Overview

This paper presents a wideband image source **method** to simulate the time-domain signal on the boundary of the spherical listening region. The proposed method considers the loudspeaker directional impulse responses (DIRs).



Figure 1: Problem setup. The ISM simulates the time-domain signal on the yellow boundary of the spherical listening region.

Loudspeaker DIRs



Figure 2: Measurement setup. The loudspeaker DIRs are measured at sampling points on the light red spherical surface of radius $r^{(s)}$.

Loudspeaker DIRs are treated as a sequence of propagating spherical wave fronts with directiondependent amplitude.

Far-field DIRs measured on a spherical surface of radius $r^{(s)}$

$$h(t, r^{(s)}, \theta^{(s)}, \phi^{(s)}) = \int_{\tau} h(\tau, r^{(s)}, \theta^{(s)}, \phi^{(s)}) \delta(t - \tau) d\tau.$$
(1)

Replace with an ideal source at $O^{(s)}$ that emits

$$d(t,\theta^{(s)},\phi^{(s)}) = \int_{\tau} d(\tau,\theta^{(s)},\phi^{(s)})\delta(t-\tau)d\tau.$$

Let

 $d(\tau, \theta^{(s)}, \phi^{(s)}) = 4\pi r^{(s)} h(\tau + r^{(s)}/c, r^{(s)}, \theta^{(s)}, \phi^{(s)}).$ Assume each spherical wave front only experiences uniform attenuation related to the traveled distance, the DIRs of this ideal source measured on the spherical surface of radius $r^{(s)}$ should follow exactly (1).

Image Source Method Based On the Directional Impulse Responses

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$$d(au, heta^{(s)}, \phi^{(s)})$$

Using the superposition principle, the SH coefficients of the observed signal

$$\zeta_n^m(t_0, r) = \frac{c}{2rr_s} \int_{\tau} \sum_{v=0}^{V} \sum_{u=-v}^{v} \gamma_v^u(\tau) \, \mathcal{P}_v^u[\cos\theta_0^{(s)}(\tau)] \\ \mathcal{P}_n^m[\cos\theta_0(\tau)] \delta_{m,u} \, \Xi(t_0, \tau) \, d\tau.$$
(2)

•
$$\Xi(t_0,\tau) = 1$$
 if $r_s - r \le c(t_0 - \tau) \le r_s + r$; else, $\Xi(t_0,\tau) = 0$.

•
$$\cos \theta_0(\tau) = [r^2 + r_s^2 - c^2(t_0 - \tau)^2]/[2rr_s].$$

• $\cos \theta_0^{(s)}(\tau) = -[c^2(t_0 - \tau)^2 + r_s^2 - r^2]/[2c(t_0 - \tau)r_s].$

sources.

Step 3 - Calculate the SH coefficients of the observed signal w.r.t. the xyz coordinate system by following (2). Also incorporate the attenuation due to wall reflections.

Step 4 - Calculate the SH coefficients of the observed signal w.r.t. the $x^{(\rho')}y^{(\rho')}z^{(\rho')}$ coordinate system by using the Wigner D-matrix.

Step 5 - Add the contributions of all image

Simulations



In (a) and (b), $d(t, \theta^{(s')}, \phi^{(s')}) = Y_0^0(\theta^{(s')}, \phi^{(s')})\delta(t)$. In (c) and (d), $d(t, \theta^{(s')}, \phi^{(s')}) = Y_1^0(\theta^{(s')}, \phi^{(s')})\delta(t - 0.01)$. In (e) and (f), $d(t, \theta^{(s')}, \phi^{(s')}) = Y_0^0(\theta^{(s')}, \phi^{(s')})\delta(t) + Y_1^0(\theta^{(s')}, \phi^{(s')})\delta(t - 0.01).$ Moreover, (a), (c) and (e) are in anechoic condition; while (b), (d), and (f) are in reverberant condition.

References

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• Room dimension [4, 6, 3] m, wall reflection coefficients [0.45, 0.7, 0.8, 0.5, 0.6, 0.75]. • Source location $\mathbf{r}_{s}^{(\varepsilon')} = [1.5, 3.4, 2.4]$ m. • The observation sphere of radius r = 0.2 m is centered at $r^{(\varepsilon')} = [1.5, 3.4, 1]$ m. $\bullet d(t, \theta^{(s')}, \phi^{(s')}) =$ $Y_0^0(\theta^{(s')},\phi^{(s')})\delta(t) + Y_1^0(\theta^{(s')},\phi^{(s')})\delta(t-0.01).$

• 24 image sources are considered.

• The sampling frequency is 16 kHz.

• SH truncation order of the observed signal is 10.

To reduce the effect of aliasing, uniformly sampled version of (2) is convolved with a low-pass filter with 257 samples and cut-off frequency at 2 kHz. [3] and [4] cover more advanced sampling and bandlimitation methods.

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