

Physically Accurate Low Latency Audio for Virtual and Augmented Reality

Ramani Duraiswami University of Maryland, and VisiSonics Corporation

Joint with Dmitry Zotkin, Zhiyun Li, Elena Grassi, Adam O'Donovan, Nail Gumerov,

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# Virtual and Augmented Reality magic (leap

#### VR/AR

Bets by Facebook, Google, and of course Microsoft

Vision of a world where the way you interact with the world is smartly intermediated by technology

Predicted to reach \$120B in 5 years

### The next platform:

Personal Computers, 1990s Internetworked Personal Computers, 200x

Mobile, 201x



Microsoft

### VR/AR

### Virtual Reality

- Create artificial world that the user believes is real
- Simulation, Gaming and Entertainment

### Augmented Reality

- Insert objects and information into the real world
- Training, Surgery, Entertainment

#### Enablers

- Hardware: Moore's law, displays, graphics, tracking,
- Software: Engines for creating virtual worlds
- Visual Perception: Improved latency, using persistence





# When rendering doesn't work: Sickness/Fatigue

#### Sickness:

- Most studied for vision/vestibular system interaction
- Use of persistence and improved frame rates mentioned by Valve/Oculus as primary improvements allowing VR
- Smaller fields of view
- Still lots of stories about gamers getting sick

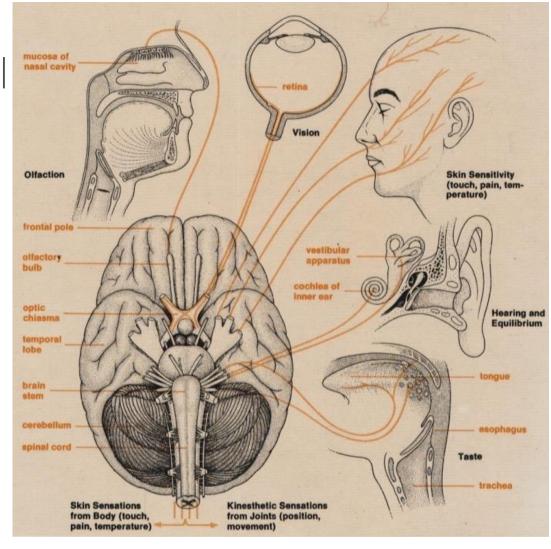
### Fatigue:

- VR Games At E3 Were Making People Sick, Get The Details
- Tendency of users to stop the VR/AR experience early
- Leave experience in minutes instead of hours
- Maybe an even bigger problem for the success of VR/AR



# Fool the Visual System?

- Visual System part of larger perceptual system, responsible for sense-making
- perceptual system is a sophisticated sensing, measuring and computing system
- Designed by evolution to perform real time measurements and take quick decisions
  - Fool this system in to believing that it is perceiving an object that is not there

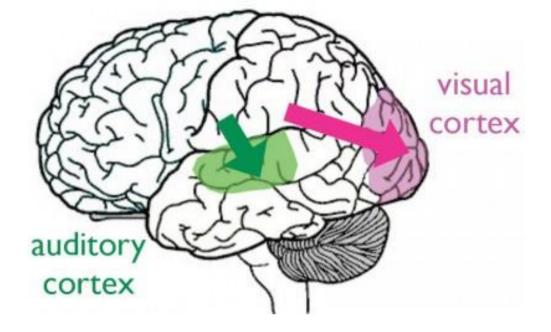


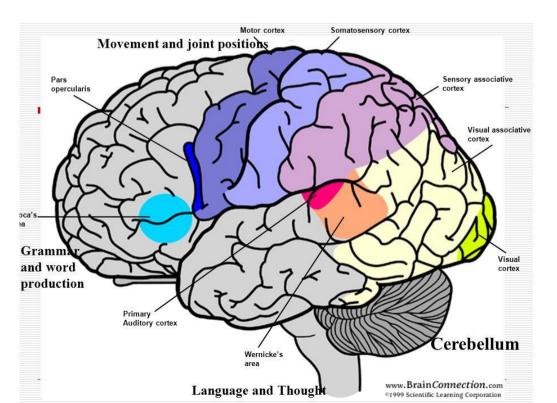




# Sensing the world auditorily

- Vision and audition are stand-off senses
  - Foveated detailed view
  - Broad knowledge of general surroundings
- Occupy nearly same area in the cortical and sensing parts of brain
- Many interconnections, including to the motor areas
- Our hypothesis: Unless the world is rendered consistently the brain experiences fatigue

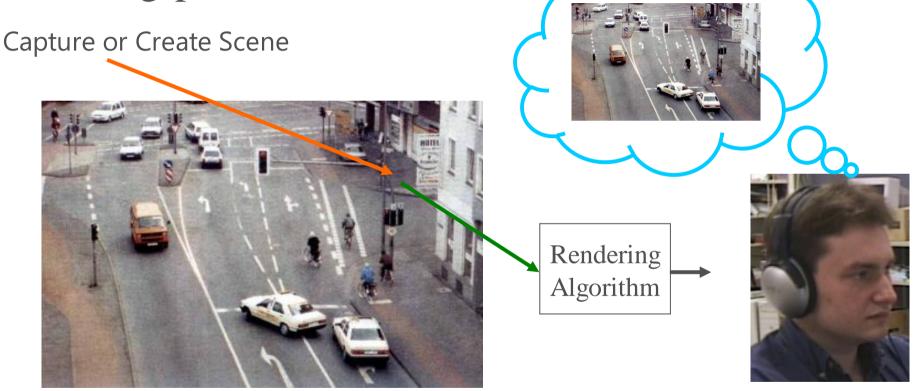




# Problem we have been working on since 2001

What physics/perception based theory can guarantee that we can

solve the following problem?



Want to quantify error in measurement and error in reproduction

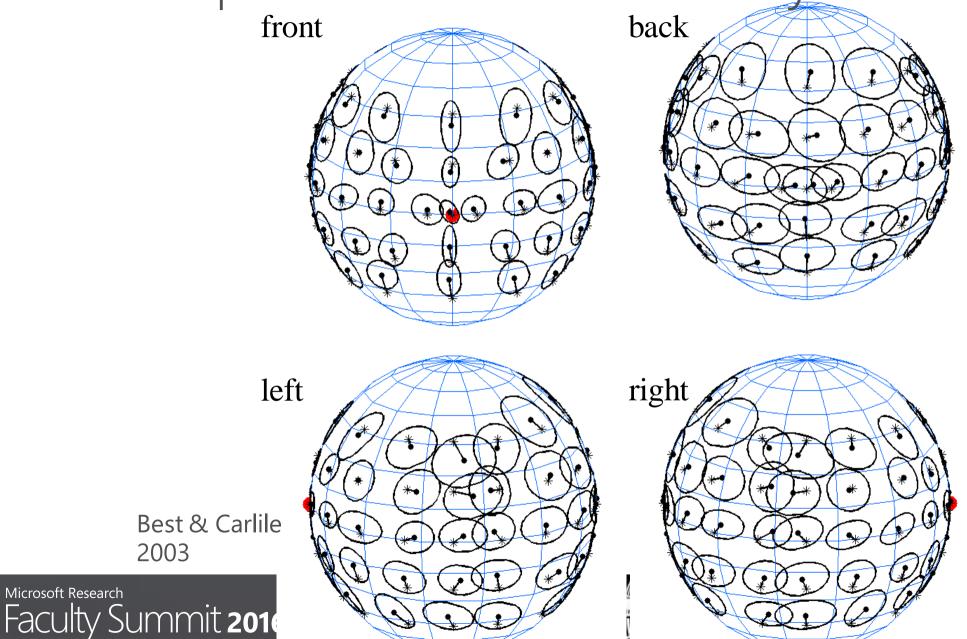




Human spatial localization ability

2003

Microsoft Research





# Hypothesis: Render Sound Correctly

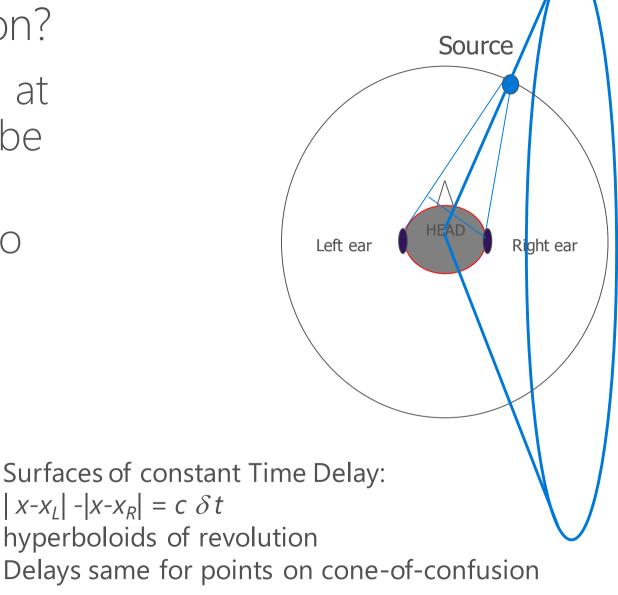
- Get the sound right at the entrances to the ear canals
- Approximately solve the audio propagation problem from sources in the scene to the ear canal
- Do what graphics and vision did
  - Move from emulation to approximate simulation
  - · Use physics based models, appropriately simplified
  - Simplify based on knowledge of what is perceptible: focus attention on things that matter
  - · Level of detail based on available computing power
  - · Capture representations of the real world that allow rendering
- Render not only objects but scenes





### How do we perceive sound location?

- Naïve time and level difference at ears are not sufficient to describe our ability
- Other mechanisms necessary to explain
  - Scattering of sound
    - Off our bodies
    - Off the environment
  - Purposive Motion





# Audible Sound Scattering

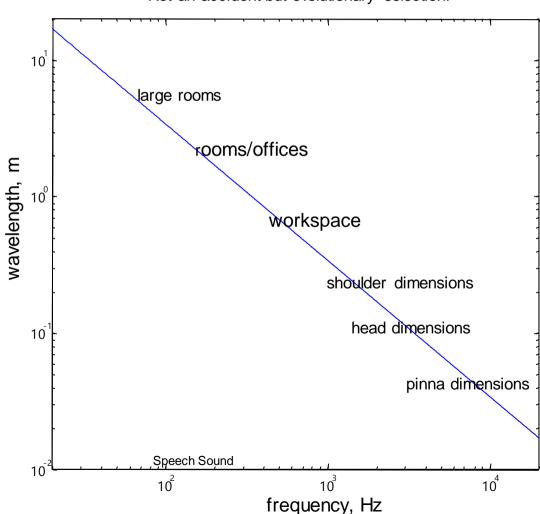
- Sound wavelengths comparable to human dimensions and dimensions of spaces we live in.
- $f\lambda = C$
- When  $\lambda >> a$  wave is unaffected by object

 $\lambda \sim a$  behavior of scattered wave is complex and diffraction effects are important.

 $\lambda << a$  wave behaves like a ray

wavelengths are comparable to our rooms, bodies, and features

Not an accident but evolutionary selection!



### Mathematical modeling of scattering

Helmholtz equation:

#### **Boundary conditions:**

Sound-hard boundaries:

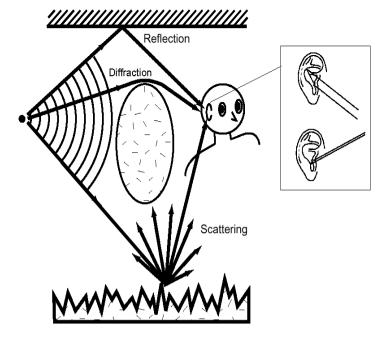
Sommerfeld radiation condition

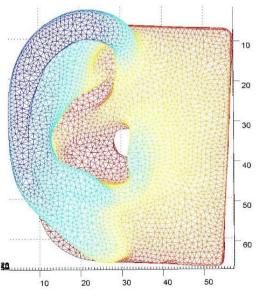
Wave equation: 
$$\frac{\partial^2 p'}{\partial t^2} = c^2 \left( \frac{\partial^2 p'}{\partial x^2} + \frac{\partial^2 p'}{\partial y^2} + \frac{\partial^2 p'}{\partial z^2} \right) = c^2 \nabla^2 p'$$
Fourier Transform from 
$$P(x, y, z, w) = \int_{-\infty}^{\infty} p'(x, y, z, t) e^{-i\omega t} dt$$
Time to Frequency Domain

$$\nabla^2 P + k^2 P = s \, \delta(x - x')$$

$$\frac{\partial P}{\partial n} = 0$$

$$\lim_{r \to \infty} r \left( \frac{\partial P}{\partial r} - ikP \right) = 0$$

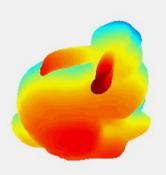


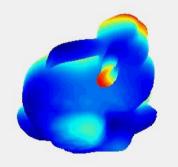


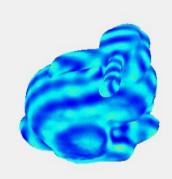
### Fast Multipole Accelerated Solver for Helmholtz equation

#### $O(kD)^2$ instead of $O(kD)^6$

Sound pressure

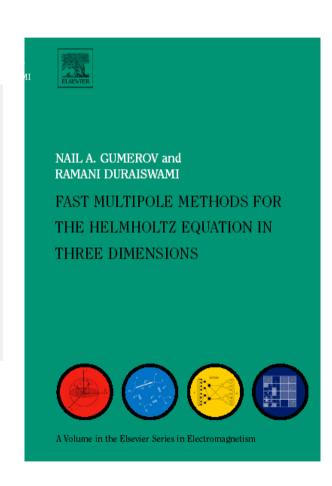






kD=0.96 (250 Hz)

kD=9.6 (2.5 kHz) kD=96 (25 kHz)



**Microsoft** 

### Accurate Approximate Scattering

- Linear systems can be characterized by impulse response (IR)
  - Knowing IR, can compute response to general source by convolution
- Response to impulsive source at a particular location
  - Scattering off person by Head Related Impulse Response (HRIR)
  - Room scattering by Room Impulse Response (RIR)
- Response differs according to source and receiver locations
  - Thus encodes source location
- HRTF and RTF are Fourier transforms of the Impulse response
  - Convolution is cheaper in the Fourier domain (becomes a multiplication)
- Motion is slow enough that a quasi-static model works





# Creating Auditory Reality

- VR/Gaming: Given a sound source and an environment build an engine that reproduces the cues
- Augmented Reality: Capture sound remotely and rerender it by reintroducing cues that exist in the real world
- Scattering of sound off the human
  - Head Related Transfer Functions
- Scattering off the Environment
  - Room Models
- Head motion
  - Head/Body Tracking

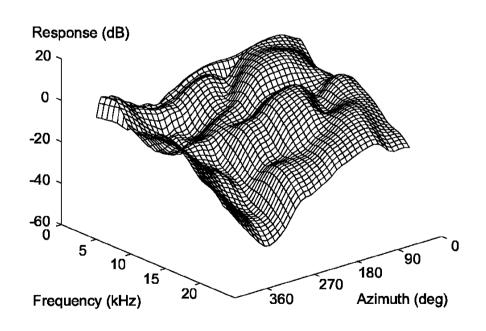


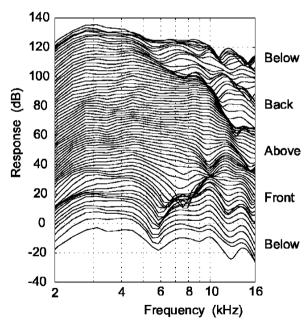


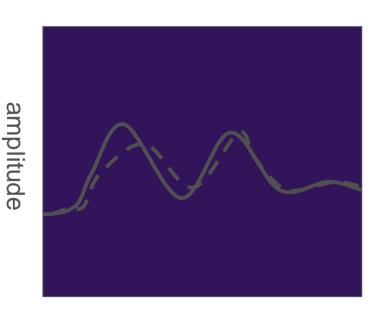
### Head Related Transfer Function

- Scattering causes frequency dependent amplification/attenuation
  - Effects can be of the order of tens of dB
  - Encodes location









frequency



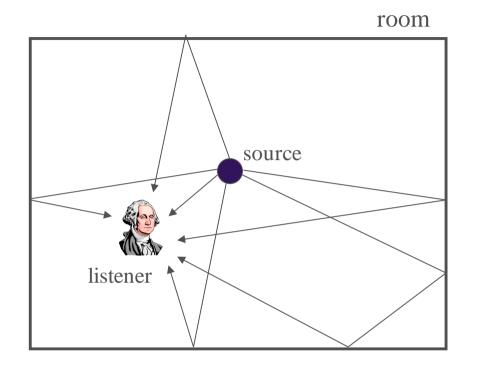


# Breaking up the Filter

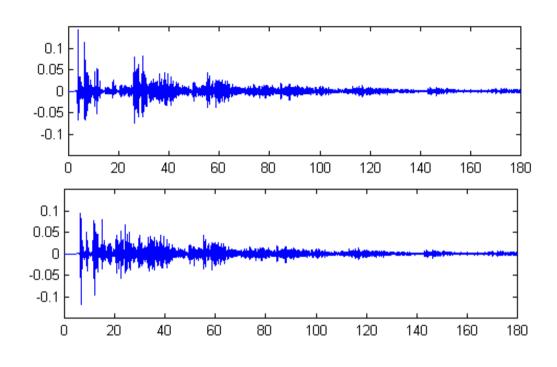
- Convolution is linear
- Early reflections are more important and time separated
  - Important for determining range
- Later reflections are a continuum
  - important for "spaciousness," "envelopment," "warmth," etc.
- Create early reflections filter on the fly
  - reflections of up to 5<sup>th</sup> or 6<sup>th</sup> order (depending on computational resources)
  - These are convolved with their HRTF
- Tail of room impulse response is approximated depending on room size

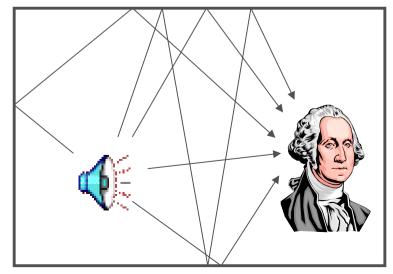


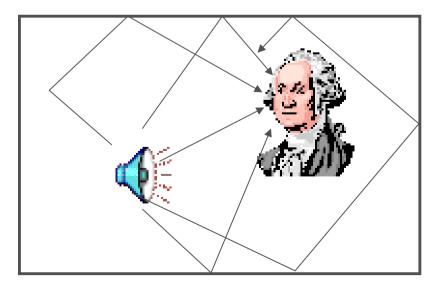
#### Room response and HRTF



- Six to eight orders have perceptive live effect
- 30 orders influence the room ambience







# HRTFs are very individual

- Humans have different sizes and shapes
- Ear shapes are very individual as well
  - Before fingerprints, Alphonse Bertillon used a system of identification of
- Even today ear shots are part of
  - Mugshots & INS photographs
- If ear shapes and body sizes are different
  - Properties of scattered wave are different
  - HRTFs will be very individual
- Need individual HRTFs for creating accurate virtual audio







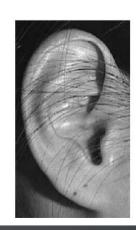












### Typically measured

- Sound presented via speakers
- Speaker locations sampled
- Takes 10 minutes to several hours
- Subject given feedback to keep pose relatively steady
- Hoop is usually >1m away (no range data)

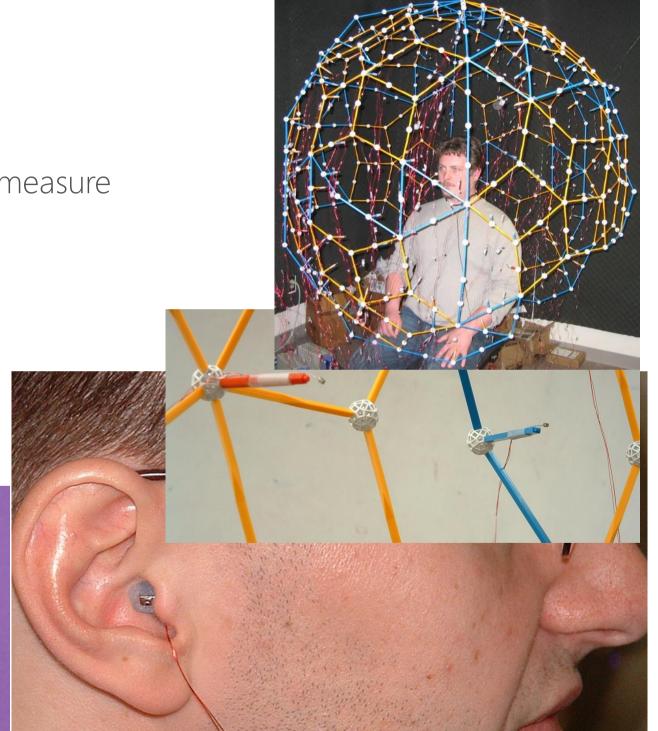




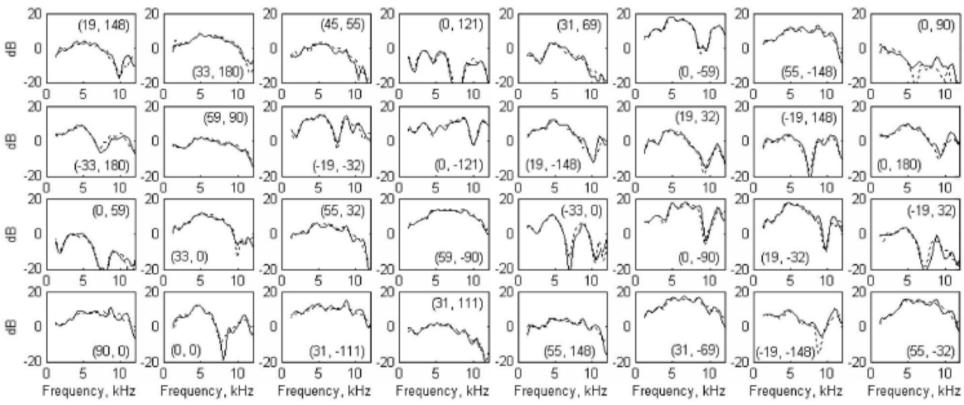
# Fast Approach

- Turned out headphone drivers
- Array of tiny microphones
- Send out a highpass signal and measure received signal
- Use analytical anthropometric representation for low frequencies and compose
- Extrapolate range





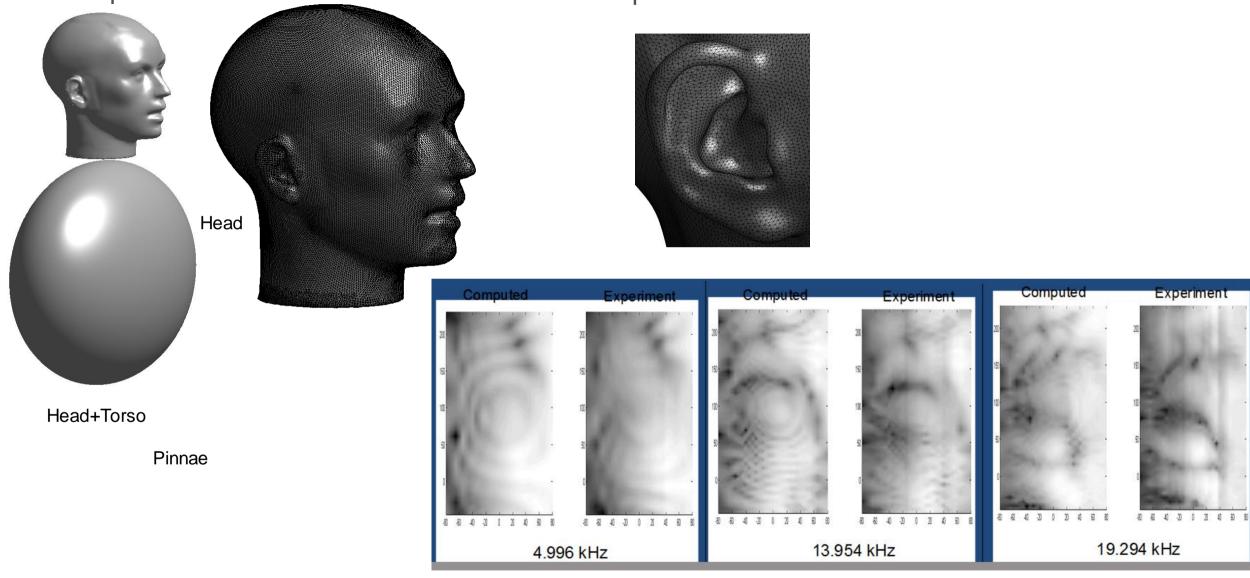
- Direct vs. Reciprocal (Zotkin et al. 2006, JASA)
- Currently reduced to under 30 s



D.N. Zotkin, R. Duraiswami, E. Grassi, and N.A. Gumerov, "Fast head-related transfer function measurement via reciprocity," J. Acoust. Soc. Am., 120:2202-14, 2006



### Compute HRTFs via Fast Multipole Acelerated BEM

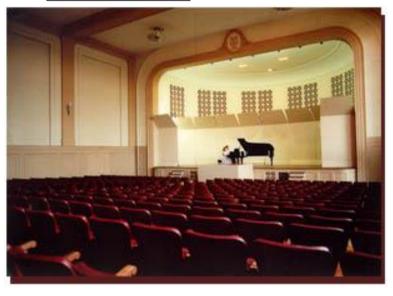






### Best Seat in the House: Telepresence

#### **RECORDING**







**PLAYBACK** 

- Place microphones at a remote location (e.g. concert hall)
- Replay spatialized audio at a remote location
- Must play it for many users
- Use rendering algorithms/ representations





### Representation via spherical wavefunctions

- sound at a point
  - So we can represent the sound at a point in terms of the local point-eigenfunctions of the Helmholtz equation

$$\psi_{in}(k;\mathbf{r}) = \sum_{n=0}^{\infty} \sum_{m=-n}^{n} A_n^m R_n^m(k;\mathbf{r}),$$

$$R_n^m(k; \mathbf{r}) = j_n(kr) Y_n^m(\theta, \varphi),$$

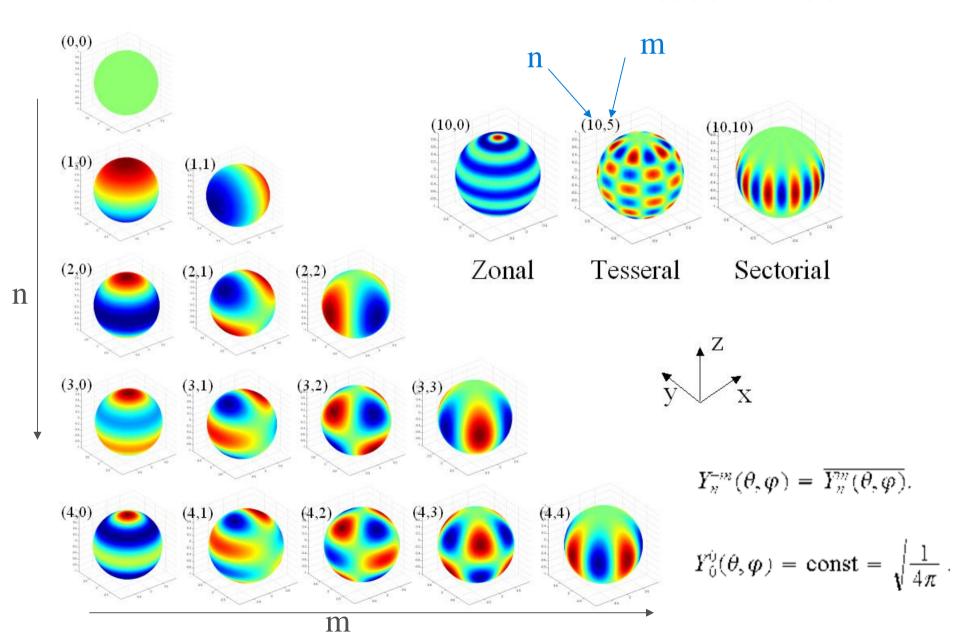
• Expand solutions in series, but truncate at p terms causing an error  $\varepsilon_{\rm p}$   $|\epsilon_p\left(\mathbf{s},\mathbf{r}\right)|\lesssim \exp\left\{-\frac{1}{3}\left[2\frac{p-kR}{\left(kR\right)^{1/3}}\right]^{3/2}\right\}=\delta_p,\quad kR\gg 1.$ 

- Error depends on frequency
  - For a given sound of wavenumber *k* this gives us minimum order for sensible representation

### Spherical Harmonics

$$Y_n^m(\theta,\varphi) = (-1)^m \sqrt{\frac{2n+1}{4\pi} \frac{(n-|m|)!}{(n+|m|)!}} P_n^{|m|}(\cos\theta) e^{im\varphi},$$
  

$$n = 0, 1, 2, ...; \qquad m = -n, ..., n.$$



### Yet another representation (Plane Waves)

- any soundfield in regular region can be expressed as an integral form of plane waves.
  - Integral over a unit sphere at the point
  - Decomposes any sound field in to a set of planewaves of various strengths
- $\psi_{in}(\mathbf{r}) = \frac{1}{4\pi} \int_{S_u} e^{ik\mathbf{s}\cdot\mathbf{r}} \mu_{in}(\mathbf{s}) dS(\mathbf{s}),$ engths

  Plane waves

  Coeffs

• Connected to spherical representation

$$e^{ik\mathbf{s}\cdot\mathbf{r}} = 4\pi \sum_{n=0}^{\infty} \sum_{m=-n}^{n} i^{n} Y_{n}^{-m}(\mathbf{s}) R_{n}^{m}(\mathbf{r}), \quad R_{n}^{m}(\mathbf{r}) = \frac{i^{-n}}{4\pi} \int_{S_{u}} e^{ik\mathbf{s}\cdot\mathbf{r}} Y_{n}^{m}(\mathbf{s}) dS(\mathbf{s}),$$

• In practice these integrals are evaluated via quadrature

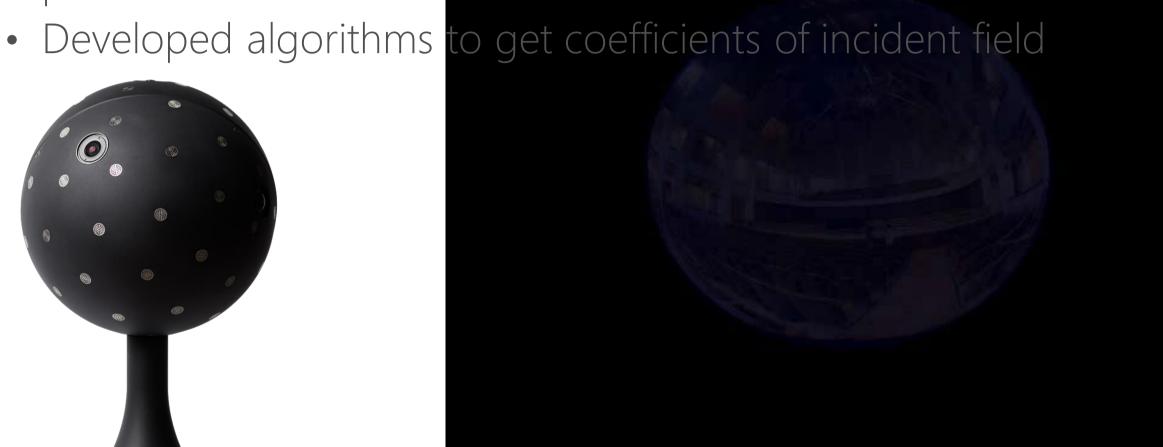
$$\int_{S_u} F(\mathbf{s}) dS = \sum_{j=0}^{L_Q-1} F(\mathbf{s}_j) w_j, \ F(\mathbf{s}) = \sum_{n=0}^{p-1} \sum_{m=-n}^{n} C_n^m Y_n^m(\mathbf{s}),$$

- Approximation error in this case is related to error in the quadrature
- Ouadrature error formula relates Lo to p

### Issues: Reconstruct coefficients from measurements

• What we measure is the response of the field with the sphere present – not the incident field









# VisiSonics RealSpace3D Engine

