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Real-time spatial audio on the Internet of Things

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Introduction

Spatially placed virtual sounds can create sounds in mid-air, or grant audio to objects that would otherwise be unable to produce sound, e.g., books, keys, utensils. Users could locate misplaced items through spatial audio guidance and/or allow users to interact with battery-less objects through auditory response.

In our envisioned future, the network of speakers on Internet-of-Things devices can provide a fabric of spatial audio throughout homes, offices, and public spaces.

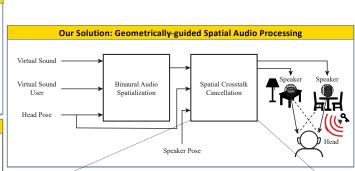
Challenge of Rendering Spatial Audio from Loudspeakers

Delivering spatial audio from loudspeakers is challenging due to crosstalk: the left sound signal

State-of-the-art crosstalk cancellation methods are expensive, requiring predesigned infrastructure and stationary observers.

- Ambisonics [1] and amplitude panning [2] require many surrounding speakers and require the user be in a specific fixed spot.
- Dynamic crosstalk cancellers for frequency domain audio processing have high latency and inaccuracy [3] or take significant calibration for the room and user transfer functions [4].

We propose a distributed spatial audio system that implements a timedomain dynamic crosstalk cancellation technique based on the geometry of a user's head position which uses estimations of amplitude decay and time delay to produce sound signals that become real-time binaural audio at the user's ears.



Spatial Cross-talk Cancellation

Due to crosstalk, each ear receives a combination of decayed (α) and delayed (δ) signals from speakers S₁ and S₂ to form left and right ear signals $E_{L}(t)$ and $E_{R}(t)$.

We use estimated α and δ information

to create S₁ and S₂ signals that, when combined at the ear, will present left and diptrimed to the servest left and diptrimed to the serv and right inputs $I_{L}(t)$ and $I_{R}(t)$ to the E_{L} and E_R respectively.

 $E_R(t) = \alpha_{1,R}S_1(t - \delta_{1,R}) + \alpha_{2,R}S_2(t - \delta_{2,R})$

 $E_L(t) = \alpha_{1,L}S_1(t - \delta_{1,L}) + \alpha_{2,L}S_2(t - \delta_{2,L})$

optimizing around predicted user movement (see below) **Sound Design**: What sounds work best for spatialization?

Future Exploration

Calibration: Can we retune system to work well in different environments?

Real-time speaker selection for optimized binaural perception, including

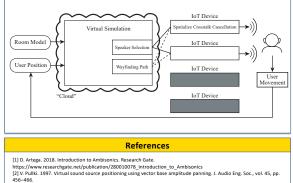
- Understanding efficacy of natural and synthesized sound patterns
- Applying spatial motion patterns to spatial sounds for localization

Integration: Can we provide real-time device-free tracking?

Gathering and applying room and speaker sound models

- Integrating with visual head tracking in a room
- Distributing audio-based computing across weak and strong IoT nodes

Cloud-based room model simulation for speaker selection



[3] H. Kurabayashi, M. Otani, K. Itoh, M. Hashimoto, M. Kayama, 2013, Development of dynamic transaural reproduction system using non-contact head tracking. In 2013 IEEE 2nd Global Conference on Consumer Electronics

(GCCE), (Tokyo, Japan) [4] M. Song, C. Zhang, D. Florencio, H. Kang. 2011. An Interactive 3-D Audio System with Loudspeakers. In IEEE

Transactions on Multimedia (Volume: 13, Issue: 5, Oct. 2011)