Implementation of real-time room auralization using a surrounding 157 loudspeaker array

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Abstract

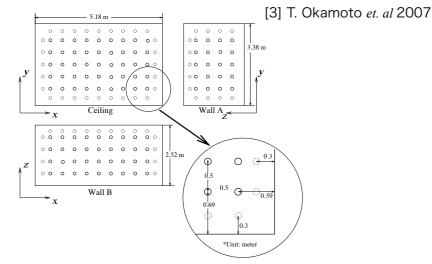
Numerous auralization systems have been developed for room acoustic simulations. Many of these systems use binaural rendering. This report presents the implementation of a real-time room acoustic auralization system using an array of 157 loudspeakers. The room acoustic model combines an iterative image-source model and feedback delay networks to create the early reflections and late reverberation. Higherorder Ambisonics (HOA) is used to generate spatial room impulse responses. A distributed network system is then used to generate the auralization output.

1 Introduction

- Auralization [1] has become a useful tool for the acoustic design of 3D architectural environments [1] M. Kleiner et. al 1993
- Most conventional techniques are computationally intensive and are therefore poorly suited to real-time applications
- M. Noisternig et. al have recently proposed a real-time system for auralization based on beam-tracing and binaural rendering [2]. [2] M. Noisternig et. al 2008

This poster presents the implementation of this auralization system on a surrounding 157 loudspeaker array, using a cluster of 5 PCs and the open source software EVERTims, VirChor and Pure Data.

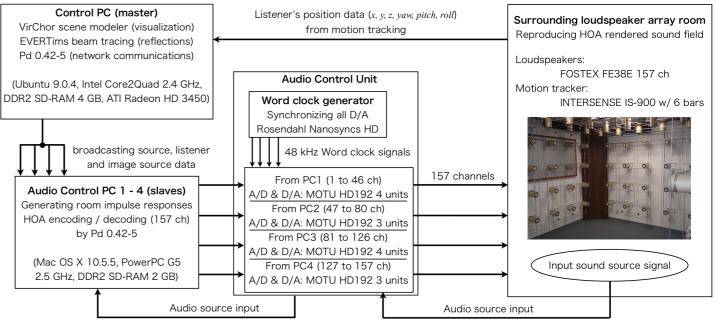
2 Surrounding 157 loudspeaker array [3]



Room dimensions and arrangement of loudspeakers

The measured mid-frequency reverberation time RT30 of the reproduction room is on the order of 0.2 s.





1. Master PC calculates the impulse responses at the listener position applying beam tracing for early reflections (EVERTims) [5] [4] J. B. Allen et. al 1979 [5] S. Laine et. al 2009 2. Client PCs calculate late reverberation by Feedback Delay Networks [6] and the 157 loudspeaker driving signals

based on Higher Order Ambisonics rendering [7] [6] J-M. Jot et. al 1991 [7] M. Poletti et. al 2005

4 Latency performance evaluation

1. All output signals of one PC are fully synchronized

2. Output signals between different PCs are not fully

synchronized with an average latency of 7.9 ms.

4.1 File playback synchronicity

ontrol signa

PC1 (Master)

controlling all PCs

ending control signal

playback TSP signal

A/D & D/A

1-46 ch

Results

(sample accurate)

4.2 Audio streaming synchronicity with splitter

PC3 (Slave)

controlled by PC1

A/D & D/A

81-126 ch

PC4 (Slave)

controlled by PC1

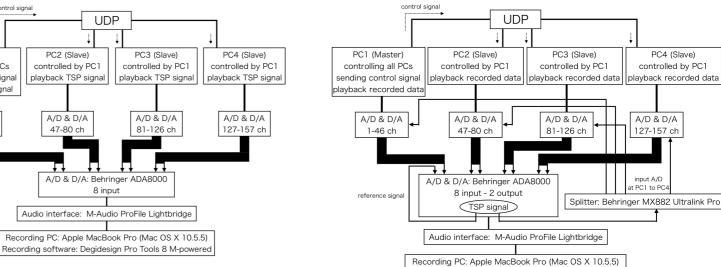
A/D & D/A

127-157 ch

input A/D

Splitter: Behringer MX882 Ultralink Pro

PC1 to PC



Results Recording software: Degidesign Pro Tools 8 M-powered

1. All output signals were completely synchronous !

- 2. The overall I/O system latency is constant 20 ms.
- 3. To further reduce the I/O latency real-time kernels and better CPU parallelization have to be applied

Acknowledgement: This study was supported by the GCOE program (CERIES) of the Grad. Sch. of Eng., Tohoku Univ.