

# Lecture

## Speech and Audio Signal Processing



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### Lecture 14: Sound reproduction by Wave Field Synthesis (WFS) and Higher Order Ambisonics (HOA)



- ❑ Wave Field Synthesis (WFS)
  - ❑ Typically using a 2D loudspeaker setup, e.g., circular or rectangular array.
  - ❑ Sound generation in a plane.
  - ❑ Targets for generation a good sound field within the plane rather independent of the listening position.
  
- ❑ Higher Order Ambisonics (HOA)
  - ❑ Often targets for the generation of a 3D sound field.
  - ❑ Typically, a precise sound field should be generated in a sweet spot.

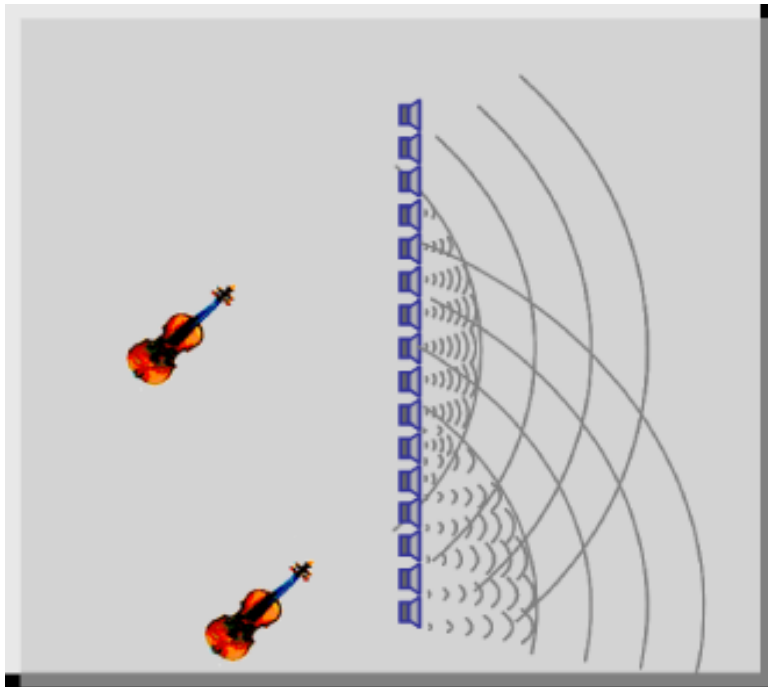


# Wave field synthesis

# WFS: Concept of Wave field synthesis

## □ The basic idea:

Reproduction of a sound source (sound waves) by the superposition of loudspeaker signals:



## □ Targets:

Typically, horizontal modelling by loudspeakers in a circular or quadratic, etc. array.

Reproduction of sound in a rather big section of the loudspeaker array

=> no sweet spot.

Limitations by the loudspeaker distance, i.e., spatial sampling which leads to spatial aliasing.

# WFS: Concept of Wave Field Synthesis

Sivantos WFS System



<https://www.quora.com/>

- Based on the wave equation:

$$\Delta p(\mathbf{x}, t) - \frac{1}{c^2} \frac{\partial^2}{\partial t^2} p(\mathbf{x}, t) = 0 \quad \mathbf{x} = [x, y, z]^T$$

- Application of the Fourier Transform => Helmholtz equation :

$$\Delta P(\mathbf{x}, \omega) + \left(\frac{\omega}{c}\right)^2 P(\mathbf{x}, \omega) = 0 \quad P(\mathbf{x}, \omega) \bullet \text{---} \circ p(\mathbf{x}, t)$$

- Plane wave solutions in the time and frequency domain:

$$p(\mathbf{x}, t) = f\left(t + \frac{1}{c} \langle \mathbf{x}, \mathbf{n} \rangle\right) \quad \langle \mathbf{x}, \mathbf{n} \rangle \text{ scalar product between the two vectors}$$

$$P(\mathbf{x}, \omega) = F(\omega) e^{j \frac{\omega}{c} \langle \mathbf{x}, \mathbf{n} \rangle} \quad \mathbf{n} \text{ Vector defining the propagation direction of the plane wave}$$



- Contribution of a spatial sound field of a point source (radiating in all directions):

$$\delta_{3D}(\mathbf{x}) = \delta(x) \delta(y) \delta(z)$$

$$q(\mathbf{x}, t) = q_0(\mathbf{x}_0, t) \delta_{3D}(\mathbf{x} - \mathbf{x}_0) \quad Q_0(\omega, \mathbf{x}_0) \bullet \text{---} \circ q_0(t, \mathbf{x}_0)$$

$$Q(\mathbf{x}, \omega) = Q_0(\mathbf{x}_0, \omega) \delta_{3D}(\mathbf{x} - \mathbf{x}_0)$$

- Spatial sound field for a single point source:

$$P_0(\mathbf{x}, \omega) = \iiint_V G_{3D}(\mathbf{x}|\mathbf{x}_0, \omega) Q(\mathbf{x}_0, \omega) d\mathbf{x}_0$$

- in free-field (Green's function):

$$G_{3D}^f(\mathbf{x}|\mathbf{x}_0, \omega) = \frac{1}{4\pi} \frac{e^{-j\frac{\omega}{c}|\mathbf{x}-\mathbf{x}_0|}}{|\mathbf{x} - \mathbf{x}_0|}$$

- **3D:** Point source, i.e., superposition of spherical waves, according to the number of point sources

$$Q_0(\mathbf{x}, \omega) = Q_0(x, y, z, \omega)$$

- **2D:** Line source, i.e., superposition of appropriate free-field model according to the number of line sources

$$Q_1(\mathbf{x}, \omega) = Q_1(x, y, \omega)$$

- **Sound field of a line source:**

$$P_1(\mathbf{x}, \omega) = \iiint_V G_{3D}(\mathbf{x}|\mathbf{x}_0, \omega) Q_1(\mathbf{x}_0, \omega) d\mathbf{x}_0$$

Does not depend on z





□ Sound field of a line source:

$$P_1(\mathbf{x}, \omega) = \iiint_V G_{3D}(\mathbf{x}|\mathbf{x}_0, \omega) Q_1(\mathbf{x}_0, \omega) d\mathbf{x}_0$$

Does not depend on z

$$P_1(\mathbf{x}, \omega) = \iint_V \underbrace{\left[ \int G_{3D}(\mathbf{x}|\mathbf{x}_0, \omega) dz_0 \right]}_{G_{2D}(\mathbf{x}|\mathbf{x}_0, \omega)} Q_1(\mathbf{x}_0, \omega) d\mathbf{x}_0$$

2-dim

□ Resulting in the free-field sound field of a line source:

$$G_{2D}(\mathbf{x}|\mathbf{x}_0, \omega) = \int_{-\infty}^{\infty} G_{3D}(\mathbf{x}|\mathbf{x}_0, \omega) dz_0$$

$$G_{2D}^f(\mathbf{x}|\mathbf{x}_0, \omega) = \frac{j}{2} H_0^{(2)} \left( \frac{\omega}{c} \|\mathbf{x} - \mathbf{x}_0\| \right)$$

with:  $H_0^{(2)} \left( \frac{\omega}{c} \rho \right) = J_0 \left( \frac{\omega}{c} \rho \right) - j N_0 \left( \frac{\omega}{c} \rho \right)$

- Relation of Green's functions of line and point sources for far-field assumptions (mathematical approximation) :

$$G_{3D}^f(\mathbf{x}|\mathbf{x}_0, \omega) = \frac{1}{4\pi} \frac{e^{-j\frac{\omega}{c}\|\mathbf{x}-\mathbf{x}_0\|}}{\|\mathbf{x}-\mathbf{x}_0\|}$$

$$G_{2D}^f(\mathbf{x}|\mathbf{x}_0, \omega) = \frac{j}{2} H_0^{(2)}\left(\frac{\omega}{c}\|\mathbf{x}-\mathbf{x}_0\|\right)$$

$$\frac{\omega}{c}\|\mathbf{x}-\mathbf{x}_0\| \gg 1$$

- Resulting in the approximation of the free-field sound field of a line source for the far-field:

$$G_{2D}^f(\mathbf{x}|\mathbf{x}_0, \omega) \approx \sqrt{\frac{2\pi\|\mathbf{x}-\mathbf{x}_0\|}{j\frac{\omega}{c}}} G_{3D}^f(\mathbf{x}|\mathbf{x}_0, \omega)$$

## □ Basic concept of WFS:

The sound field within the Volume  $V$  can be expressed by the known sound pressure and its gradient on the surface of the boundary volume, assuming no sound sources within the volume

$$P_0(\mathbf{x}, \omega) = - \oint\!\!\!\oint_{\partial V} \left[ G_{3D}(\mathbf{x}|\mathbf{x}_0, \omega) \frac{\partial}{\partial \mathbf{n}} P_0(\mathbf{x}_0, \omega) - P_0(\mathbf{x}_0, \omega) \frac{\partial}{\partial \mathbf{n}} G_{3D}(\mathbf{x}|\mathbf{x}_0, \omega) \right] d\mathbf{x}_0$$

$\mathbf{x} \in V$

## □ Then the sound pressure outside the volume is zero

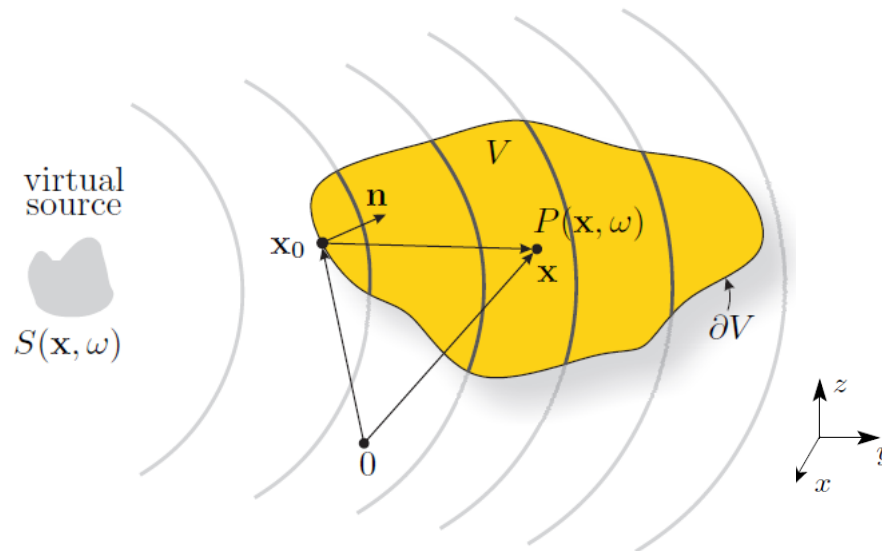
$$P_0(\mathbf{x}, \omega) = 0 \quad \mathbf{x} \notin V$$

## □ This is a condition which is not really necessary, since we are interested in the sound field within the volume. However, we can use a relaxation of this property for a later simplification of the Kirchhoff-Helmholtz-Integral.

# WFS: Kirchhoff-Helmholtz-Integral: 3D and 2D

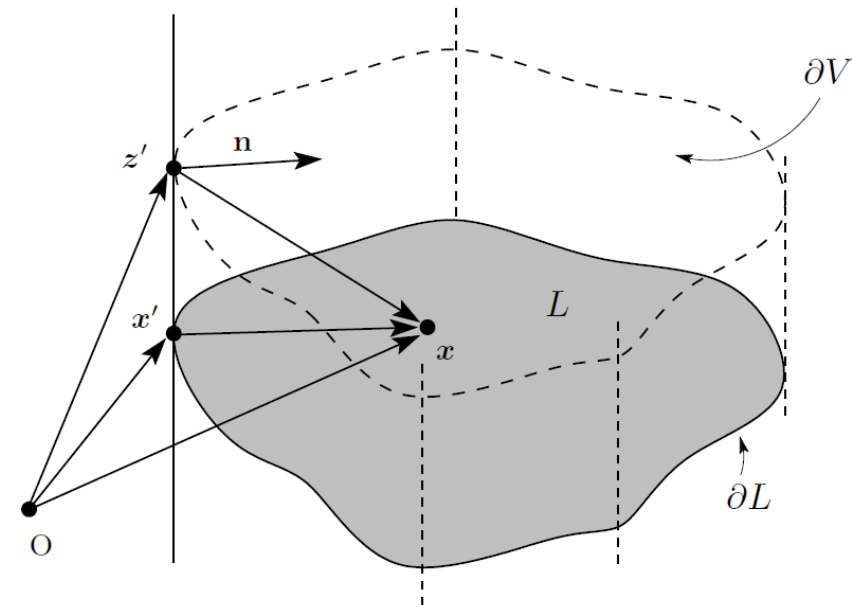
## Basic concept of WFS:

General 3D setup



Prism 3D setup

=> leading to the 2D setup  
which models the sound field in a plane L.



## □ 2D WFS equation in the plane:

Surface integration over the prism, invariant sound field within the z-direction  $P_1(\mathbf{x}_0, \omega) = P_0(\mathbf{x}_0, \omega)$

$$\begin{aligned} P_0(\mathbf{x}, \omega) &= - \oiint_{\partial V} \left[ G_{3D}(\mathbf{x}|\mathbf{x}_0, \omega) \frac{\partial}{\partial \mathbf{n}} P_0(\mathbf{x}_0, \omega) - P_0(\mathbf{x}_0, \omega) \frac{\partial}{\partial \mathbf{n}} G_{3D}(\mathbf{x}|\mathbf{x}_0, \omega) \right] d\mathbf{x}_0 \\ &= - \oint_{\partial L} \int_{z_0=-\infty}^{\infty} \left[ G_{3D}(\mathbf{x}|\mathbf{x}_0, \omega) \frac{\partial}{\partial \mathbf{n}} P_0(\mathbf{x}_0, \omega) \right. \\ &\quad \left. - P_0(\mathbf{x}_0, \omega) \frac{\partial}{\partial \mathbf{n}} G_{3D}(\mathbf{x}|\mathbf{x}_0, \omega) \right] dx_0 dy_0 dz_0 \\ &= - \oint_{\partial L} \left[ \int_{z_0=-\infty}^{\infty} G_{3D}(\mathbf{x}|\mathbf{x}_0, \omega) dz_0 \right] \frac{\partial}{\partial \mathbf{n}} P_0(\mathbf{x}_0, \omega) \\ &\quad - P_0(\mathbf{x}_0, \omega) \frac{\partial}{\partial \mathbf{n}} \left[ \int_{z_0=-\infty}^{\infty} G_{3D}(\mathbf{x}|\mathbf{x}_0, \omega) dz_0 \right] dx_0 dy_0 \end{aligned}$$

$$P_1(\mathbf{x}, \omega) = - \oint_{\partial L} \left[ G_{2D}(\mathbf{x}|\mathbf{x}_0, \omega) \frac{\partial}{\partial \mathbf{n}} P_1(\mathbf{x}_0, \omega) - P_1(\mathbf{x}_0, \omega) \frac{\partial}{\partial \mathbf{n}} G_{2D}(\mathbf{x}|\mathbf{x}_0, \omega) \right] d\mathbf{x}_0$$

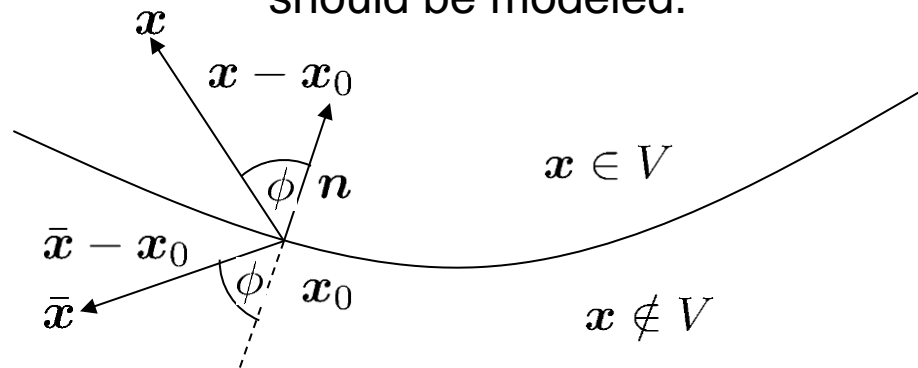
## WFS: Choice of Green's functions

### □ Concept:

Choose Green's functions such that it is symmetric with respect to the boundary of the surface.

Only possible if the property  $P_0(\mathbf{x}, \omega) = 0 \quad \mathbf{x} \notin V$

is relaxed.  $\Rightarrow$  No problem: Only the sound field within the volume should be modeled.



- Choice of the Green's function at the boundary as the sum of symmetric free-field Green's function with respect to the volume's surface

$$G_{iD}(\mathbf{x}|\mathbf{x}_0, \omega) = G_{iD}^f(\mathbf{x}|\mathbf{x}_0, \omega) + G_{iD}^f(\bar{\mathbf{x}}|\mathbf{x}_0, \omega)$$

$$i \in [2, 3]$$

## □ Resulting in:

Dipole:  $\frac{\partial}{\partial \mathbf{n}} G_{iD}(\mathbf{x}|\mathbf{x}_0, \omega) = 0$

Monopole:  $G_{iD}(\mathbf{x}|\mathbf{x}_0, \omega) = 2 G_{iD}^f(\mathbf{x}|\mathbf{x}_0, \omega)$

No dipoles, only  
monopoles model the  
Kirchhoff-Helmholtz-Integral

## □ Kirchhoff-Helmholtz-Integrals only with monopoles:

$$P_0(\mathbf{x}, \omega) = - \oint\!\!\!\oint_{\partial V} 2 G_{3D}^f(\mathbf{x}|\mathbf{x}_0, \omega) \frac{\partial}{\partial \mathbf{n}} P_0(\mathbf{x}_0, \omega) d\mathbf{x}_0 \quad \mathbf{x} \in V \quad 3 \text{ dimensional}$$

$$P_1(\mathbf{x}, \omega) = - \oint_{\partial L} 2 G_{2D}^f(\mathbf{x}|\mathbf{x}_0, \omega) \frac{\partial}{\partial \mathbf{n}} P_1(\mathbf{x}_0, \omega) d\mathbf{x}_0 \quad \mathbf{x} \in L \quad 2 \text{ dimensional}$$



## □ Replacing line sources by point sources:

$$G_{2D}^f(\mathbf{x}|\mathbf{x}_0, \omega) \approx \sqrt{\frac{2\pi\|\mathbf{x} - \mathbf{x}_0\|}{j\frac{\omega}{c}}} G_{3D}^f(\mathbf{x}|\mathbf{x}_0, \omega)$$

- **Allows to model a 2D sound field not by line but by point sources** => desired setup for WFS in a horizontal plane generated by loudspeakers (point source):

$$P_1(\mathbf{x}, \omega) = - \oint_{\partial L} 2 G_{2D}^f(\mathbf{x}|\mathbf{x}_0, \omega) \frac{\partial}{\partial \mathbf{n}} P_1(\mathbf{x}_0, \omega) d\mathbf{x}_0 \quad \mathbf{x} \in L \quad \text{2 dimensional}$$

$$P_1(\mathbf{x}, \omega) \approx - \oint_{\partial L} 2 \sqrt{\frac{2\pi\|\mathbf{x} - \mathbf{x}_0\|}{j\frac{\omega}{c}}} G_{3D}^f(\mathbf{x}|\mathbf{x}_0, \omega) \frac{\partial}{\partial \mathbf{n}} P_1(\mathbf{x}_0, \omega) d\mathbf{x}_0 \quad \mathbf{x} \in L$$

## □ The concept of driving signals / functions:

$$P_1(\mathbf{x}, \omega) \approx - \oint_{\partial L} 2 \sqrt{\frac{2 \pi \|\mathbf{x} - \mathbf{x}_0\|}{j \frac{\omega}{c}}} G_{3D}^f(\mathbf{x}|\mathbf{x}_0, \omega) \frac{\partial}{\partial \mathbf{n}} P_1(\mathbf{x}_0, \omega) d\mathbf{x}_0 \quad \mathbf{x} \in L$$

$$\approx - \oint_{\partial L} G_{3D}^f(\mathbf{x}|\mathbf{x}_0, \omega) D(\mathbf{x}|\mathbf{x}_0, \omega) d\mathbf{x}_0$$

$$\text{Driving signal: } D(\mathbf{x}|\mathbf{x}_0, \omega) = 2 \sqrt{\frac{2 \pi \|\mathbf{x} - \mathbf{x}_0\|}{j \frac{\omega}{c}}} \frac{\partial}{\partial \mathbf{n}} P_1(\mathbf{x}_0, \omega)$$

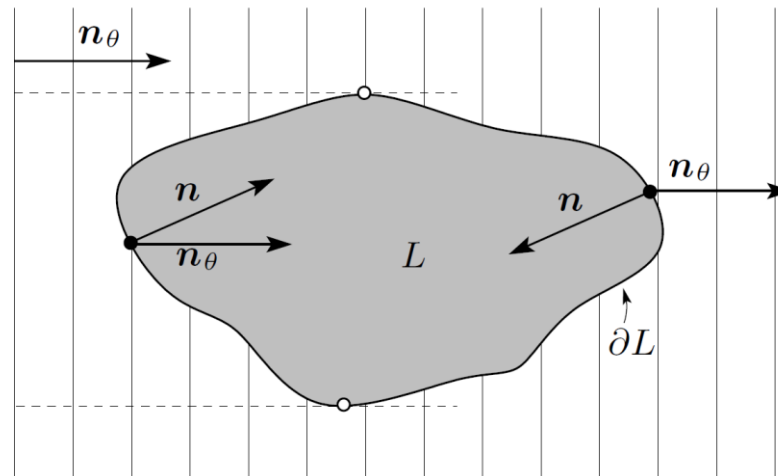
## □ Driving signals:

Signals of point sources to generate a sound field, i.e., the driving signals define the WFS loudspeaker signals.

- Obtaining driving signals from a recorded wave field:  
=> **Data base rendering**

# WFS: Boundary conditions

- ❑ **Model:** Generating a sound field by the superposition of plane waves  
Plane wave with direction  $\mathbf{n}_\theta$



- ❑ Monopoles on the right section emanate into  $L$  in the opposite propagation direction than the plane wave.
- ❑ The Kirchhoff-Helmholtz equation with monopoles & dipoles avoid this problem (no sound outside  $L$ ). With monopoles only => Countermeasures necessary!

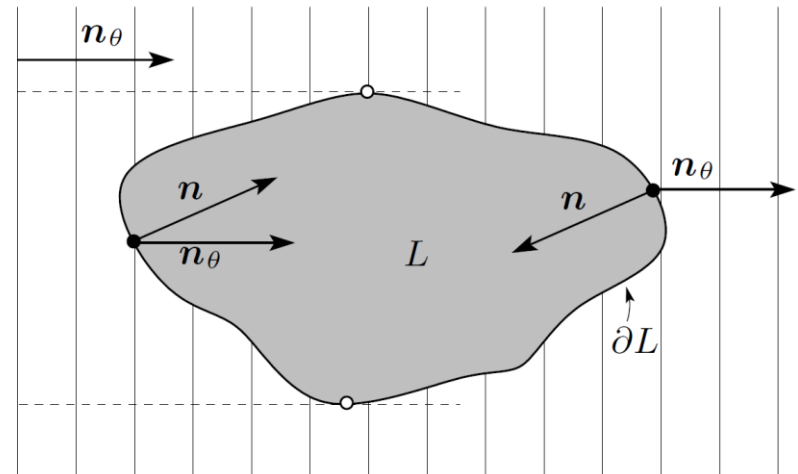
# WFS: Boundary conditions

## Realized countermeasure:

Multiplication with a window function:

$$a(\mathbf{x}) = \begin{cases} 1 & \langle \mathbf{n}, \mathbf{n}_\theta \rangle > 0 \\ 0 & \text{else} \end{cases}$$

=> Sources with propagation components opposite to the plane wave propagation are set to zero.



## Results in a modified driving function for monopoles in 2D:

$$D(\mathbf{x}|\mathbf{x}_0, \omega) = 2 a(\mathbf{x}_0) \sqrt{\frac{2 \pi \|\mathbf{x} - \mathbf{x}_0\|}{j \frac{\omega}{c}}} \frac{\partial}{\partial \mathbf{n}} P_1(\mathbf{x}_0, \omega)$$

## □ Plane wave:

$$S_{pw}(\mathbf{x}, \omega) = \hat{S}_{pw}(\omega) e^{-j\frac{\omega}{c} \mathbf{n}_{pw}^T \mathbf{x}}$$

$$\frac{\partial}{\partial \mathbf{n}} S_{pw}(\mathbf{x}, \omega) = -j\frac{\omega}{c} \hat{S}_{pw}(\omega) e^{-j\frac{\omega}{c} \mathbf{n}_{pw}^T \mathbf{x}} \mathbf{n}_{pw}^T \mathbf{x}$$

## □ resulting in the following driving function:

$$D(\mathbf{x}|\mathbf{x}_0, \omega) = -2 a_{pw}(\mathbf{x}_0) \sqrt{\frac{2\pi \|\mathbf{x} - \mathbf{x}_0\|}{j\frac{\omega}{c}}} \left[ j\frac{\omega}{c} \hat{S}_{pw}(\omega) \mathbf{n}_{pw}^T \mathbf{x}_0 \right] e^{-j\frac{\omega}{c} \mathbf{n}_{pw}^T \mathbf{x}_0}$$

## □ Spherical wave:

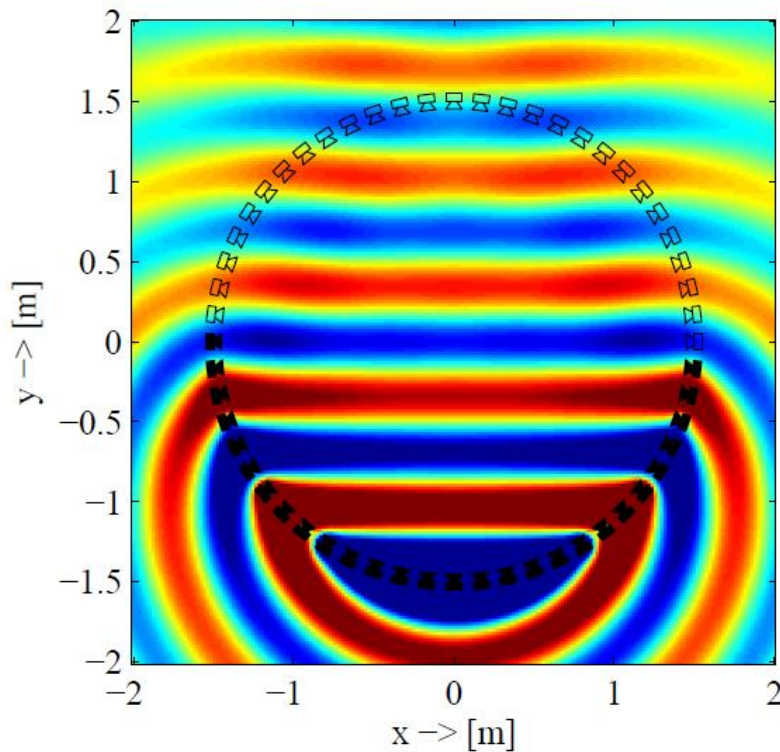
$$S_{sw}(\mathbf{x}, \omega) = \hat{S}_{sw}(\omega) \frac{e^{-j\frac{\omega}{c}\|\mathbf{x}-\mathbf{x}_s\|}}{\|\mathbf{x}-\mathbf{x}_s\|}$$

$$\frac{\partial}{\partial \mathbf{n}} S_{sw}(\mathbf{x}, \omega) = -\frac{[\mathbf{x}-\mathbf{x}_s]^T \mathbf{n}(\mathbf{x})}{\|\mathbf{x}-\mathbf{x}_s\|^2} \left[ \frac{1}{\|\mathbf{x}-\mathbf{x}_s\|} + \frac{j\omega}{c} \right] \hat{S}_{sw}(\omega) e^{-j\frac{\omega}{c}\|\mathbf{x}-\mathbf{x}_s\|}$$

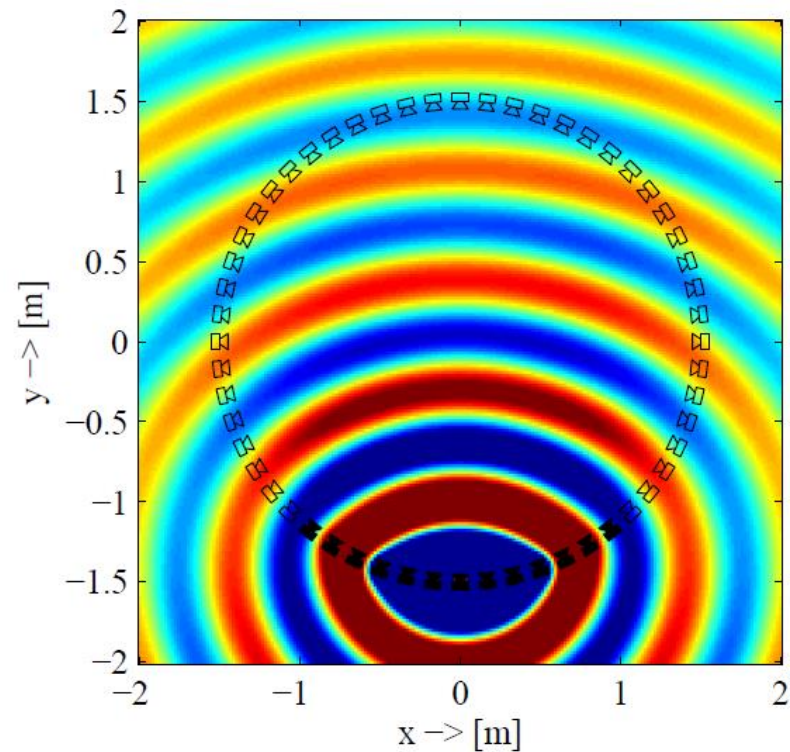
## □ resulting in the following driving function:

$$D(\mathbf{x}|\mathbf{x}_0, \omega) = -2 a_{sw}(\mathbf{x}_0) \sqrt{\frac{2\pi\|\mathbf{x}-\mathbf{x}_0\|}{j\frac{\omega}{c}}} \frac{[\mathbf{x}_0-\mathbf{x}_s]^T \mathbf{n}(\mathbf{x}_0)}{\|\mathbf{x}_0-\mathbf{x}_s\|^2} \left[ \frac{1}{\|\mathbf{x}_0-\mathbf{x}_s\|} + \frac{j\omega}{c} \right] \hat{S}_{sw}(\omega) e^{-j\frac{\omega}{c}\|\mathbf{x}_0-\mathbf{x}_s\|}$$

## □ Plane & Spherical waves generated with a circular loudspeaker array:



(a) plane wave ( $f_{\text{pw}} = 500$  Hz,  $\mathbf{n}_{\text{pw}} = [0 \ 1]^T$ )



(b) spherical wave ( $f_{\text{sw}} = 500$  Hz,  $\mathbf{x}_S = [0 \ -2]^T$  m)



## WFS: Dependency of the driving function

- Driving function (signals of the WFS loudspeakers) depend on the loudspeaker position  $\mathbf{x}_0$  and the **listener position**  $\mathbf{x}$  within the plane  $L$ .

$$D(\mathbf{x}|\mathbf{x}_0, \omega)$$

- However, the loudspeaker signals should be independent of the listeners' position (otherwise a location tracking of listeners would be necessary!).
- Fortunately, the dependence on the listeners position is rather mild.
- Typically, the driving function is fixed for a fixed listener's position:

$$D_L(\mathbf{x}_0, \omega) = D(\mathbf{x}_L|\mathbf{x}_0, \omega)$$

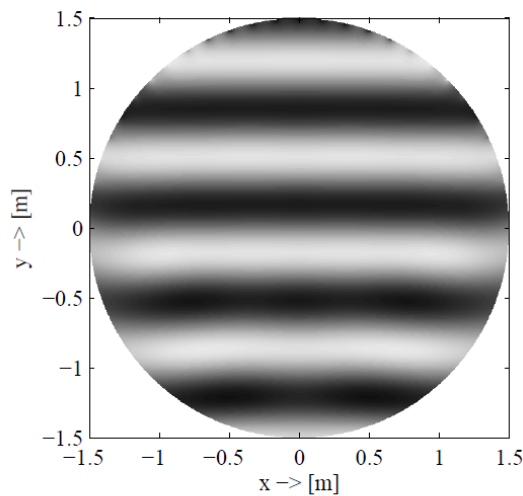
- In case the listener is close to the loudspeakers => rather big error.

- ❑ WFS is generated with loudspeakers at discrete positions.
- ❑ => Spatial aliasing occurs. However, the analysis of the aliasing is rather complicated and dependent on the specific geometry / aperture of the WFS system.
- ❑ In general:
  - ❑ Spatial aliasing increases with increasing frequency, i.e., bandwidth
  - ❑ Ex.: For typical loudspeaker distances of 10-30 cm => Aliasing > 1kHz
  - ❑ However, listeners are not too sensitive to spatial aliasing, typically a spatial coloration occurs.
  - ❑ Pre-filtering with the term  $\sqrt{\frac{j\omega}{c}}$  in the driving functions should only be performed below the spatial aliasing frequency (~1 kHz).
  - ❑ WFS systems with monopoles generate sound field outside L or V => in case of a WFS in a room => reflections from the wall generate artifacts.

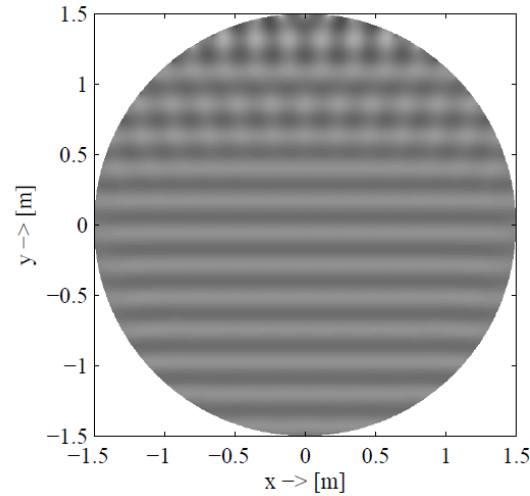
# WFS: Spatial aliasing: Plane waves generated by circ. array



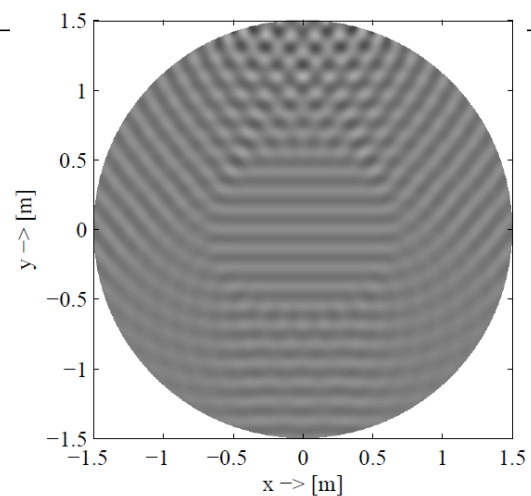
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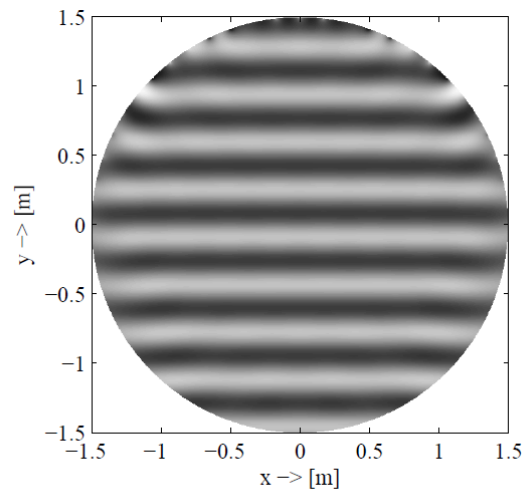
(a) WFS ( $f_{pw} = 500$  Hz)



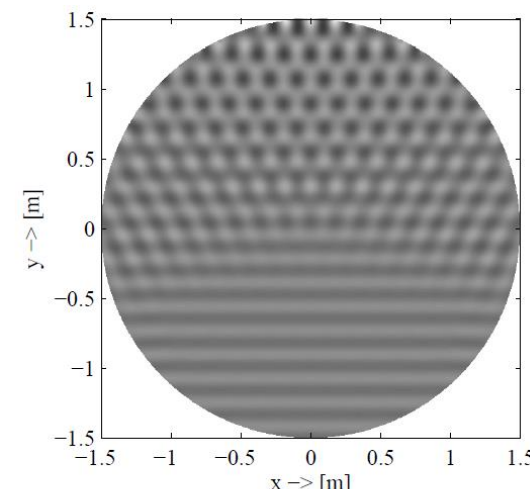
(e) WFS ( $f_{pw} = 1500$  Hz)



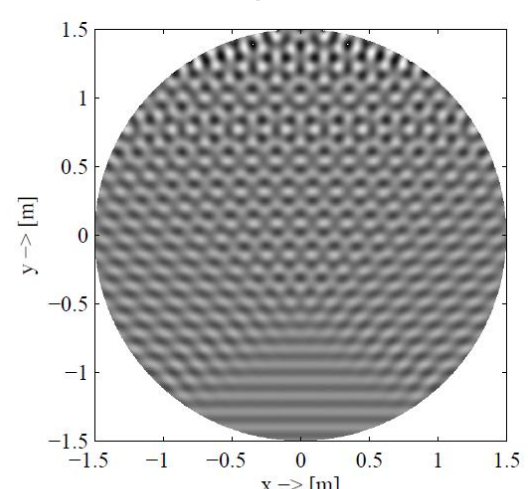
(d) HOA ( $f_{pw} = 2500$  Hz)



(c) WFS ( $f_{pw} = 1000$  Hz)



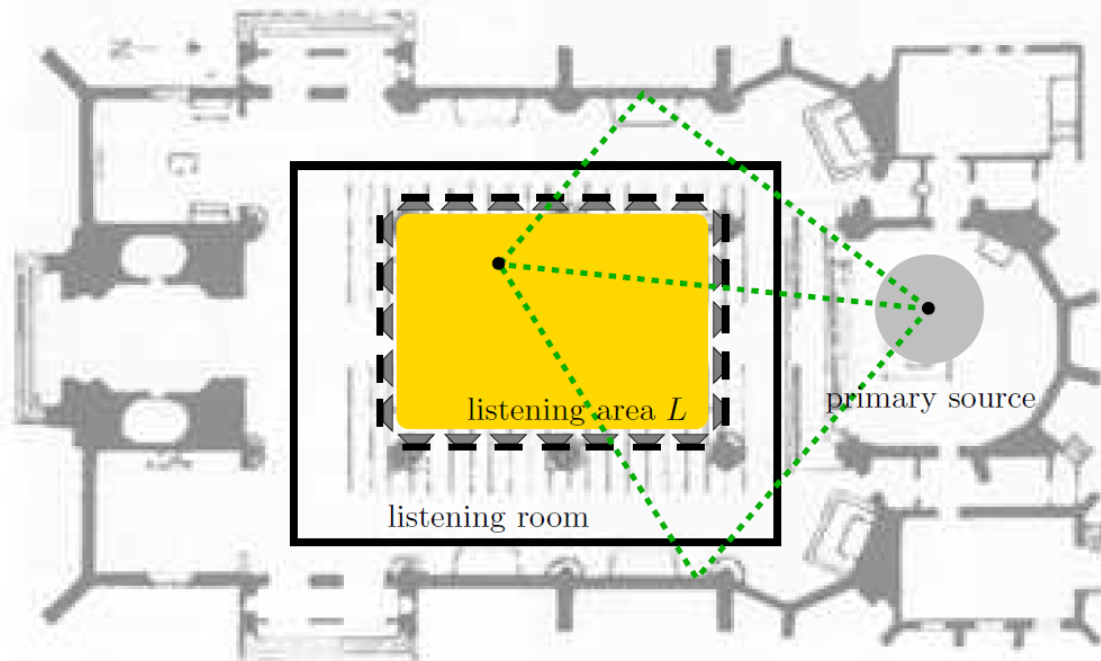
(a) WFS ( $f_{pw} = 2000$  Hz)



(e) WFS ( $f_{pw} = 3000$  Hz)

# WFS: Example of virtual acoustics

- Reproduction of a sound field of a church:
  - Virtual scene model is based on the decomposition of the sound signal in the church into plane waves.
  - Image model of reverberation: Sound in the listening area is a multitude of point sources. => Superposition of plane waves.





# Higher Order Ambisonics

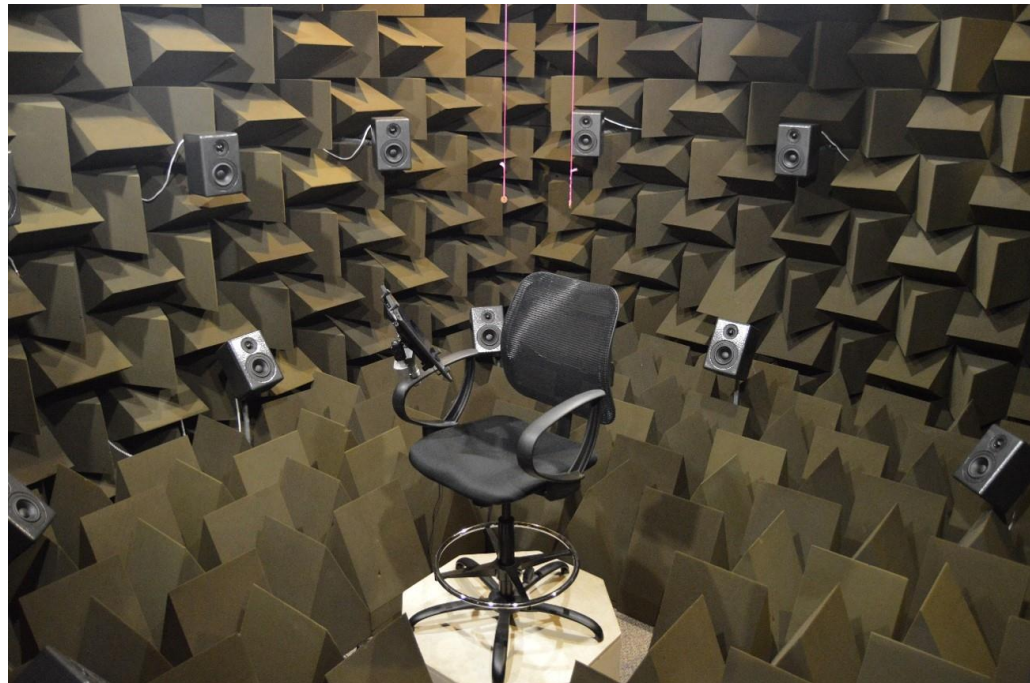


# HOA: Examples



Source: <https://www.southampton.ac.uk>

Source: <http://sites.psu.edu/spral/current-projects/auras/>



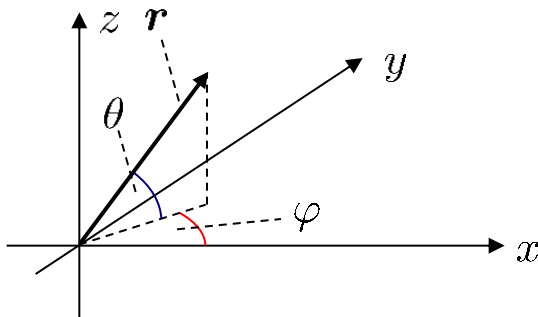
## □ The basic idea:

- Decomposition of a wave field into spherical modes.
- Performing a transform into these modes.
- Mode description independent of the recording (microphone) or the generation (loudspeaker) setup.
- Order determines the precision of the decomposition / generation.
- For an order  $M$  there are  $(M+1)^2$  microphones / loudspeakers necessary for its recording / generation.
- The setup is generally used to reproduce a 3D sound field precisely at a sweet spot within a loudspeaker setup.
- The higher the order is, the bigger is the sweet spot or the max. frequency given the radius of the target sphere.



# HOA: Theoretical sound field decomposition

- Definition of a subspace  $\Omega_1$  where the sources are located and a subspace  $\Omega_2$  where no sources are located (listening area).
- General setup: All sources outside a radius,  $R_2$  i.e., the listening area is inside.



- Comparable to sinus components for the Fourier transform, the sound field can be decomposed into “spherical harmonics”

# HOA: Theoretical sound field decomposition



- The decomposition is expressed as follows:

$$p(\mathbf{r}, \omega) = \sum_{m=0}^{\infty} i^m j_m(kr) \sum_{n=0}^m \sum_{\sigma=\pm 1} B_{mn}^{\sigma}(\omega) Y_{mn}^{\sigma}(\varphi, \theta) \quad r = |\mathbf{r}| \quad k = \frac{\omega}{c}$$

HOA components  
Spherical functions

- with the following spherical functions:

$j_m(kr)$  Spherical Bessel function of first kind

$$Y_{mn}^{\sigma}(\varphi, \theta) = \sqrt{(2m+1) \epsilon_n \frac{(m-n)!}{(m+n)!}} P_{mn}(\sin \theta) \text{sc}(n \varphi)$$

$$\text{sc}(n \varphi) = \begin{cases} \cos(n \varphi) & : \sigma = +1 \\ \sin(n \varphi) & : \sigma = -1 \end{cases} \quad \epsilon_n = \begin{cases} 1 & : n = 0 \\ 2 & : n > 0 \end{cases}$$

$$P_{mn}(\sin \theta) = \frac{d^n P_m(\sin \theta)}{d(\sin \theta)^n}$$

Legendre function n. order, m. degree

$$P_n(z) = \frac{1}{2^n n!} \frac{d^n}{dz^n} (z^2 - 1)^n$$

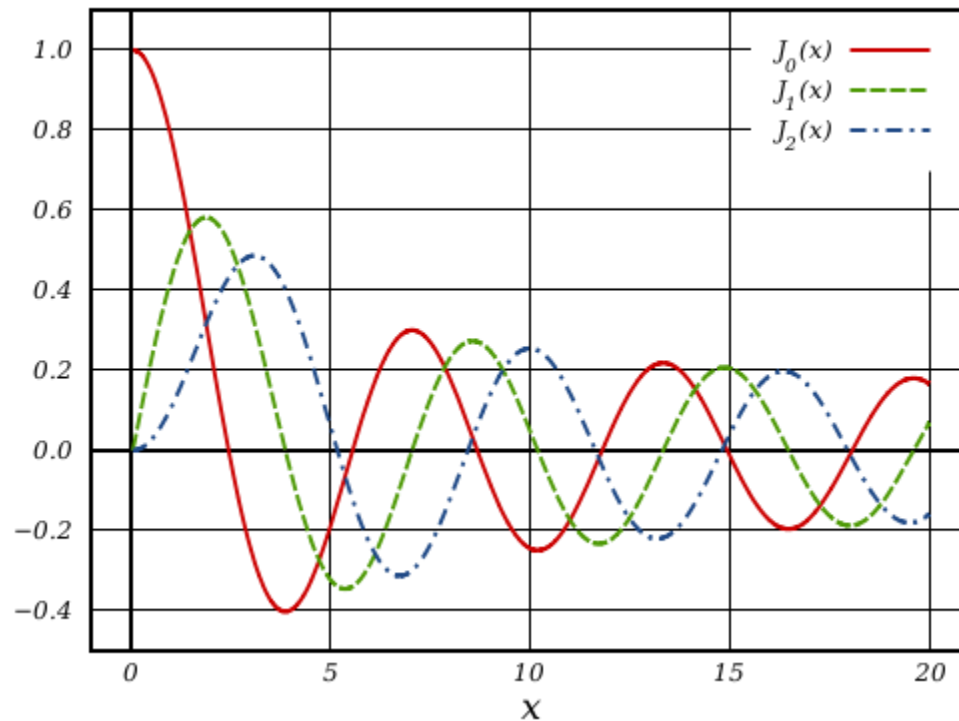
Legendre function n. order, 1st degree

- and the HOA components  $B_{mn}^{\sigma}(\omega)$  i.e., a universal format to encode a sound field.

# HOA: Spherical Bessel functions

$j_m(kr)$  Spherical Bessel function  
of first kind

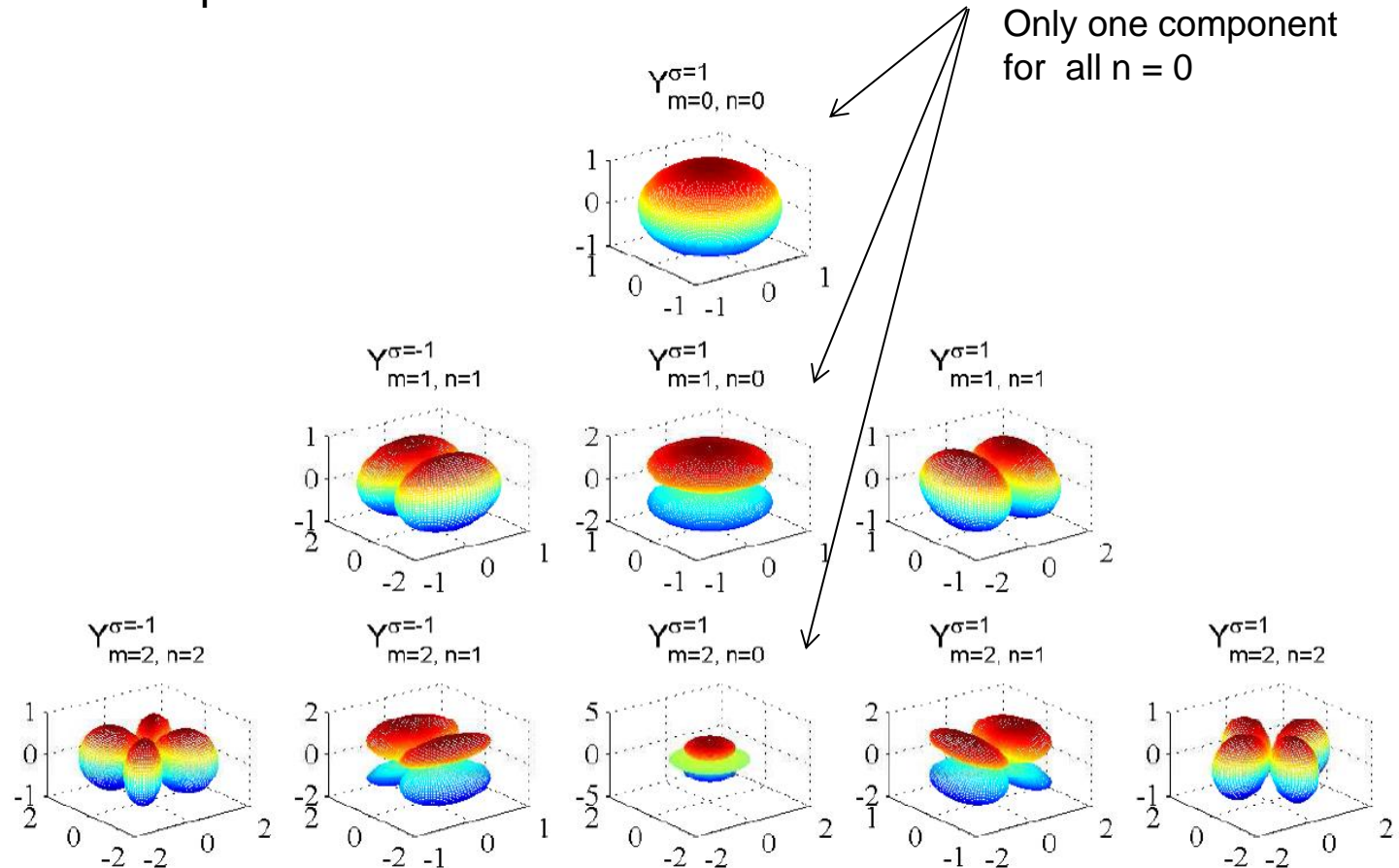
$$j_m(x) = \frac{1}{2\pi} \int_{-\pi}^{\pi} e^{i(m\tau - x \sin \tau)} d\tau$$



Source: Wikipedia

# HOA: Theoretical sound field decomposition

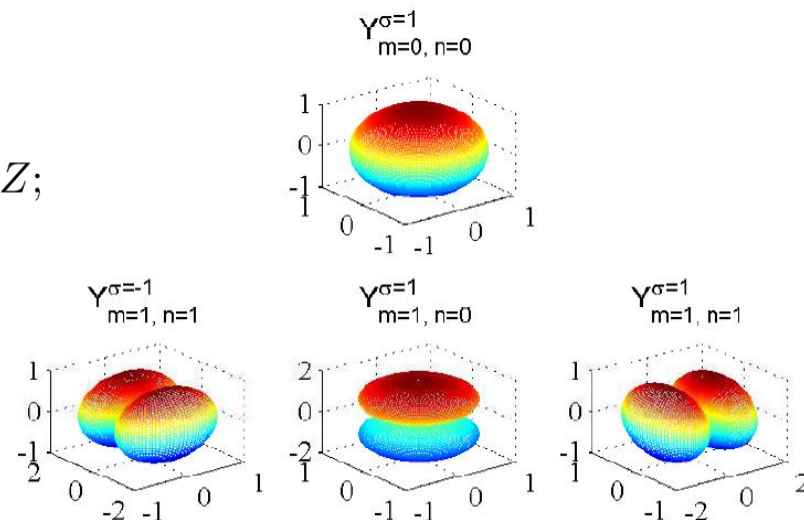
## Plot of the spherical functions:



# HOA: Theoretical sound field decomposition

- Truncation of the infinite sum to  $M = 1$ , results in the well-known B-Format according to Gerzon, i.e., one onmi and three dipole components:

$$B_{00}^1 = W; \quad B_{11}^1 = X; \quad B_{11}^{-1} = Y; \quad B_{10}^1 = Z;$$



- When limiting the sum, a truncation / reconstruction error occurs. The error can be assumed to be negligible for  $kr \leq M$

=> Sweet spot:

$$r_{\max} = \frac{M c}{\omega_0}$$

Given the radius of sphere of targeted reconstruction, the reconstruction is only valid up to:

$$\omega_{\max} = \frac{M c}{r_0}$$

- The HOA components can in general be calculated as follows:

$$B_{mn}^{\sigma}(\omega) = EQ(k r_M) \frac{1}{4\pi r_M^2} \int_{\varphi=0}^{2\pi} \int_{\theta=-\frac{\pi}{2}}^{\frac{\pi}{2}} p(r_M, \varphi, \theta, \omega) Y_{mn}^{\sigma}(\varphi, \theta) \cos \theta d\theta d\varphi$$

$$EQ(k r_M) = \frac{1}{i^m j_m(k r_M)} \quad : \text{normalization}$$

$$p(r_M, \varphi, \theta, \omega) \quad : \text{recorded sound field over a sphere with radius } r_M$$

- Important to note: The HOA components are totally independent of the recording setup, including the radius of the recorded sphere.

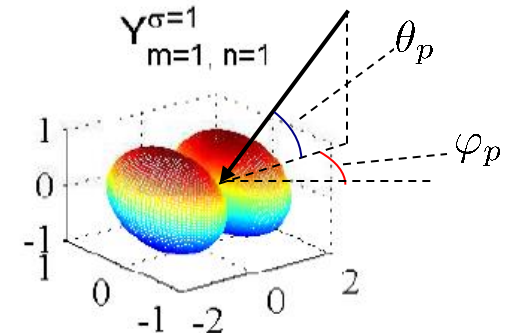
## HOA: Encoding for specific cases

- For specific cases, i.e., plane and spherical waves the HOA components can be directly calculated based on the spherical functions:  $Y_{mn}^{\sigma}(\varphi, \theta)$

- 1) Plane waves arriving from a direction  $\varphi_p, \theta_p$  with amplitude  $O_p$

$$B_{mn}^{\sigma}(\omega) = \frac{O_p}{4\pi} Y_{mn}^{\sigma}(\varphi_p, \theta_p)$$

$B_{mn}^{\sigma}(\omega)$  can be directly calculated by evaluating  $Y_{mn}^{\sigma}(\varphi_p, \theta_p)$  at the direction of arrival of the plane wave.



- 2) Spherical waves where the point of origin  $\mathbf{r}_s = (r_s, \varphi_s, \theta_s)$  is placed outside of the listening area with radius  $R_2$ , i.e.  $r_s > R_2$  with the amplitude  $O_s$

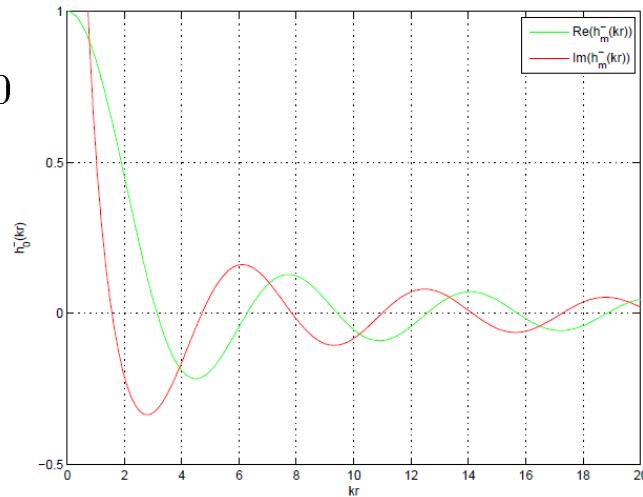
$$B_{mn}^{\sigma}(\omega) = \frac{O_s}{4\pi} i^{-(m+1)} \frac{h_m^-(k r_s)}{k} Y_{mn}^{\sigma}(\varphi_s, \theta_s)$$

$h_m^-(k r_s)$  : spherical Hankel functions of second kind:  $h_m^-(k r_s) = H_m^{(2)}(k r_s)$

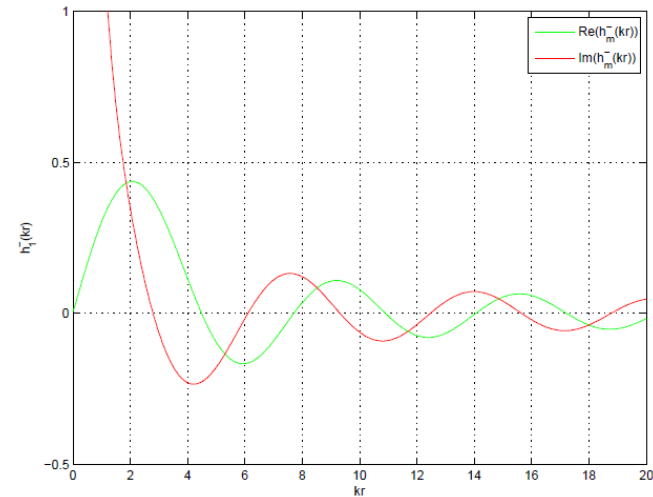


# HOA: Spherical Hankel functions of second kind

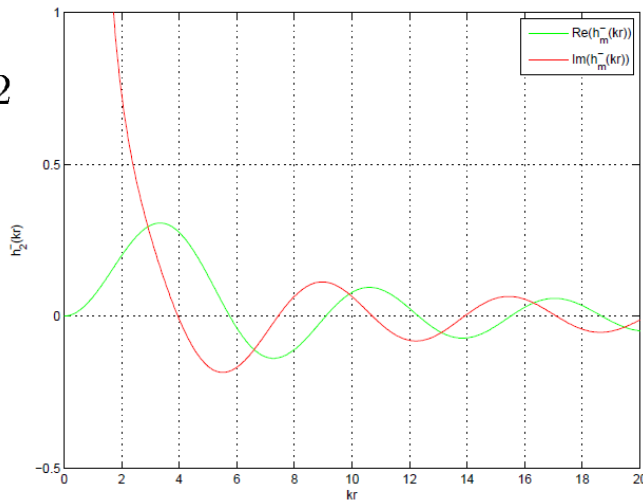
$m = 0$



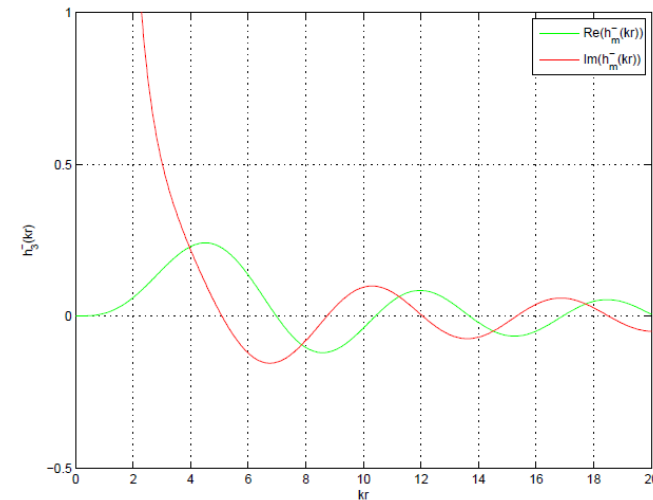
$m = 1$



$m = 2$



$m = 3$



# HOA: Encoding by spatial sampling

- ❑ **Task:** Record a sound field and calculate the HOA components
- ❑ **Concept:** Using  $N_M$  microphones positioned on a sphere with radius  $r_M$ , where  $q$  is the position of the  $q^{\text{th}}$  microphone.

$$p_q(\omega) = p(r_M, \varphi_q, \theta_q, \omega)$$
$$p_q(\omega) = \sum_{m=0}^{\infty} i^m j_m(k r_M) \sum_{n=0}^m \sum_{\sigma=\pm 1} B_{mn}^{\sigma}(\omega) Y_{mn}^{\sigma}(\varphi_q, \theta_q)$$



# HOA: Encoding by spatial sampling

- For  $N_M$  microphones one can obtain  $N_M$  equations; approximating the sound field up to order  $M$ .

$$p_q(\omega) = \sum_{m=0}^M i^m j_m(k r_M) \sum_{n=0}^m \sum_{\sigma=\pm 1} B_{mn}^{\sigma}(\omega) Y_{mn}^{\sigma}(\varphi_q, \theta_q) \quad N_M \geq (M + 1)^2$$

- In Matrix vector notation these equations can be written as follows:

$$\mathbf{p} = \mathbf{Y}_M \mathbf{W}_M \mathbf{b}$$

with:

$$\mathbf{Y}_M = \begin{bmatrix} Y_{00}^1(\varphi_1, \theta_1) & Y_{10}^1(\varphi_1, \theta_1) & \cdots & Y_{MM}^{-1}(\varphi_1, \theta_1) \\ Y_{00}^1(\varphi_2, \theta_2) & Y_{10}^1(\varphi_2, \theta_2) & \cdots & Y_{MM}^{-1}(\varphi_2, \theta_2) \\ \vdots & \vdots & \ddots & \vdots \\ Y_{00}^1(\varphi_{N_M}, \theta_{N_M}) & Y_{10}^1(\varphi_{N_M}, \theta_{N_M}) & \cdots & Y_{MM}^{-1}(\varphi_{N_M}, \theta_{N_M}) \end{bmatrix}$$

$$\mathbf{b} = [B_{00}^1(\omega) \quad B_{10}^1(\omega) \quad B_{11}^1(\omega) \quad B_{11}^{-1}(\omega) \quad \cdots \quad B_{MM}^{-1}(\omega)]^T$$

$$\mathbf{W}_M = \text{diag} \{ [j_0(k r_M) \quad i^1 j_1(k r_M) \quad i^1 j_1(k r_M) \quad i^1 j_1(k r_M) \cdots i^M j_M(k r_M)] \}$$

## □ Size analysis of vectors and matrices:

$$\mathbf{p} = \mathbf{Y}_M \mathbf{W}_M \mathbf{b}$$

Diagram illustrating the dimensions of the vectors and matrices in the equation  $\mathbf{p} = \mathbf{Y}_M \mathbf{W}_M \mathbf{b}$ :

- $\mathbf{p}$  is a vector of length  $(M + 1)^2$ .
- $\mathbf{W}_M$  is a matrix of size  $(M + 1)^2 \times (M + 1)^2$ .
- $\mathbf{Y}_M$  is a matrix of size  $N_M \times (M + 1)^2$ .

## □ Least squares solutions:

$$\mathbf{W}_M \hat{\mathbf{b}} = (\mathbf{Y}_M^T \mathbf{Y}_M)^{-1} \mathbf{Y}_M^T \mathbf{p}$$

$$\Rightarrow \hat{\mathbf{b}} = \mathbf{E}_M (\mathbf{Y}_M^T \mathbf{Y}_M)^{-1} \mathbf{Y}_M^T \mathbf{p}$$

With the diagonal normalization matrix:

$$\mathbf{E}_M = \text{diag} \left\{ \left[ \frac{1}{j_o(k r_M)} \quad \frac{1}{i^1 j_1(k r_M)} \quad \frac{1}{i^1 j_1(k r_M)} \quad \frac{1}{i^1 j_1(k r_M)} \cdots \frac{1}{i^M j_M(k r_M)} \right] \right\}$$

## HOA: Encoding by spatial sampling

- If the spatial distribution of the microphones fulfills the orthonormality property:

$$(\mathbf{Y}_M^T \mathbf{Y}_M) = \mathbf{1}$$

$$\hat{\mathbf{b}} = \mathbf{E}_M (\mathbf{Y}_M^T \mathbf{Y}_M)^{-1} \mathbf{Y}_M^T \mathbf{p} \Rightarrow \hat{\mathbf{b}} = \mathbf{E}_M \mathbf{Y}_M^T \mathbf{p}$$

- which results in the discrete version of the general coding equation:

$$B_{mn}^\sigma(\omega) = \frac{1}{j_m(k r_M)} \frac{1}{N_M} \sum_{q=0}^{N_M} p_q(\omega) Y_{mn}^\sigma(\varphi_q, \theta_q)$$

- Reminder: general coding equation:

$$B_{mn}^\sigma(\omega) = EQ(k r_M) \frac{1}{4\pi r_M^2} \int_{\varphi=0}^{2\pi} \int_{\theta=-\frac{\pi}{2}}^{\frac{\pi}{2}} p(r_M, \varphi, \theta, \omega) Y_{mn}^\sigma(\varphi, \theta) \cos \theta d\theta d\varphi$$

$$EQ(k r_M) = \frac{1}{i^m j_m(k r_M)} : \text{normalization}$$

# HOA: Encoding by spatial sampling

## Thoughts about the recording setup

- The matrix  $\epsilon$  quantifies the “non-orthogonality” of the microphone distribution (i.e., recording setup, e.g., the „Eigenmike“):

$$\epsilon = \mathbf{1} - \frac{1}{N_M} (\mathbf{Y}_M^T \mathbf{Y}_M)$$

$$p_q(\omega) = \sum_{m=0}^{\infty} i^m j_m(k r_M) \sum_{n=0}^m \sum_{\sigma=\pm 1} B_{mn}^{\sigma}(\omega) Y_{mn}^{\sigma}(\varphi_q, \theta_q)$$

- Lower spatial aliasing with **reduced radius**  $r_M$  :

The Bessel functions act as a low-frequency filter which reduces spatial sampling like an anti-aliasing filter.

The cutoff frequency increases with  $r_M$ . Therefore, decreasing the radius will minimize the spatial aliasing.

- Better conditioned inverse matrix with **increased radius**  $r_M$ :

$$(\mathbf{Y}_M^T \mathbf{Y}_M)$$

- **HOA decoding task:** Calculation of the loudspeaker signals of an HOA loudspeaker array with  $N_L$  loudspeakers.

- Loudspeaker positions:

$$\mathbf{r}_{L,l} = (r_{L,l}, \varphi_{L,l}, \theta_{L,l})$$

- Sound pressure of the  $l$ -th loudspeaker at position  $\mathbf{r}$  where  $s_l(\omega)$  is the driving signal:  $s_l(\omega) p_l(\mathbf{r}, \omega)$

- HOA decomposition of the signal radiated by the  $l$ -th loudspeaker signal. I.e.,  $L_{l,mn}^\sigma(\omega)$  is known by the measurement of the loudspeaker array.

$$p_l(\mathbf{r}, \omega) = \sum_{m=0}^{\infty} i^m j_m(kr) \sum_{n=0}^m \sum_{\sigma=\pm 1} L_{l,mn}^\sigma(\omega) Y_{mn}^\sigma(\varphi_q, \theta_q) \quad \leftarrow \text{1 loudspeaker}$$

$$\begin{aligned} p_{\text{LS}}(\mathbf{r}, \omega) &= \sum_{l=1}^{N_L} s_l(\omega) p_l(\mathbf{r}, \omega) \\ &= \sum_{l=1}^{N_L} s_l(\omega) \sum_{m=0}^{\infty} i^m j_m(kr) \sum_{n=0}^m \sum_{\sigma=\pm 1} L_{l,mn}^\sigma(\omega) Y_{mn}^\sigma(\varphi_q, \theta_q) \quad \leftarrow \text{All loudspeakers} \end{aligned}$$

## □ Mode matching

Target: calculate the excitation signal  $s_l(\omega)$  of the loudspeakers in order to approximate the sound field given by an HOA description  $B_{mn}^\sigma(\omega)$

$$p(\mathbf{r}, \omega) = \sum_{m=0}^{\infty} i^m j_m(kr) \sum_{n=0}^m \sum_{\sigma=\pm 1} B_{mn}^\sigma(\omega) Y_{mn}^\sigma(\varphi, \theta) \quad \leftarrow \text{General description}$$

$$p_{\text{LS}}(\mathbf{r}, \omega) = \sum_{m=0}^{\infty} i^m j_m(kr) \sum_{n=0}^m \sum_{\sigma=\pm 1} \sum_{l=1}^{N_L} s_l(\omega) L_{l,mn}^\sigma(\omega) Y_{mn}^\sigma(\varphi_q, \theta_q) \quad \leftarrow \text{LS description}$$

$$\Rightarrow B_{mn}^\sigma(\omega) \stackrel{!}{=} \sum_{l=1}^{N_L} s_l(\omega) L_{l,mn}^\sigma(\omega) \quad \text{equal to: } \mathbf{b} = \mathbf{L} \mathbf{s}$$

$$\mathbf{L} = \begin{bmatrix} L_{1,00}(\varphi_1, \theta_1) & L_{2,00}(\varphi_2, \theta_2) & \cdots & L_{N_L,00}(\varphi_{N_L}, \theta_{N_L}) \\ L_{1,10}(\varphi_1, \theta_1) & L_{2,10}(\varphi_2, \theta_2) & \cdots & L_{N_L,10}(\varphi_{N_L}, \theta_{N_L}) \\ \vdots & \vdots & \ddots & \vdots \\ L_{1,MM}(\varphi_1, \theta_1) & L_{2,MM}(\varphi_2, \theta_2) & \cdots & L_{N_L,MM}(\varphi_{N_L}, \theta_{N_L}) \end{bmatrix}$$

$$\mathbf{s} = [s_1(\omega) \ s_2(\omega) \ \cdots \ s_{N_L}(\omega)]^T \quad : \text{excitation signal vector}$$

$$\mathbf{b} = [B_{00}^1(\omega) \ B_{10}^1(\omega) \ B_{11}^1(\omega) \ B_{11}^{-1}(\omega) \ \cdots \ B_{MM}^{-1}(\omega)]^T$$



- Size analysis of vectors and matrices:

$$\mathbf{b} = \mathbf{L} \mathbf{s} \leftarrow \text{Vector of length } N_L$$

↑  
Matrix of size  $(M + 1)^2 \times N_L$

- Least squares solutions:

$$\hat{\mathbf{s}} = \mathbf{L}^T (\mathbf{L} \mathbf{L}^T)^{-1} \mathbf{b}$$

- Decoding matrix:

$$\mathbf{D} = \mathbf{L}^T (\mathbf{L} \mathbf{L}^T)^{-1}$$

- Regular loudspeaker distribution on a sphere of radius  $r_L$
- Regular distribution in case the following equation is fulfilled:  $(\mathbf{L}^T \mathbf{L}) = \mathbf{1}$
- Then the decoding matrix reduces to  $\mathbf{D} = \mathbf{L}^T$

□ Special cases for loudspeakers radiating specific types of waves

□ Spherical waves:

$$\begin{aligned} \mathbf{L} &= \mathbf{W}_L^T \mathbf{Y}_L \\ \hat{\mathbf{s}} &= \left( \mathbf{W}_L^T \mathbf{Y}_L \right)^T \mathbf{b} \end{aligned} \quad \mathbf{Y}_L = \begin{bmatrix} Y_{00}^1(\varphi_1, \theta_1) & Y_{00}^1(\varphi_2, \theta_2) & \cdots & Y_{00}^1(\varphi_{N_L}, \theta_{N_L}) \\ Y_{10}^1(\varphi_1, \theta_1) & Y_{10}^1(\varphi_2, \theta_2) & \cdots & Y_{10}^1(\varphi_{N_L}, \theta_{N_L}) \\ \vdots & \vdots & \ddots & \vdots \\ Y_{MM}^{-1}(\varphi_1, \theta_1) & Y_{MM}^{-1}(\varphi_2, \theta_2) & \cdots & Y_{MM}^{-1}(\varphi_{N_L}, \theta_{N_L}) \end{bmatrix}$$

$$\mathbf{W}_L = \text{diag} \left\{ \left[ -i \quad -\frac{h_1^-(kr_L)}{k} i^{-M-1} \quad -\frac{h_M^-(kr_L)}{k} \right] \right\}$$

$$w_m = i^{-(m+1)} \frac{h_m^-(kr_s)}{k}$$

□ Plane waves:

$$\begin{aligned} \mathbf{L} &= \mathbf{Y}_L \\ \hat{\mathbf{s}} &= \mathbf{Y}_L^T \mathbf{b} \end{aligned}$$

- ❑ Introduction of WFS and HOA for sound field generation of loudspeaker arrays.
- ❑ Typically specific fields of applications:
- ❑ HOA: 3D sound in a sweet spot  
=> more precise reproduction
- ❑ WFS: 2D in larger plane => listening area  
=> more natural sound experience

# References

- [1] R. Nicol , “Sound spatialization by higher order ambisonics: Encoding and decoding a sound scene in practice from a theoretical point of view,” in *Proceedings of the 2nd International Symposium on Ambisonics and Spherical Acoustics*, Paris, France (May 6–7, 2010)
- [2] S. Spors, J. Ahrens. "A comparison of wave field synthesis and higher-order ambisonics with respect to physical properties and spatial sampling." Audio Engineering Society Convention 125. Audio Engineering Society, 2008.
- [3] S. Spors, R. Rabenstein, and Jens Ahrens. "The theory of wave field synthesis revisited." 124th AES Convention. 2008.
- [4] R. Rabenstein, S. Spors, and P. Steffen. "Wave field synthesis techniques for spatial sound reproduction." *Topics in Acoustic Echo and Noise Control* 5 (2006): 517-545.
- [5] A. Sontacchi. Dreidimensionale Schallfeldreproduktion für Lautsprecher-und Kopfhöreranwendungen. Dissertation, Uni Graz, 2003.

- ❑ **In case of questions:**
  - ❑ Please contact me by eMail.
  
- ❑ **Lecture in Summer Term:** „Adaptive Filters“:  
Theory and applications