

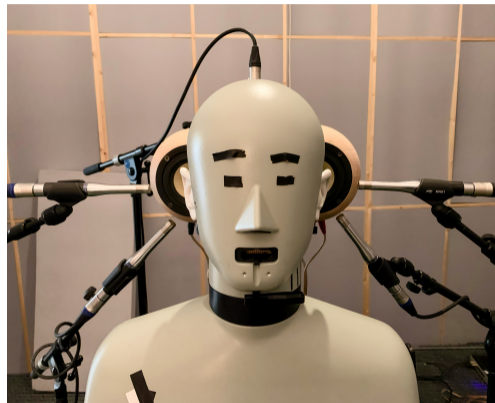
# Assessment of Simulations in FAUST and TASCAR for the Development of Audio Algorithms in Acoustic Environments

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# Introduction

- Research on active noise control (ANC) in time-variant environments
- ANC → hard real-time constraints
- Difficult to evaluate & validate algorithms
  - ▶ Offline evaluation often not possible
  - ▶ Evaluation on hardware (DSP/FPGA)



# Introduction

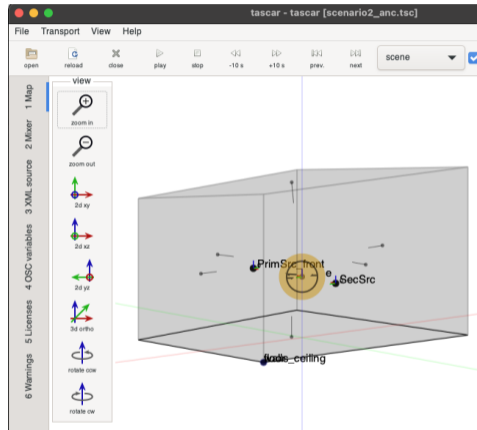
## Evaluation on hardware

- Bare metal programming
  - hardware-bound, time consuming, hard to adapt
- Conversion from higher language (Syfala/dSPACE/SpeedGoat)
  - costly, dedicated hardware
- General disadvantages
  - ▶ Hardware availability/costs
  - ▶ Limited repeatability
  - ▶ Measurement space

→ Real-time simulation of acoustic scenarios

# TASCAR

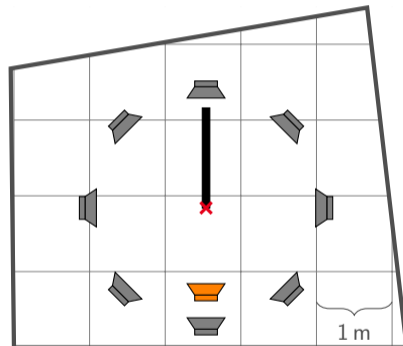
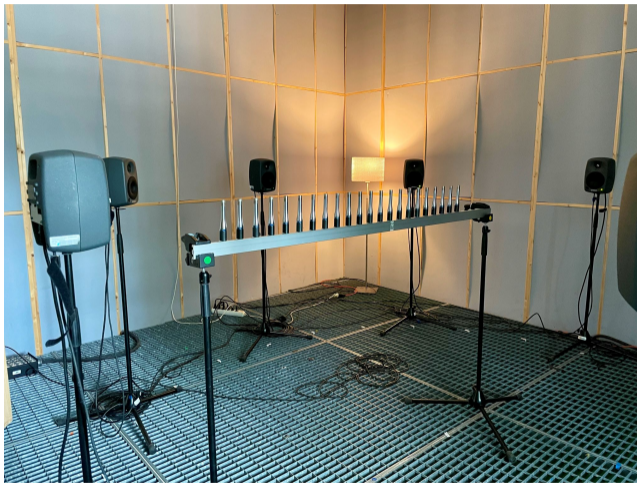
- “Toolbox for Acoustic Scene Creation and Rendering”
- Open-source audio virtualization tool (University of Oldenburg)
- Widespread use for hearing-aid research, audiology, and deep learning
- Offline & real-time sample-based auralization of acoustic scenes
- Using Jack Audio Connection Kit (I/O, time-line)
- Direct user-interaction via OSC



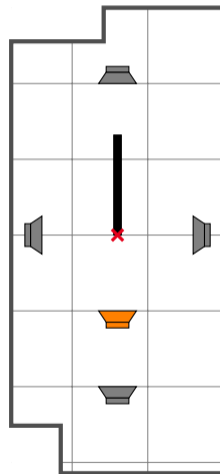
# Experiment Setup

- Compare accuracy of simulation to measurements/co-simulations
  - ▶ Soundfield properties
  - ▶ Behavior of acoustic algorithms
- Two different rooms
  - ▶ Semi-anaechoic measurement room
  - ▶ Anisotropic meeting room
- Different source configurations (1/4/8 sources)
- Reference recording with linear microphone arrangement

# Measurement setup - measurement chamber



# Measurement setup - meeting room



# Simulation setup

- Remodel simplified room model in TASCAR based on high-precision 3D scan
- Estimated absorption coefficients for each surface
- Varying simulation accuracy
  0. Free-field conditions
  1. 2<sup>nd</sup> order image source model (ISM)
  2. 2<sup>nd</sup> order ISM + modeled source directivity
  3. 2<sup>nd</sup> order ISM + directivity + FDN reverb
- Rendered offline (roomacoustic parameters) or online (RT signal processing)

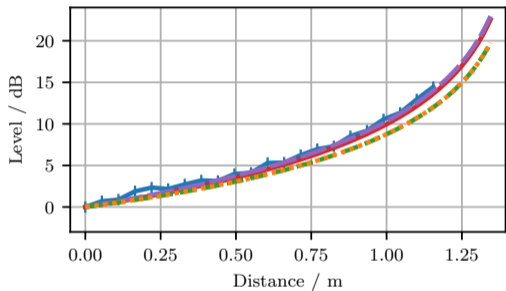


# Soundfield analysis

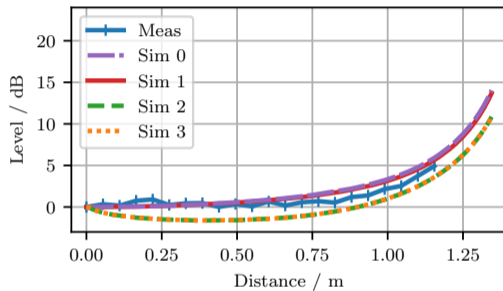
## Analyzed properties

- Level distribution along array
- Spatial coherence/correlation to central position
  - ▶ Spatial correlation widely used measure for description of sound fields
  - ▶ Correlation only valid for test signal → time consuming
  - ▶ Spatial coherence with broadband excitation for fast/easier measurement

# Level distribution - measurement room

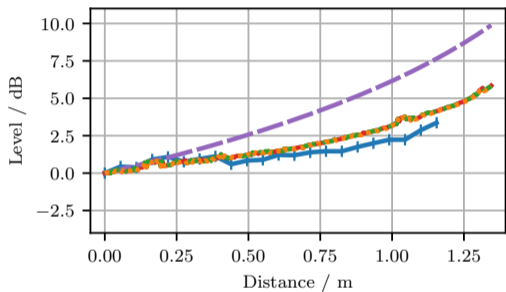


1 source

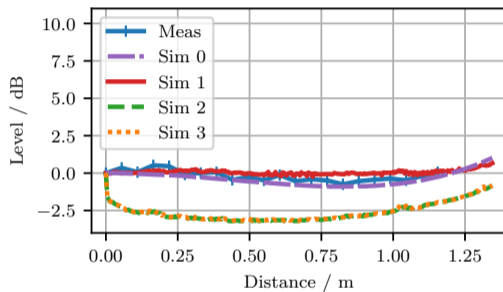


8 sources

# Level distribution - meeting room

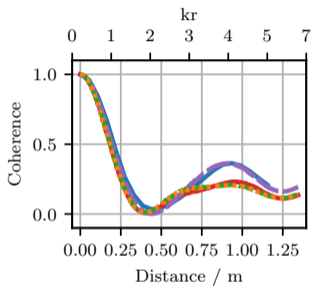


1 source

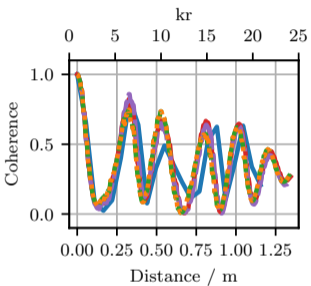


4 sources

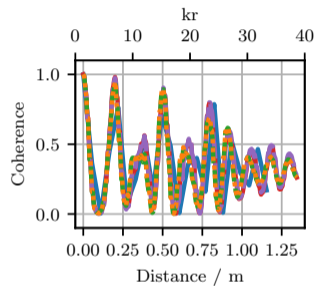
# Spatial coherence - measurement room (4ch)



$f = 300 \text{ Hz}$

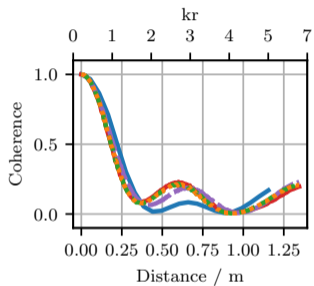


$f = 1000 \text{ Hz}$

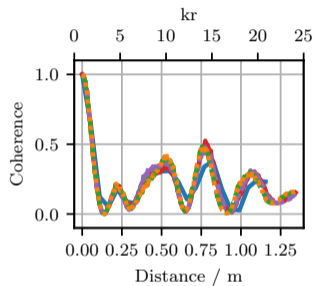


$f = 1700 \text{ Hz}$

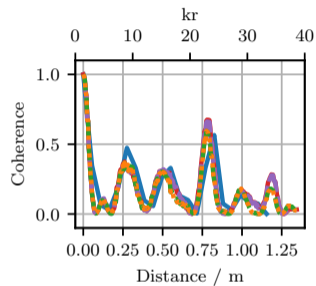
# Spatial coherence - measurement room (8ch)



$f = 300$  Hz

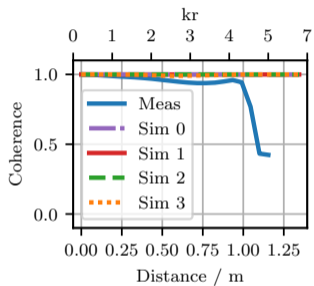


$f = 1000$  Hz

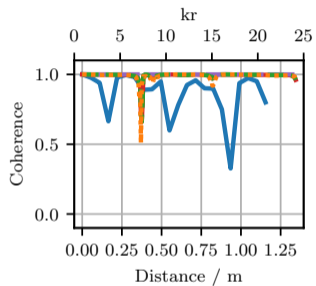


$f = 1700$  Hz

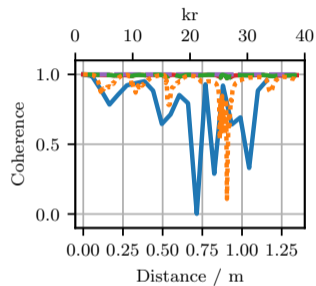
# Spatial coherence - meeting room (1ch)



$f = 300$  Hz

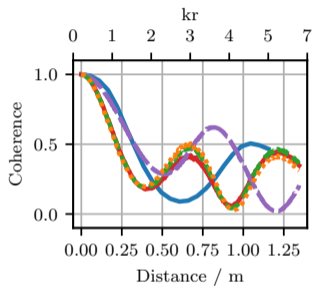


$f = 1000$  Hz

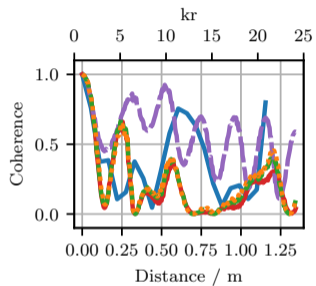


$f = 1700$  Hz

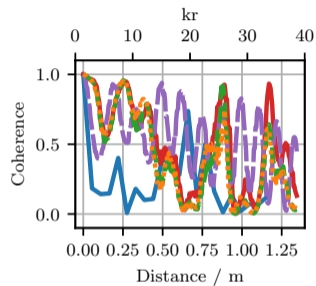
# Spatial coherence - meeting room (4ch)



$f = 300 \text{ Hz}$



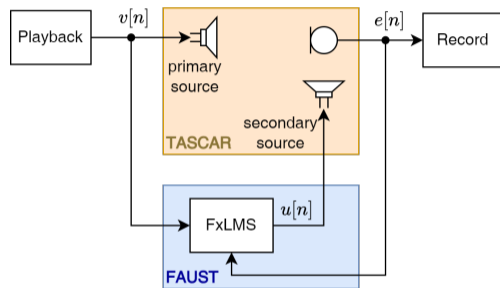
$f = 1000 \text{ Hz}$



$f = 1700 \text{ Hz}$

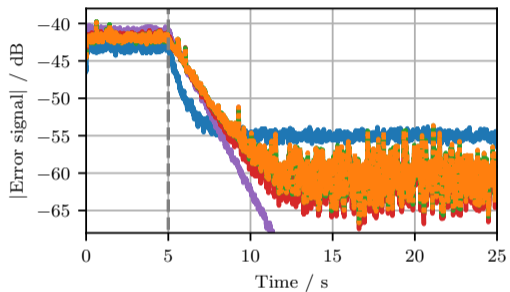
# Real-time simulation

- Feedforward ANC as example
  - ▶ Disturbances emitted by primary source
  - ▶ Play antinoise signal to cancel noise at listening position
  - ▶ Antinoise generated by adaptive filter
- Acoustic simulation in TASCAR
- Signal processing in FAUST (faust2jack)
- Latency defined by JACK settings
- Compare adaptation properties to measurement-based co-simulation

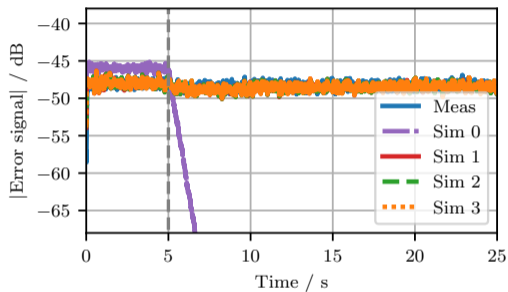




# Adaptation speed

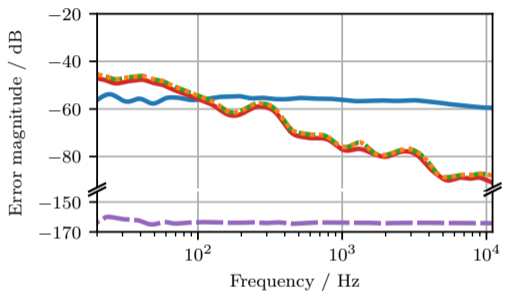


Measurement room

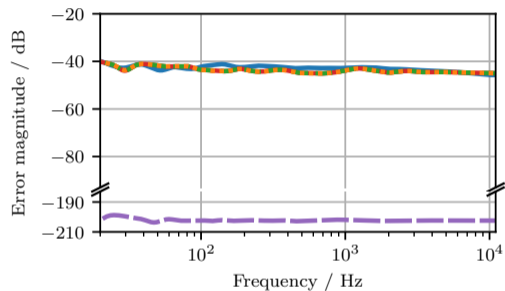


Meeting room

# Steady state spectrum



Measurement room



Meeting room

# Conclusion

- FAUST and TASCAR can be used in combination to evaluate audio algorithms
- Good accuracy in low reflective environments
- Limited validity in diffuse settings
- Handy tool for quick experiments and pre-studies
- Can't replace physical measurements completely

## Outlook

- Time-variant virtualization by using external sensors
- `faust2tascar`

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Paper



Slides



Code

