



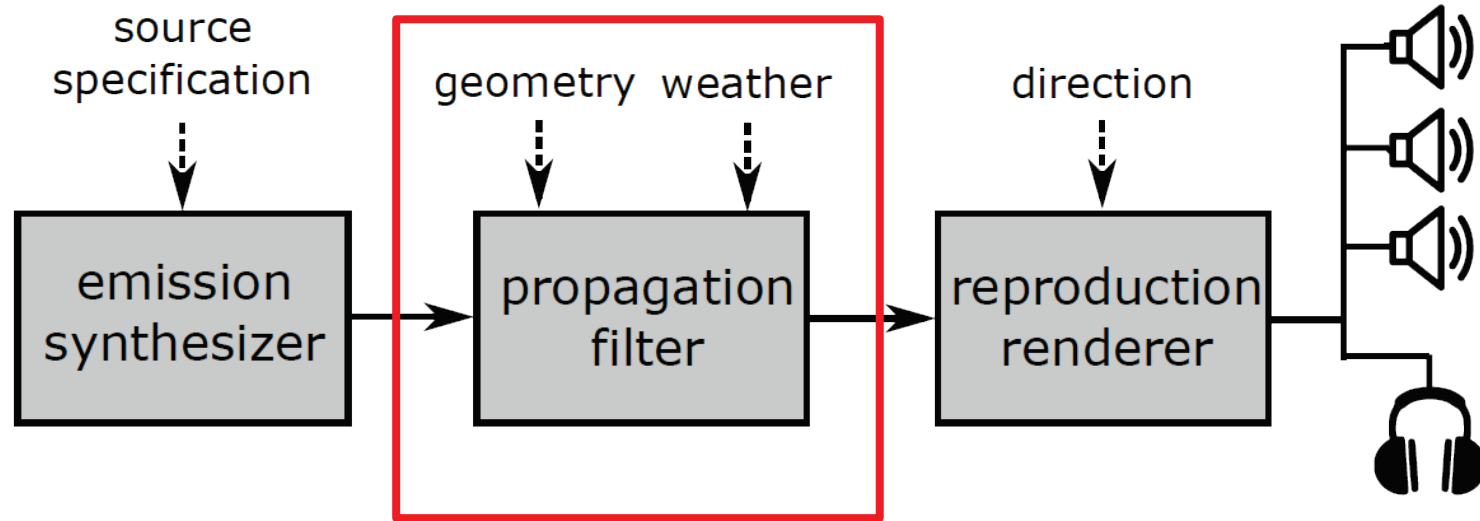
# Acoustics II

## Auralization: Propagation simulation

Reto Pieren

2024

# Flexible auralization approach



# Propagation simulation

# Propagation simulation for fixed sources



# Applications: Evaluation and design of...

Object	Sound sources
Performance space like concert hall	Musical instruments
Radio/TV studio	Natural speaker
Electronic announcement system	Distributed loudspeakers
Open plan office	Many natural speakers + noise masking system
Vehicle interior	Engine, fans, loudspeakers,...

# Principle: Requirements

- Propagation simulation should capture
  - Attenuation → correct absolute sound level, loudness
  - Spectral modification → coloration, timbre
  - Echos, reflection patterns, reverberation → spatial impression, diffusivity
  - Directional information → spatial impression, localization

# Principle: Characteristics

- Assumptions on sound propagation conditions
  - No non-linear distortions → Linearity
  - No movements of sources, receiver and objects + air at rest → Time-invariance
- Linear, time-invariant (LTI) system
  - Multiple sources: Receiver signal = sum of contributions from all sources

$$y_{\text{tot}}(t) = \sum_i y_i(t)$$

- Propagation description: input-output relation by impulse response

# Principle: Characteristics

- Basic case: Point-to-point propagation with omni-directional source and omni-directional receiver

$$y(t) = (x * h)(t) \triangleq \int_{-\infty}^{\infty} x(t - \tau)h(\tau)d\tau$$
$$\approx \int_0^{T_h} x(t - \tau)h(\tau)d\tau$$

$y(t)$ : sound pressure at receiver location

$x(t)$ : dry source signal

$h(t)$ : impulse response for propagation

$T_h$ : truncation length

Operator  $*$ : linear convolution

# Principle: Source and receiver directivity

- Usually of relevance:
  - Directional source, i.e. voice, loudspeaker, instrument
  - Directional receiver for spatial audio reproduction
- Relate source components to receiver components through multiple spatial impulse responses

$$\begin{bmatrix} y_1(t) \\ \vdots \\ y_M(t) \end{bmatrix} = \begin{bmatrix} h_{11}(t) & \cdots & h_{1N}(t) \\ \vdots & \ddots & \vdots \\ h_{M1}(t) & \cdots & h_{MN}(t) \end{bmatrix} * \begin{bmatrix} x_1(t) \\ \vdots \\ x_N(t) \end{bmatrix}$$

$x$ : source component, e.g. spherical harmonics

$y$ : receiver components, e.g. speaker channels or B-format

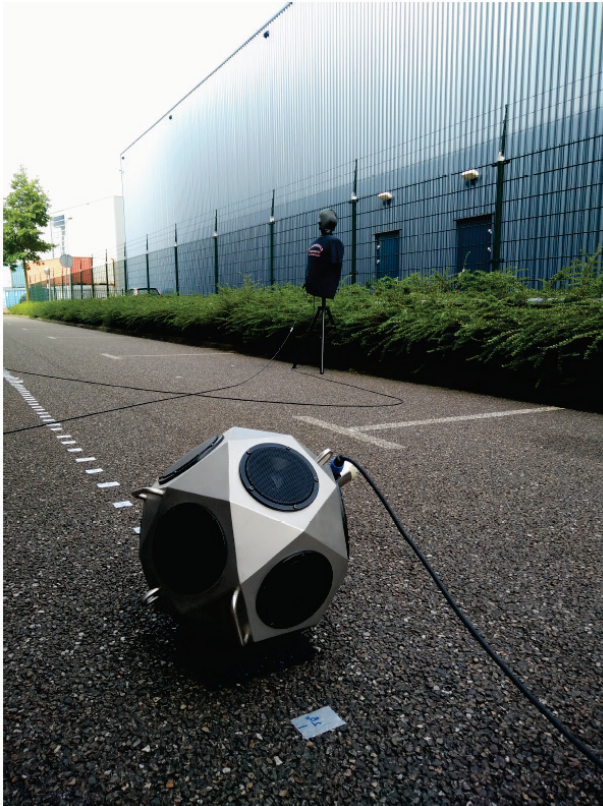
$h$ : spatial impulse responses

# Principle: Impulse responses

$$\begin{bmatrix} y_1(t) \\ \vdots \\ y_M(t) \end{bmatrix} = \begin{bmatrix} h_{11}(t) & \cdots & h_{1N}(t) \\ \vdots & \ddots & \vdots \\ h_{M1}(t) & \cdots & h_{MN}(t) \end{bmatrix} * \begin{bmatrix} x_1(t) \\ \vdots \\ x_N(t) \end{bmatrix}$$

- Where to get **IRs** from?
  1. Real world: In-situ measurements
  2. In the lab: Physical models
  3. From the computer: Computational models

# In-situ measurements



- Impulse response measurements with
  - a controlled source
  - specific signals (MLS, sweep,...)
  - microphone arrays
- Source conditions
  - Compensation of speaker characteristics
  - Calibration of source
  - Consideration of source directivity

[Georgiou 2018. Modeling for auralization of urban environments: incorporation of directivity in sound propagation and analysis of a framework for auralizing a car pass-by. TU Eindhoven, PhD thesis.]

# Physical models



# Physical models (scale model measurements)

- Wave phenomena correctly reproduced, e.g. diffraction, reflection, interference
- Build model in scale 1:N, typical 1:10
  - Wavelength in model:  $\lambda/N$
  - Frequency in model:  $fN \rightarrow$  ultrasound
  - Use of frequency-matching materials
  - Specific loudspeakers and microphones
  - Difference in air absorption: Compensation of impulse responses necessary

# Physical model: Elbphilharmonie Hamburg



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# Physical model: Copenhagen Concert Hall



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Photos: Kurt Eggenschwiler

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# Physical model: Goetheanum Dornach



Bild: Zeitschrift «Das Goetheanum»

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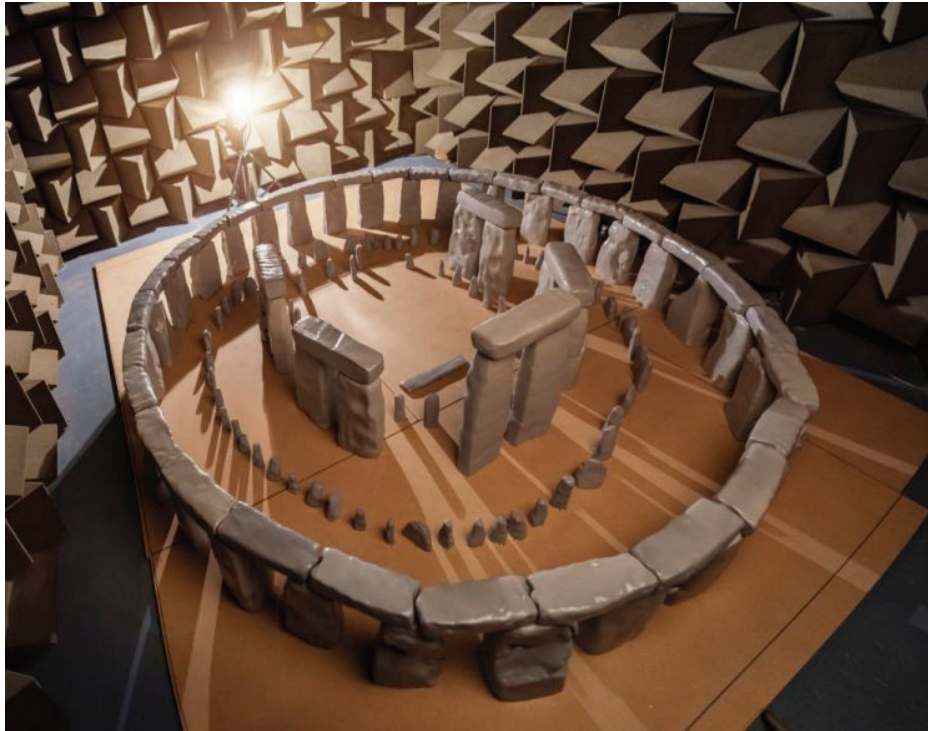


R. Pieren

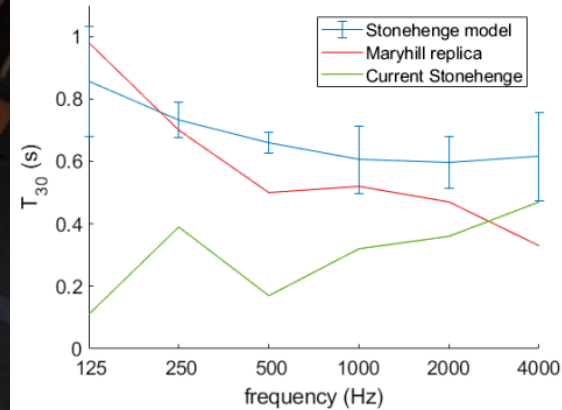
Photo in real room: Kurt Eggenschwiler

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# Physical models (scale model measurements)



- Stonehenge 1:12 model (Trevor Cox, 2020)
- 3d printed model inside anechoic chamber



# Computational models

# Computational models

- Wave-based methods
  - Limit in upper frequency range
  - Ongoing research
- Geometrical acoustical models
  - Used in practice
  - High-frequency approximation
  - Types
    - Image models: mirror source
    - Ray tracing models
    - Hybrid models

[Savioja, L. & Svensson, U.P. 2015. Overview of geometrical room acoustic modeling techniques. *Journal of the Acoustical Society of America*, 138(2).]

# Computational models: Hybrid geometrical models

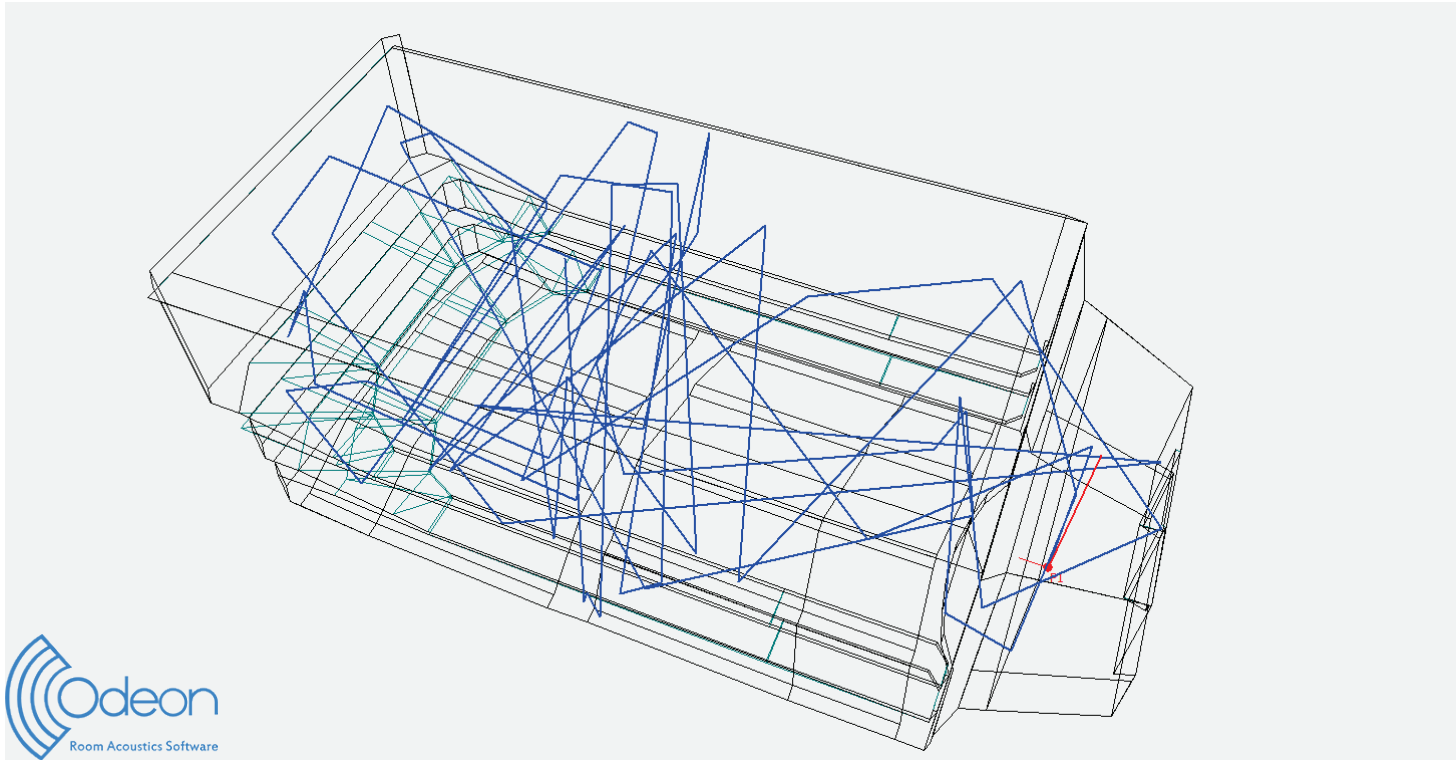
- Combination of mirror source and raytracing
  - Mirror source: First, discrete reflections
  - Ray tracing: Diffuse sound/reverberation
- Commercial software available for room acoustics
  - ODEON, CATT-Acoustic,...



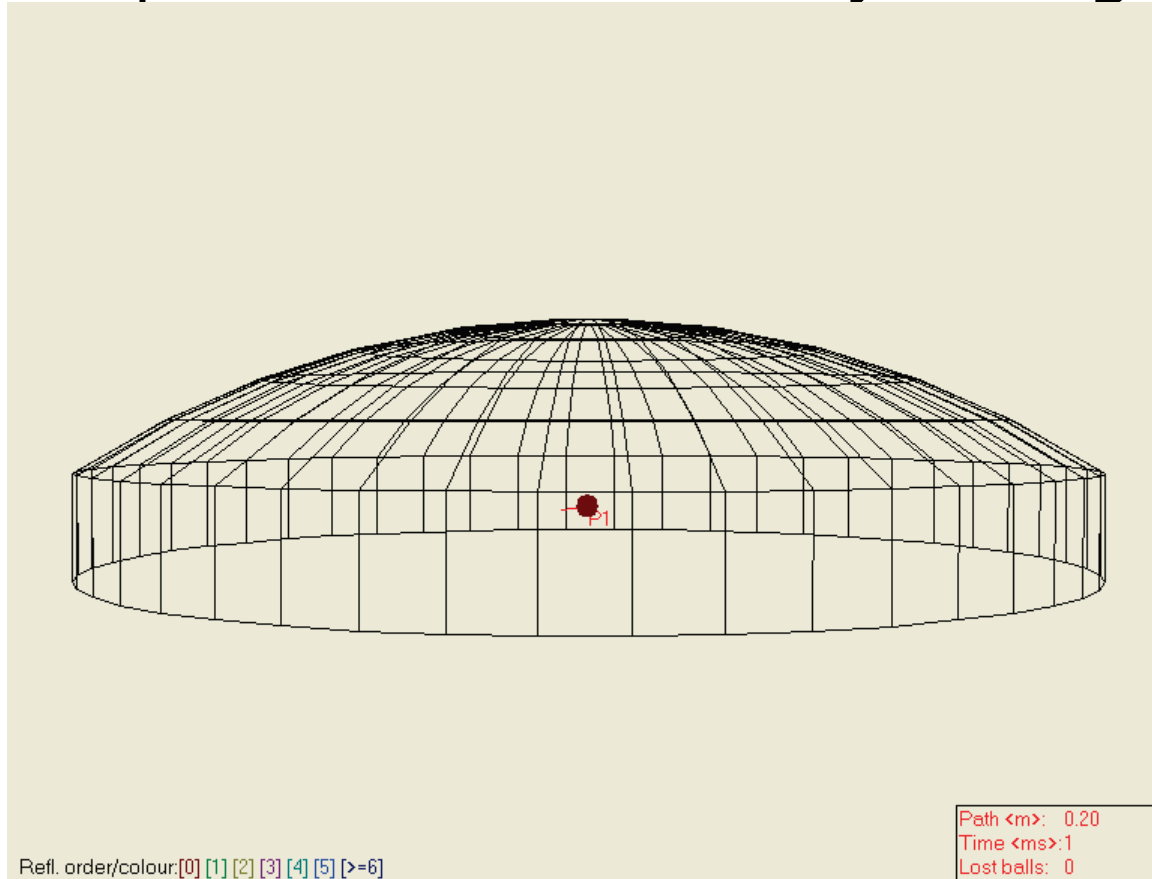
# Computational models: Properties of hybrid geometrical models

- Consideration of directional
  - sources → spectral directivity
  - receivers → binaural or first-order Ambisonics
- Modelling reflections at surfaces
  - Energy loss: frequency-dependent, non-directional absorption coefficient
  - Scattering: using Lambert's law and a scattering coefficient
- Calculation of energy impulse responses  
→ conversion into pressure impulse responses

# Computational models: Ray tracing

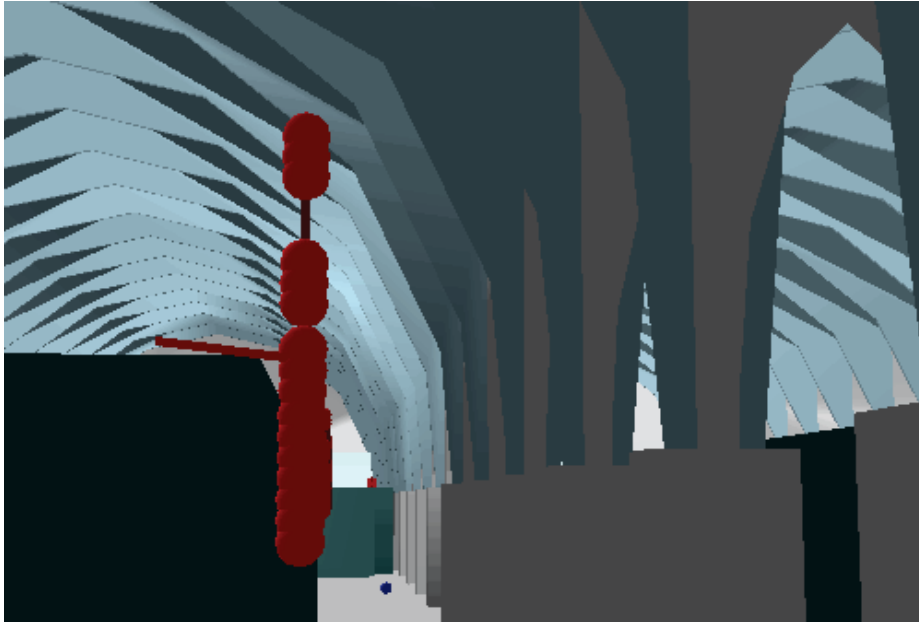


# Computational models: Ray tracing



- Flutter echo and focussing from concave surface

# ODEON example: Array loudspeakers in Copenhagen Central Station (RT=4s)



Near loudspeaker:



Remote loudspeaker:

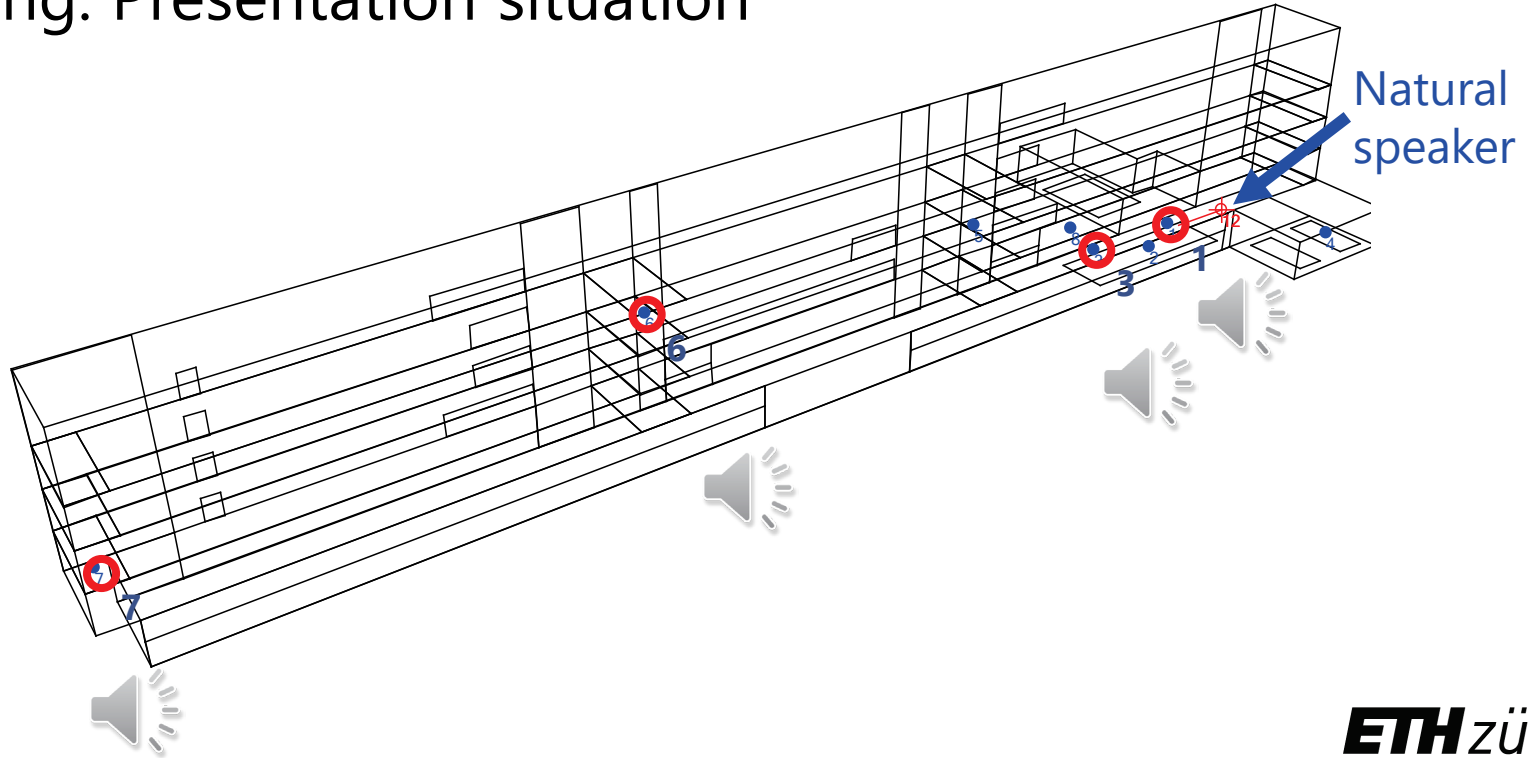


Complete system:



<https://odeon.dk/copenhagen-central-station/>

# ODEON example: Long atrium in an office building: Presentation situation



# Propagation simulation for moving sources

# Moving sources

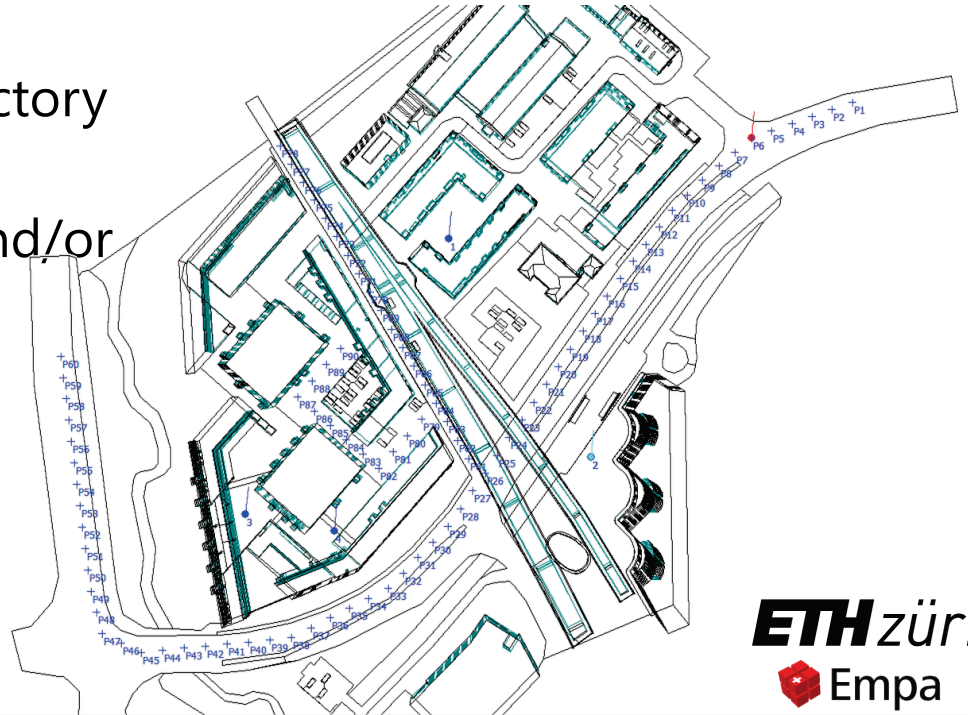
- Application: Traffic noise with outdoor propagation
- Sources: road and rail vehicles, aircraft
- Temporal changes:
  - Source radiation direction relative to receiver
  - Doppler effect
  - Propagation geometry changes over time
    - propagation effects vary with time (no LTI system)
    - time-variable processing of source signals

# Interpolation of impulse responses



# Series of point-to-point impulse responses

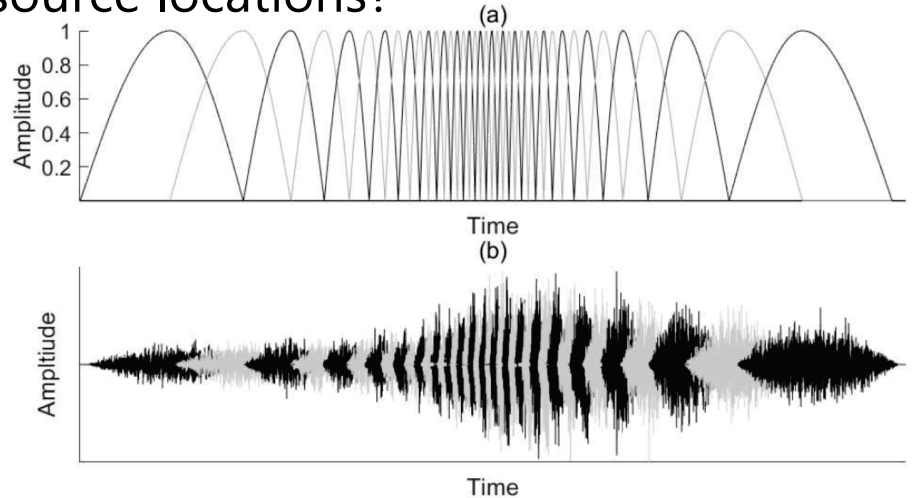
- Discretization of source trajectory
- Obtain series of IRs via measurements, ray tracing and/or wave-based simulation



# Series of point-to-point impulse responses

## ■ How to interpolate IRs between source locations?

- Equal-power crossfading (Georgiou 2018)
  - assumes incoherency
  - no Doppler effect
  - difficulties with impulsive and tonal sounds



- Derivation of generic time-aligned IRs to be interpolated (Heutschi 2020)
  - ambiguous solutions in complex cases
- Ongoing research

# Time-variant propagation filters

# Network of time-variant propagation filters

- Dynamic propagation path tracing
- Representation of discrete propagation effects by filter operations
  - Doppler effect
  - Divergence
  - Air absorption
  - Ground effect
  - Edge diffraction
  - Turbulence effects

# Geometrical divergence of point source

- Sound pressure amplitude of point source

- Spherical waves ( $\rightarrow$  Acoustics I)

- Sound pressure drops 6dB/distance doubling

- frequency-independent

$$\underline{p}(r, t) = \frac{1}{r} \cdot \underline{p}_{\text{plane.wave}} = \frac{1}{r} \hat{p} e^{j(-kr+\phi)} e^{j\omega t}$$

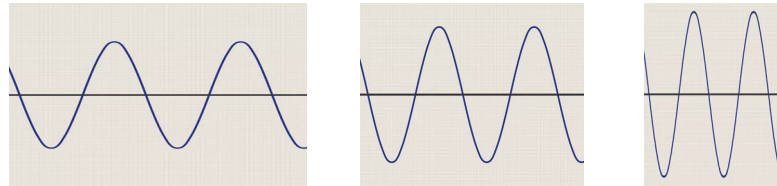
- Slow amplitude modulation:  $p_{\text{rec}}[k] = \frac{p_{1\text{m}}[k]}{r[k]}$

$p_{1\text{m}}[k]$ : emission sound pressure at source time index  $k$  at virtual reference distance of 1 meter

$r$ : source-receiver distance in meters

# Doppler effect

- = Doppler frequency shift + Doppler amplification
- Kinematic source effect
- Application **before** frequency-dependent propagation effects
- Mechanism: Time-varying propagation delay  $\rightarrow$  compression and dilatation of time axis
- Simulation by time-varying delay (vibrato effect) and AM



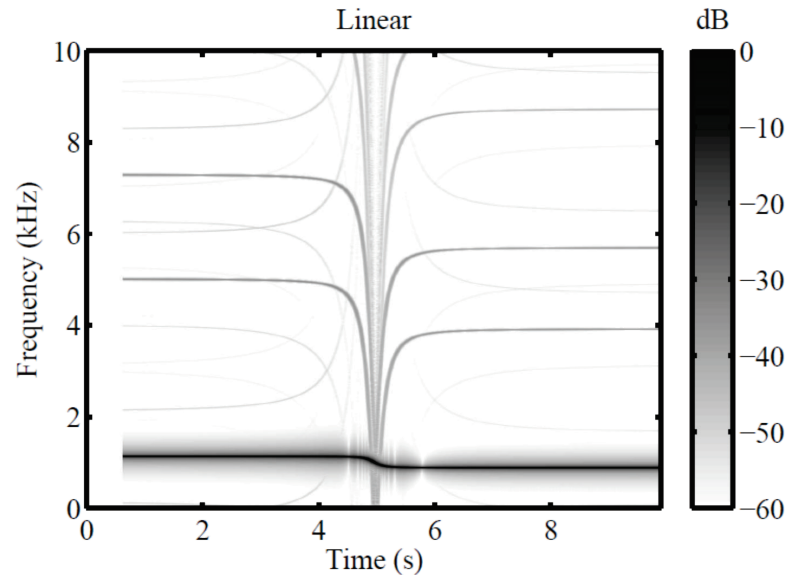
# Doppler frequency shift

- Warped time at receiver:  $t' = t + \Delta t(t) = t + \frac{r(t)}{c}$
- Discrete time  $\rightarrow$  non-integer sampling indices  
 $\rightarrow$  interpolation needed
- Interpolation methods
  - Nearest neighbor  $\rightarrow$  zipper noise
  - Linear interpolation  $\rightarrow$  sometimes sufficient at low speeds
  - Various methods

# Doppler frequency shift: Interpolation artefacts

- Averaging of samples  $\rightarrow$  attenuation of high frequencies
- Non-linear distortions  $\rightarrow$  creation of new frequency components

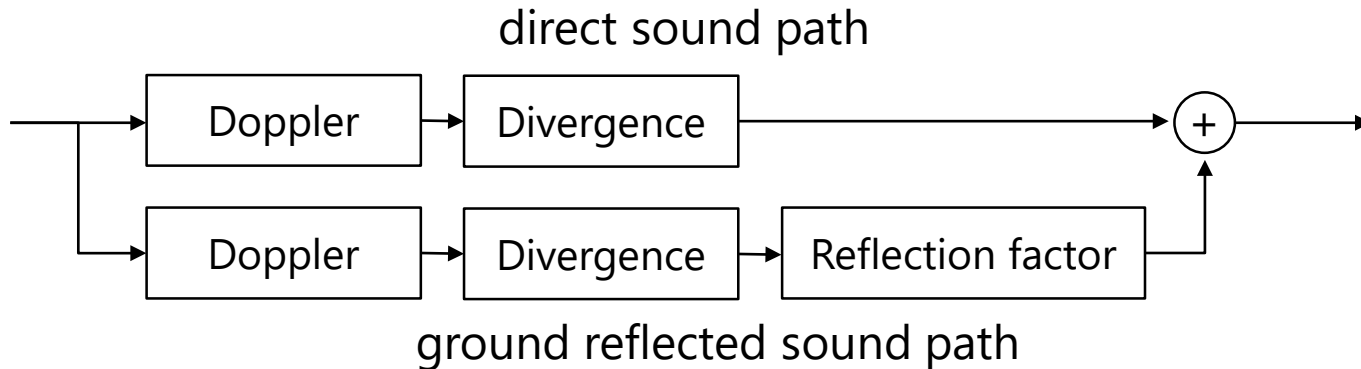
1kHz source moving at  
150 km/h at 7.5 distance





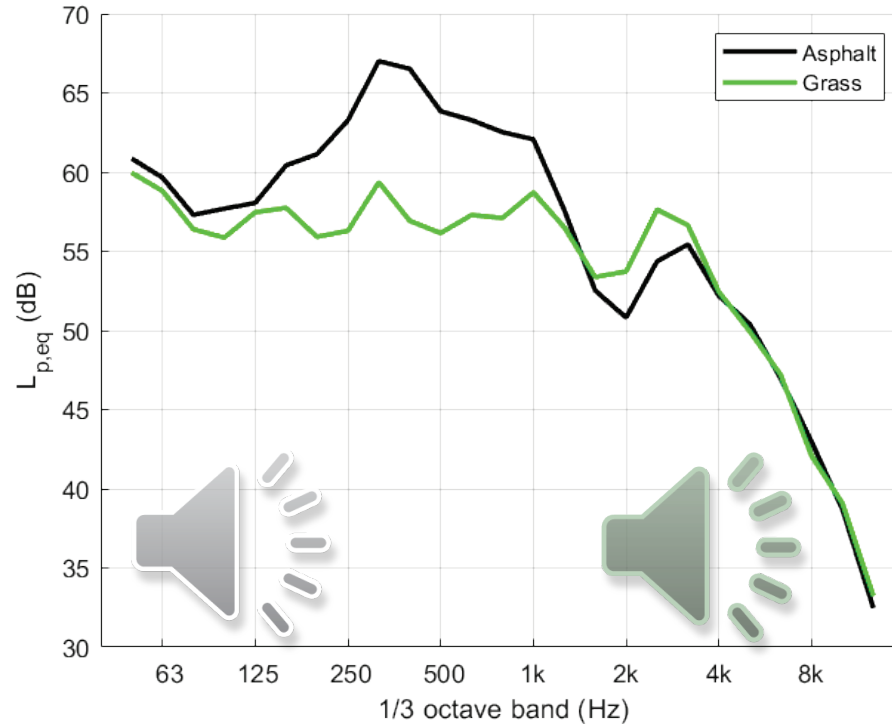
# Ground effect

- Coloration created by interference of direct and ground reflected sound (comb filter effect)
- Simulation by summation of additional signal path



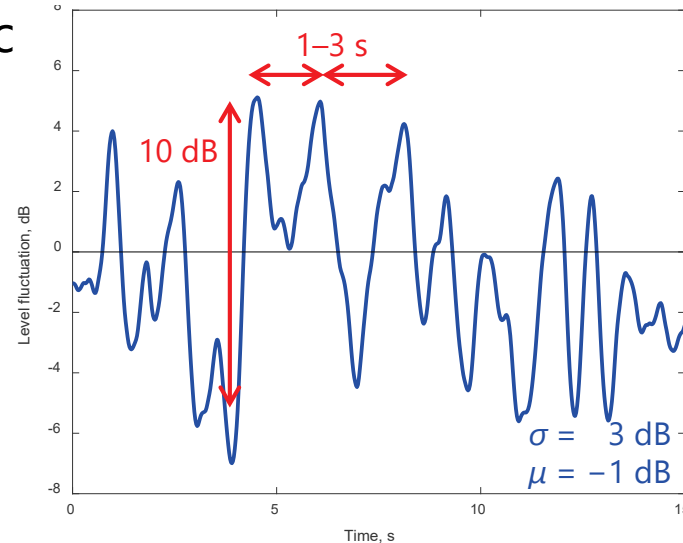
# Ground effect example

- Train pass-by
- Observer standing on ground at 34 m from track
- Propagation over flat ground



# Turbulence-induced amplitude modulation

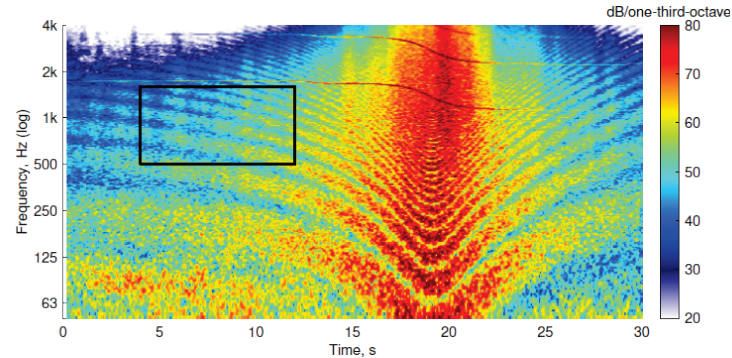
- Long-distance propagation through turbulent atmosphere  
→ refraction → random AM
- Depends on distance, frequency, degree of turbulence
- Simulation run of stochastic high-frequency AM



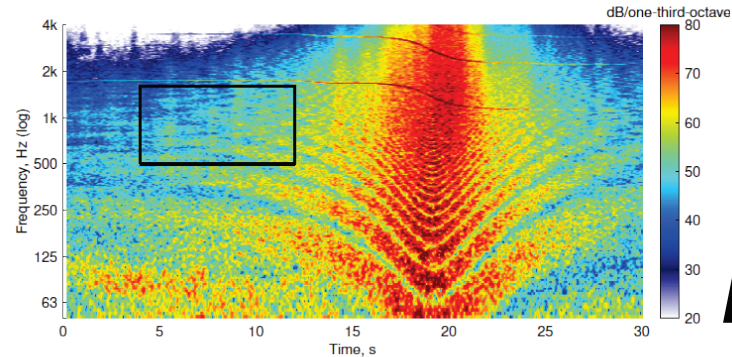
# Turbulence-induced coherence loss

- Propagation through turbulent atmosphere
  - loss of coherence between propagation paths
  - affects ground effect at high frequencies
  - reduction of comb filter effect

[Pieren, R. & Lincke, D. 2022. Auralization of aircraft flyovers with turbulence-induced coherence loss in ground effect. *Journal of the Acoustical Society of America*, 151.]



(a) Synthesis without coherence loss simulation



(b) Synthesis with coherence loss simulation