{P}(\phi_{\mathrm{pw}}, \mathbf{x}_{\mathrm{c}}, \omega) \mathrm{e}^{-\mathrm{j}k\langle x' | n_{\mathrm{pw}} \rangle} \, \mathrm{d}\phi_{\mathrm{pw}}$$

Apply conventional WFS driving | function to each plane wave

LWFS Driving Signal in Time-Frequency Domain

$$D^{\text{LWFS}}(\mathbf{x}_0, \omega) = S(\omega) \frac{1}{2\pi} \int_{-\pi}^{\pi} \bar{P}(\phi_{\text{pw}}, \mathbf{x}_{\text{c}}, \omega) D_{\text{pw}}^{\text{WFS}}(\mathbf{x}'_0, \phi_{\text{pw}}, \omega) d\phi_{\text{pw}}$$

LWFS-SBL

Driving Signal II

LWFS Driving Signal in Continuous Time Domain

$$d^{\text{LWFS}}(\mathbf{x}_0,t) = s(t) *_t \frac{1}{2\pi} \int_{-\pi}^{\pi} \bar{p}(\phi_{\text{pw}},\mathbf{x}_{\text{c}},t) *_t d_{\text{pw}}^{\text{WFS}}(\mathbf{x}_0',\phi_{\text{pw}},t) \, \mathrm{d}\phi_{\text{pw}}$$

Temporal Sampling
$$f(t) \rightarrow f(nT_s) \rightarrow f[n]$$

LWFS Driving Signal in Discrete Time Domain

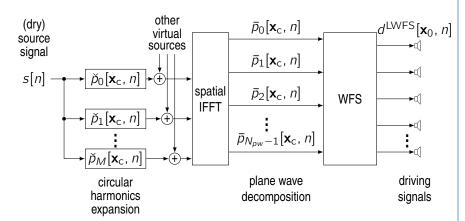
$$d^{\text{LWFS}}[\mathbf{x}_{0}, n] = s[n] *_{n} \frac{1}{2\pi} \int_{-\pi}^{\pi} \bar{p}[\phi_{\text{pw}}, \mathbf{x}_{\text{c}}, n] *_{n} d_{\text{pw}}^{\text{WFS}}[\mathbf{x}'_{0}, \phi_{\text{pw}}, n] d\phi_{\text{pw}}$$

Rectangle Method with equi-angular Sampling
$$\int f(\phi_{\mathrm{pw}}) o f\left(rac{2\pi m}{N_{\mathrm{pw}}}
ight) o f_m$$

$$\begin{split} d^{\text{LWFS}}[\mathbf{x}_0, n] &\approx s[n] *_n \frac{1}{N_{\text{pw}}} \sum_{m=0}^{N_{\text{pw}}-1} \bar{p}_m[\mathbf{x}_{\text{c}}, n] *_n d_{\text{pw},m}^{\text{WFS}}[\mathbf{x}_0', n] \\ \text{with } \bar{p}_m[\mathbf{x}_{\text{c}}, n] &= \sum_{\mu=-M}^{M} j^{\mu} \check{p}_{\mu}[\mathbf{x}_{\text{c}}, n] \, \mathrm{e}^{+j\frac{2\pi}{N_{\text{pw}}}\mu m} \end{split}$$

LWFS-SBL

System Layout



! Open Task: Time-Realisation of Circular Harmonics Coefficients

Circular Harmonics Expansions

Plane Wave

$$\check{P}_{\mu}(\mathbf{x}_{c},\omega)=\mathsf{j}^{-\mu}\mathsf{e}^{-\mathsf{j}\mu\phi_{\mathsf{pw}}}$$

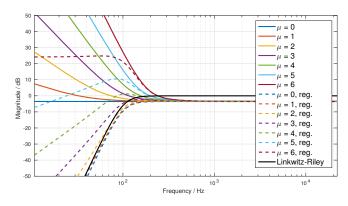
+ straightforward implementation possible

Point Source

$$\check{P}_{\mu}(\mathbf{x}_{\mathrm{c}},\omega) \approx \frac{\mathsf{j}^{|\mu|-\mu}}{4\pi} \left(-\mathsf{j}\frac{\omega}{\mathsf{c}}\right) h_{|\mu|}^{(2)} \left(\frac{\omega}{\mathsf{c}} \rho_{\mathrm{ps}}'\right) \mathrm{e}^{-\mathsf{j}\mu\phi_{\mathrm{ps}}'}$$

- + analytical expression in Laplace-Domain available → IIR implementation
- pole of order $|\mu|$ at $\omega=0$ \rightarrow unstable
- ! regularisation/pole compensation needed

Regularisation for Point Source



- ullet apply Linkwitz-Riley highpass filter of order $2\eta \geq M$ to all modes μ
- ! explicit combination of highpass filter and spherical Hankel functions necessary
- discrete-time realisation via bilinear transform
- results in strongly highpass filtered source signal

Dual-Band Approach for Point Source

- conventional WFS is aliasing-free and stable at low frequency
- combine conventional WFS and LWFS-SBL via Linkwitz-Riley frequency crossover

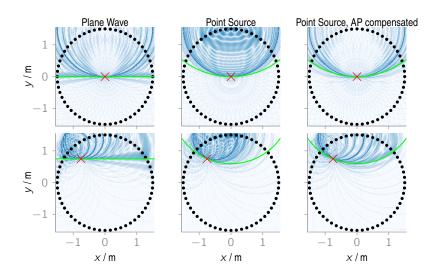
$$D_{ps}^{Dual}(\mathbf{x}_0, \omega) = D_{ps}^{WFS}(\mathbf{x}_0, \omega) L P^{2\eta}(\omega) + D_{ps}^{LWFS}(\mathbf{x}_0, \omega) H P^{2\eta}(\omega)$$

with

- $LP^{2\eta}(\omega)$ Linkwitz-Riley lowpass filter of order 2η
- $HP^{2\eta}(\omega)$ Linkwitz-Riley highpass filter of order 2η
- $AP^{2\eta}(\omega) = LP^{2\eta}(\omega) + HP^{2\eta}(\omega)$ joined allpass characteristic
- ullet allpass characteristic may be compensated using Backward-Filtering o non-causal

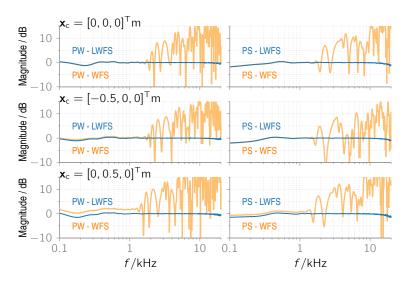
Simulations

Reproduced Sound Field, 56 Loudspeaker, M = 27



Simulations

Magnitude Spectrum, 56 Loudspeaker, M = 27



Conclusion & Future Work

Conclusion

- system layout for the time-domain realisation of LWFS-SBL
- existing WFS software (e.g. SoundScape Renderer) may be used to synthesise plane wave decomposition
- realisation of point source requires additional regularisation and crossover between conventional WFS and LWFS-SBL
- implementation available as part of the Sound Field Synthesis Toolbox

Future Work

- fair evaluation and comparison to other approaches w.r.t to run time^a
- perceptual evaluation^b

^aHahn et al., Synthesis of a Spatially Band-Limited Plane Wave in the Time-Domain Using Wave Field Synthesis, EUSIPCO 2017

^bWinter et al., Colouration in 2.5D Local Wave Field Synthesis Using Spatial Bandwidth-Limitation, WASPAA 2017

Thank you for your attention!



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