

ICASSP Tutorial T1: Personalising sound over loudspeakers

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Introduction to personalising sound over loudspeakers

Why sound?

- In evolutionary terms, **sensing** of sound and vibration pre-dates vision
- Hearing provides **connection** to the world around us, through the grapevine, the jungle drum, the market hubbub, the buzz
- It is especially adept at detecting transient events, as with unexpected sounds that turn the head and grab **attention**
- As a truly panoramic sensing modality, we can interpret **immersive** environments



Why personal?

- Increasingly common to multi-task during **shared** experiences
- Even in company, **personal** life continues (alerts, notifications)
- We like **choice** to control our experiences, including selective enhancement, content filters or supplementary feeds
- Personalisation improves autonomy, intimacy, and sense of privacy



- **Object-based audio** (e.g., MPEG-H) offers many personalisation features

Bleidt et al., 2014. Object-based audio: Opportunities for improved listening..., SMPTE

ITU, 2015. ITU-R BS.2076-0: Audio Definition Model

Herre et al., 2015. MPEG-H 3D Audio - The new standard for coding of immersive spatial audio, JSTSP

Coleman et al., 2018. An audio-visual system for object-based audio: from recording to listening, TMM

Why loudspeakers?

- Loudspeakers can be incorporated into the built environment
 - Clean, safe, convenient
- Absence of wearables (c.f. headphones, earphones, headsets)
 - Not weighed down, good comfort and hygiene, free to move, open to alerts
- Creates space for shared experiences
 - Being together in the space, maintains open communication channel
- Provide consistent spatial and timbral quality





What are the practical challenges?

- Sound is invisible!
- Sources spread sound out in all directions
- Air moves and changes temperature, pressure and humidity
- Environments reflect sound all over the place, can be noisy and vary
- Sound diffracts around obstacles and permeates barriers
- Transducers have noise, colouration, non-linear distortion and frequency/power limits
- Listeners are complex, whose perception changes with context





Vision for future ears-free personal sound

The promise of sound is in its power:

- To tell a story
- To transport you to another world
- To envelope you in it
- To transcend the mundane, to evoke emotion, passion and encounter empathy
- To connect with your loved ones and your tribe!
grab our attention

To fulfil this, we need capability:

- To deliver you sound that is reliable, enveloping and tailored to your needs





What will we cover in this tutorial?

- **Formulation**
 - Definitions of personal sound zones and spatial audio
 - Key approaches to manipulate sound fields
 - Measures of performance
- **Engineering**
 - Design parameters
 - Regularization
 - Room effects
 - Practical sound zone filter design
- **Experience**
 - Soundbar listening demonstration
 - Perceptual models
 - Alternative approaches and applications
- Summary of conclusions and perspectives





Definitions: Personal sound zone and spatial audio problems



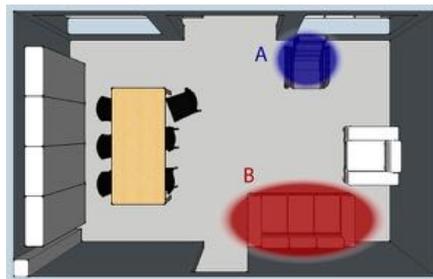


Zone size

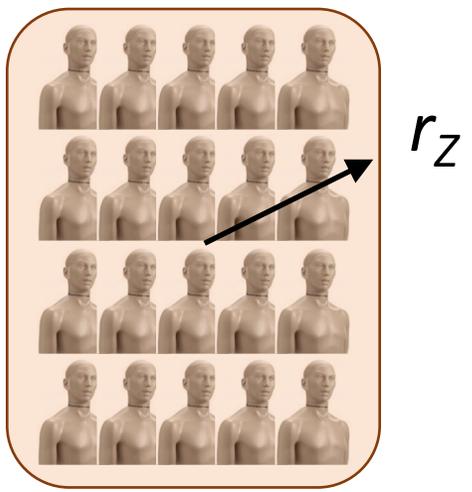
Theatre (*auditorium*)



Lounge (*head*)



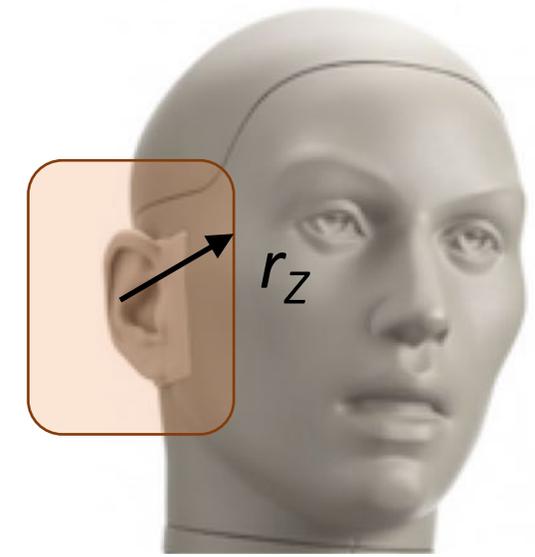
Person (*ear*)



André et al., 2014, IJHCS



Olik et al., 2013, AES conf



Hollebon et al., 2019, AES conf



Zone count

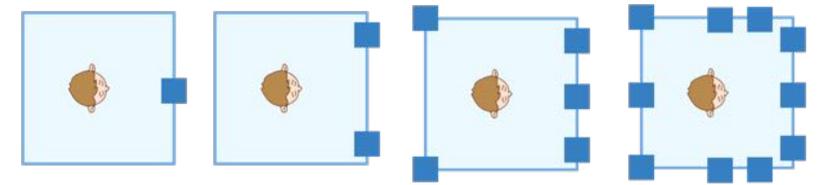
With Z zones, each bright zone has $Z-1$ dark zones:

Bright	Dark
{A}	
{A}	{B}
{A}	{BC}
	⋮
{A}	{BCDEF}



Spatial sound formats

- 1D (distance): mono
- 2D (distance+azimuth): stereo, 5.1 surround
- 3D (distance+azimuth+height): 9.1, 22.2, etc.



- Object-based audio



- Real vs. virtual channels



Rumsey, 2001. Spatial audio

Coleman et al., 2014. Stereophonic personal audio reproduction using planarity control optimization, ICSV

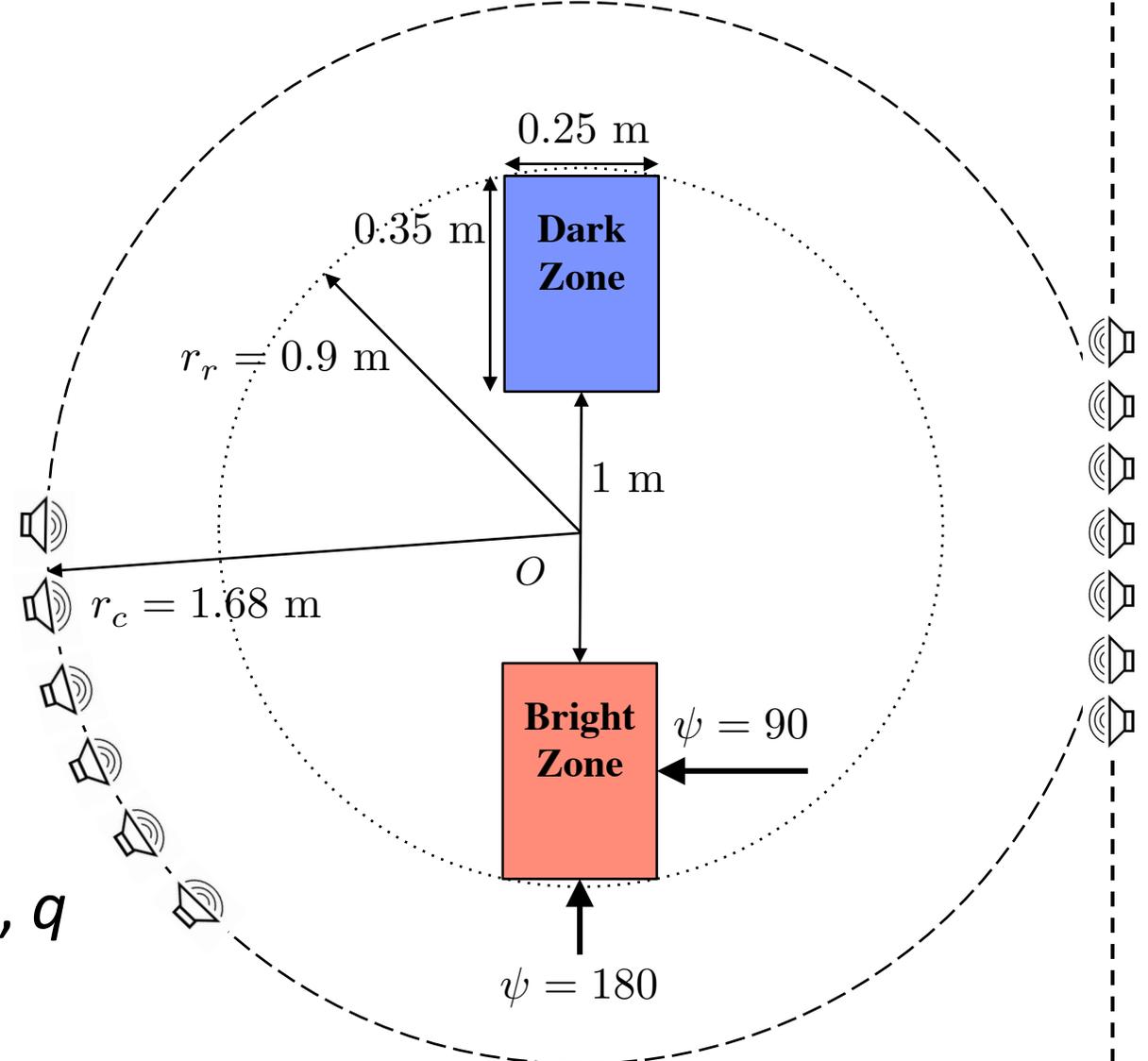
Coleman & Jackson, 2016. Planarity-based sound field optimization for multi-listener spatial audio, AES

ITU, 2015. ITU-R BS.2076-0: Audio Definition Model

Thresh & Kearney, 2017. A direct comparison of localisation performance when using first..., AES

Reproduction system

- Number of channels, L
- Loudspeaker arrangement:
 - Uniform line array
 - Uniform circular array
 - Arbitrary positions
- Assuming:
 - Calibrated gain
 - Flat full-range frequency response
 - Perfect synchronisation
- Ideal (plane or) monopole source, q



Fazi, 2010. Sound field reproduction. PhD thesis

Møller & Olsen, 2011. Sound zones, MSc thesis

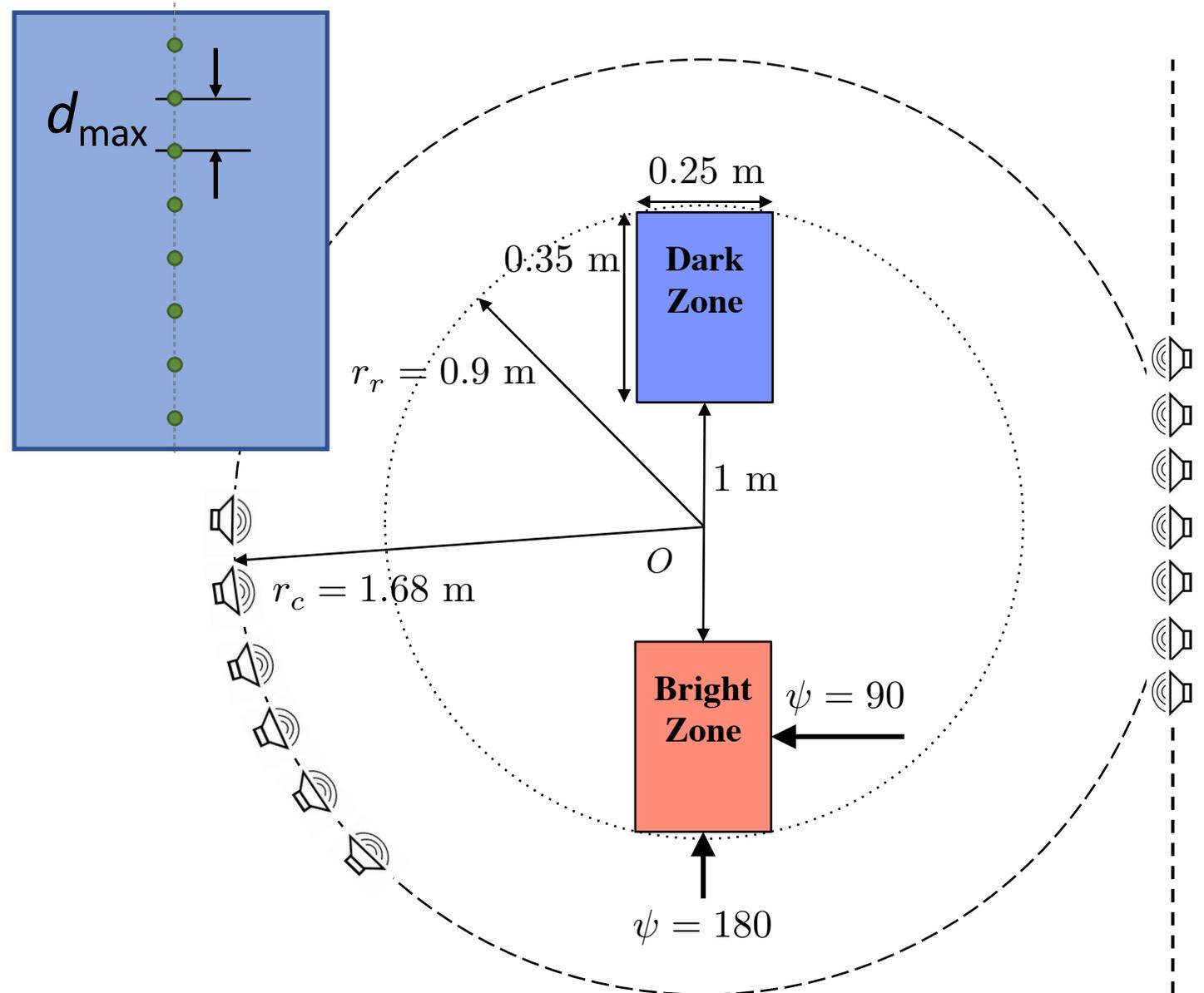
Coleman, 2014. Loudspeaker array processing for personal sound zone reproduction. PhD thesis

Zone sampling

Nyquist principle imposes maximum spacing at half the wavelength of the highest frequency:

$$d_{\max} = \frac{c}{2f}$$

e.g.,
 $d_{\max} = 5 \text{ cm}$ at 3.4kHz





Essential notation

Sources

$$\mathbf{q}(f)=[q_1,\dots,q_l,\dots,q_L]^T$$

Control mics

$$\mathbf{p}_A(f)=[p_{A,1},\dots,p_{A,n},\dots,p_{A,N}]^T$$

Transfer functions

$$\mathbf{G}_A(f) = \begin{bmatrix} G_{A,1,1} & \dots & G_{A,1,L} \\ \vdots & \ddots & \vdots \\ G_{A,N,1} & \dots & G_{A,N,L} \end{bmatrix}; \mathbf{G}_B(f) = \begin{bmatrix} G_{B,1,1} & \dots & G_{B,1,L} \\ \vdots & \ddots & \vdots \\ G_{B,N,1} & \dots & G_{B,N,L} \end{bmatrix}$$

Relations

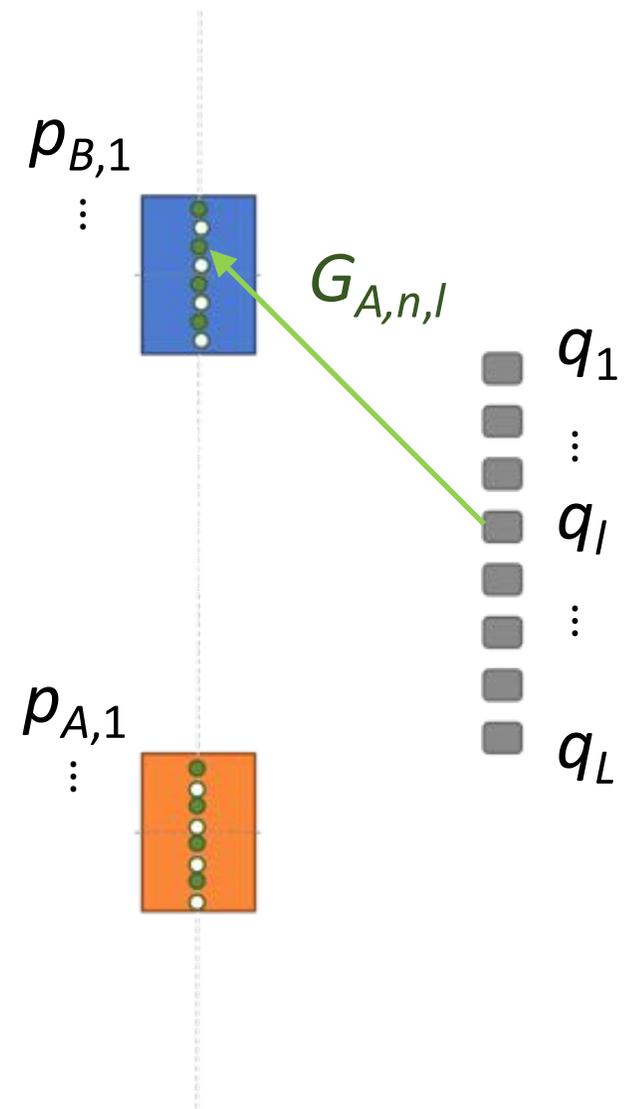
$$\mathbf{p}_A = \mathbf{G}_A \mathbf{q}$$

and

$$\mathbf{p}_B = \mathbf{G}_B \mathbf{q}$$

or

$$\begin{bmatrix} \mathbf{p}_A \\ \mathbf{p}_B \end{bmatrix} = \begin{bmatrix} \mathbf{G}_A \\ \mathbf{G}_B \end{bmatrix} [\mathbf{q}]$$



Definitions summary

- Number, size and spacing of zones
- Spatial audio objectives
- Reproduction setup
- Control and monitor microphones
- For monopole sources:

$$G_{n,l} = \frac{j\rho f}{2R} e^{j2\pi f R/c},$$

where $R = |\mathbf{r}_{n,l}|$





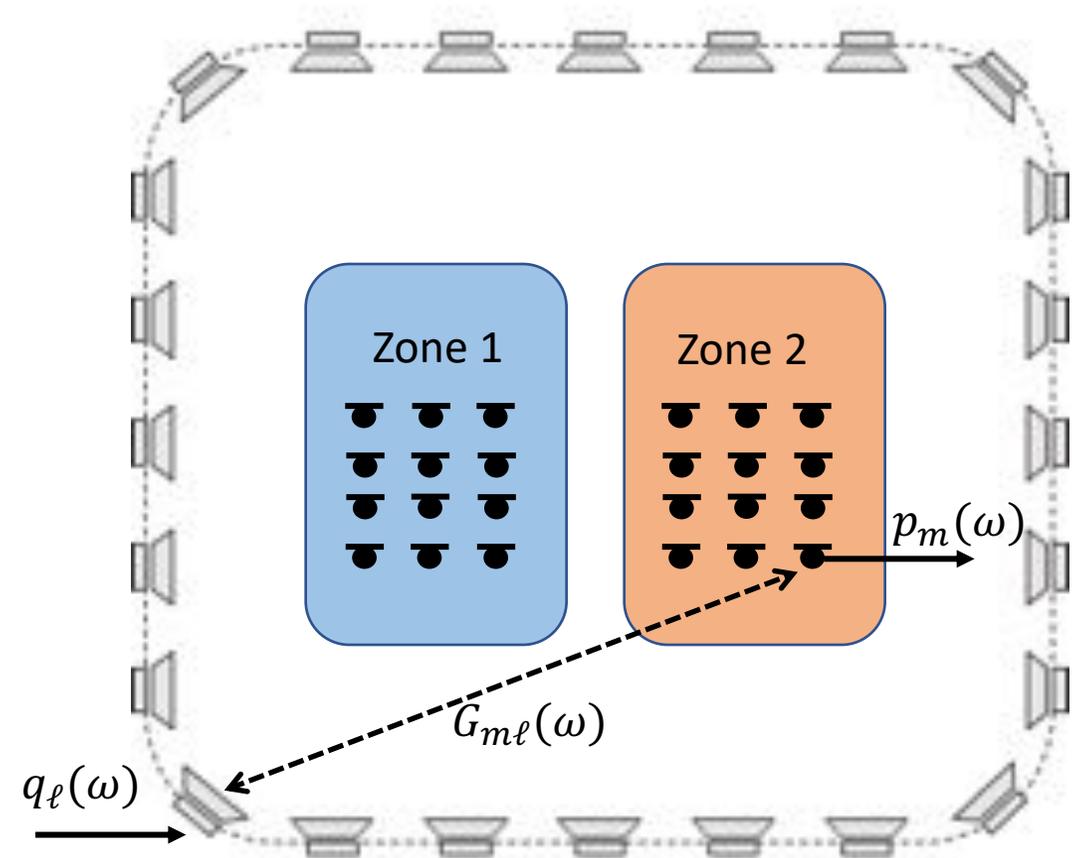
Key approaches to filter design for personal sound zones



Notation

- $q_\ell(\omega)$ driving signal of the ℓ -th loudspeaker
- $p_m(\omega)$ signal of the m -th microphone/control point
- $G_{m\ell}(\omega)$ electroacoustical transfer function between the ℓ -th speaker and the m -th control point

$$p_m(\omega) = \sum_{\ell=1}^L G_{m\ell}(\omega) q_\ell(\omega)$$



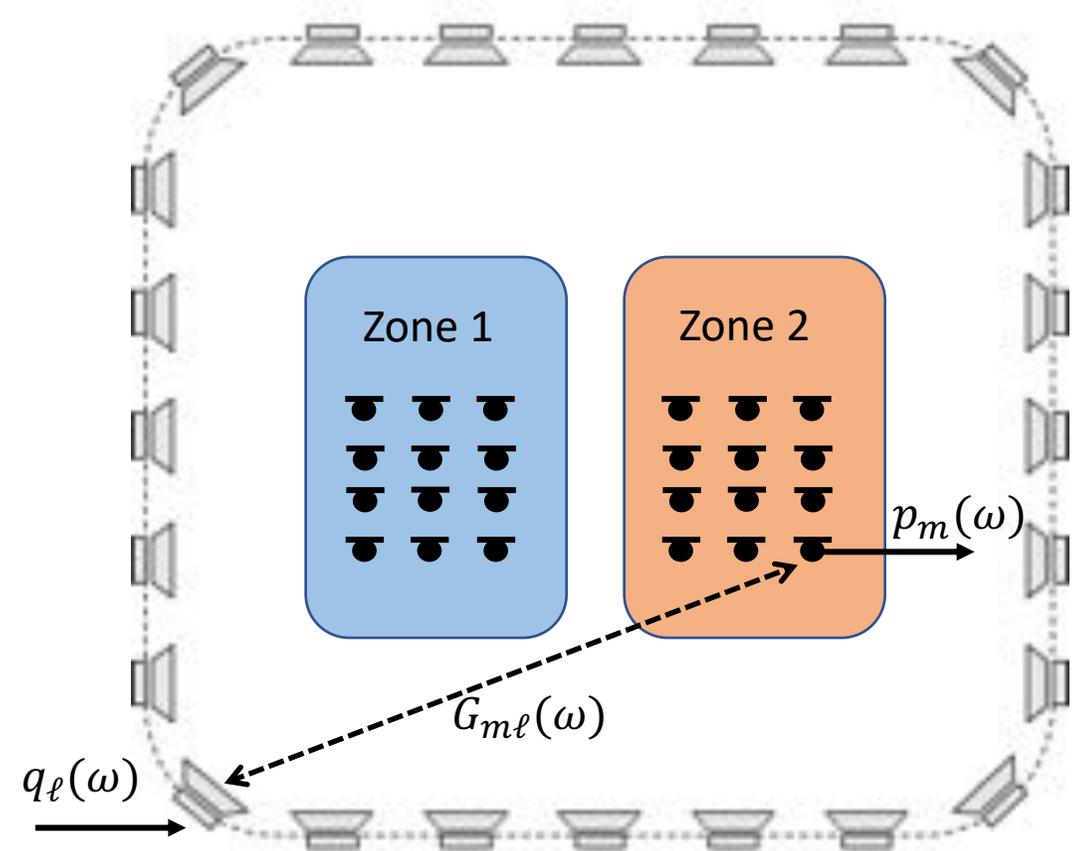
Notation

$$\mathbf{q} = [q_1(\omega), q_2(\omega), \dots, q_L(\omega)]^T$$

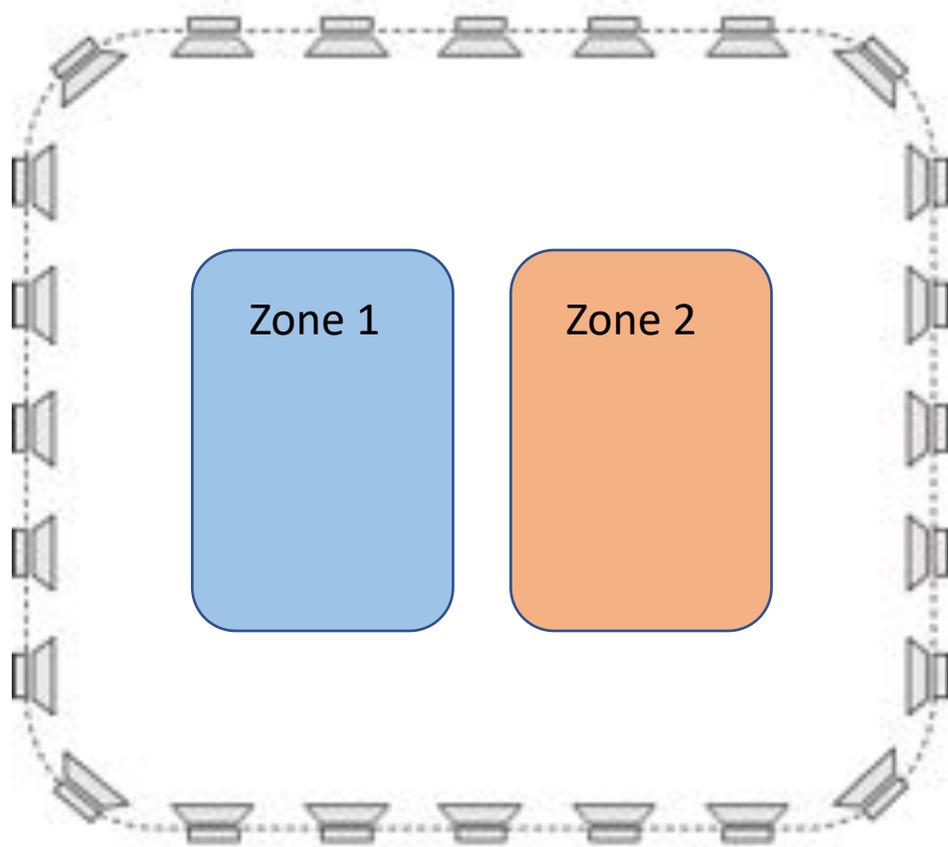
$$\mathbf{p} = [p_1(\omega), p_2(\omega), \dots, p_N(\omega)]^T$$

$$\mathbf{G} = \begin{bmatrix} G_{1,1}(\omega) & \dots & G_{1,L}(\omega) \\ \vdots & \ddots & \vdots \\ G_{N,1}(\omega) & \dots & G_{N,L}(\omega) \end{bmatrix}$$

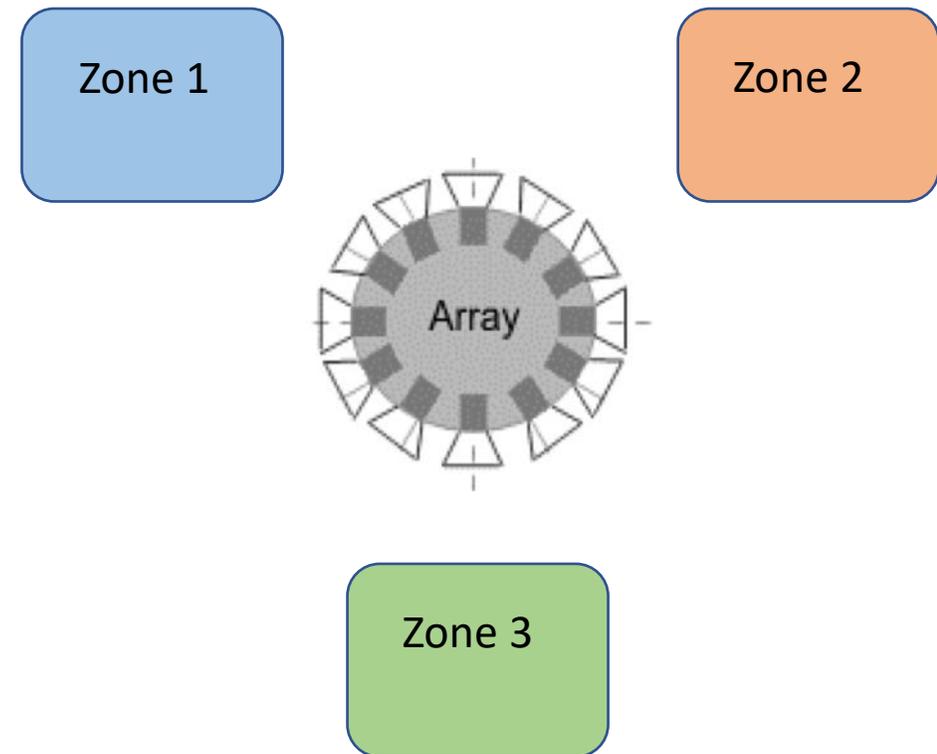
$$\mathbf{p} = \mathbf{G} \mathbf{q}$$



Interior problem



Exterior problem



Pressure matching

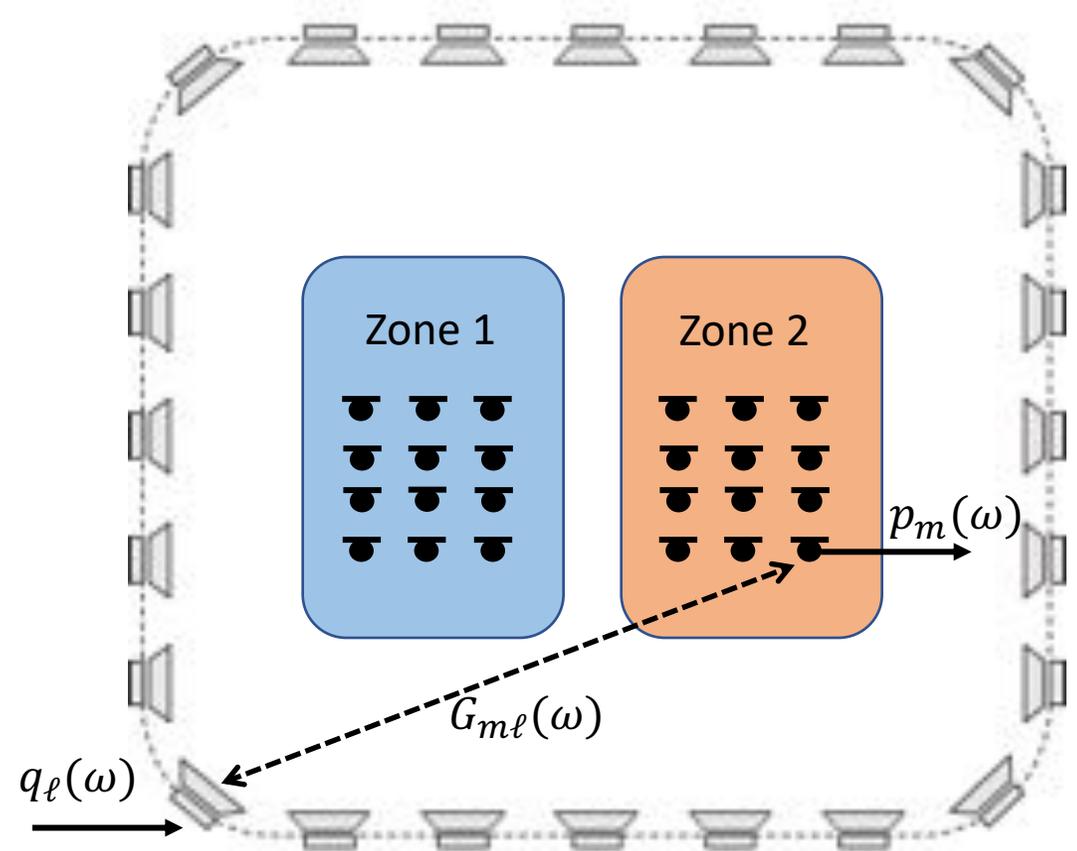
\mathbf{p}_T is the target pressure vector in the bright zone

$$\mathbf{p}_T = \left[\underbrace{p_1, p_2, \dots, p_{N_B}}_{\text{Zone 1}}, \underbrace{0, 0, \dots, 0}_{\text{Zone 2}} \right]^T$$

$N > L \rightarrow$ overdetermined problem

$$J = \|\mathbf{p} - \mathbf{p}_T\|^2 = \|\mathbf{G} \mathbf{q} - \mathbf{p}_T\|^2$$

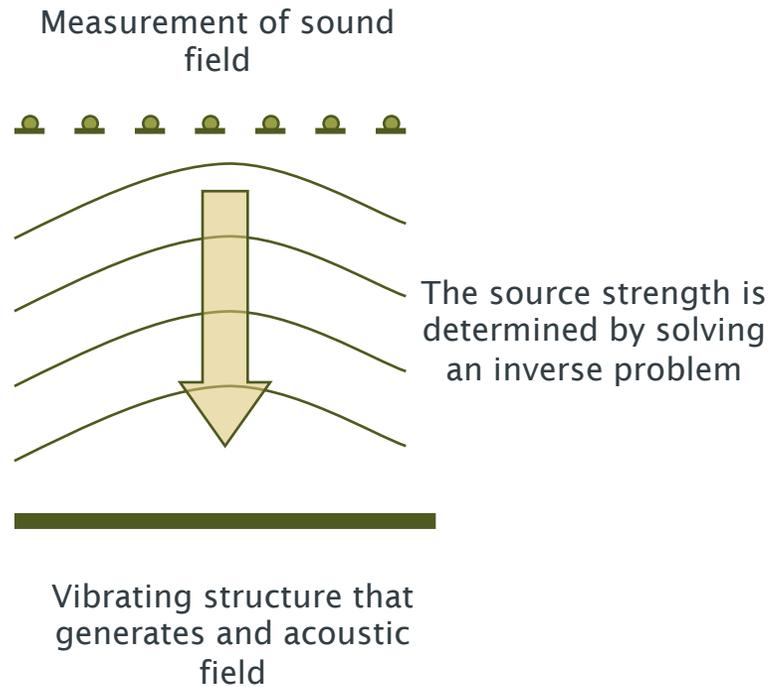
$$\mathbf{q}_{opt} = \mathbf{G}^\dagger \mathbf{p}_T = (\mathbf{G}^H \mathbf{G})^{-1} \mathbf{G}^H \mathbf{p}_T$$



$$\mathbf{p} = \mathbf{G} \mathbf{q}$$

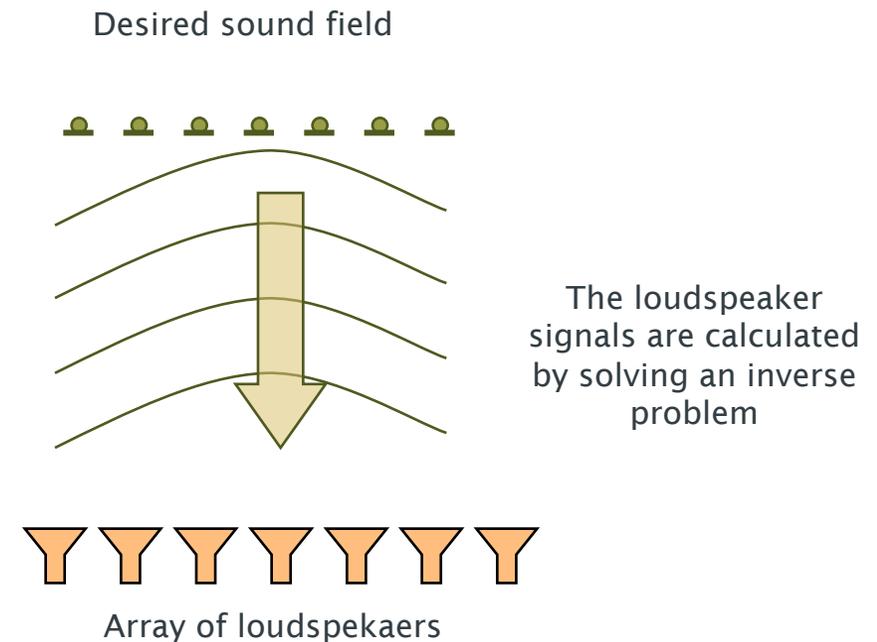
Relation to acoustical holography

ACOUSTICAL HOLOGRAPHY



- The (physical) solution exists

SOUND FIELD CONTROL



- The solution might not exist!

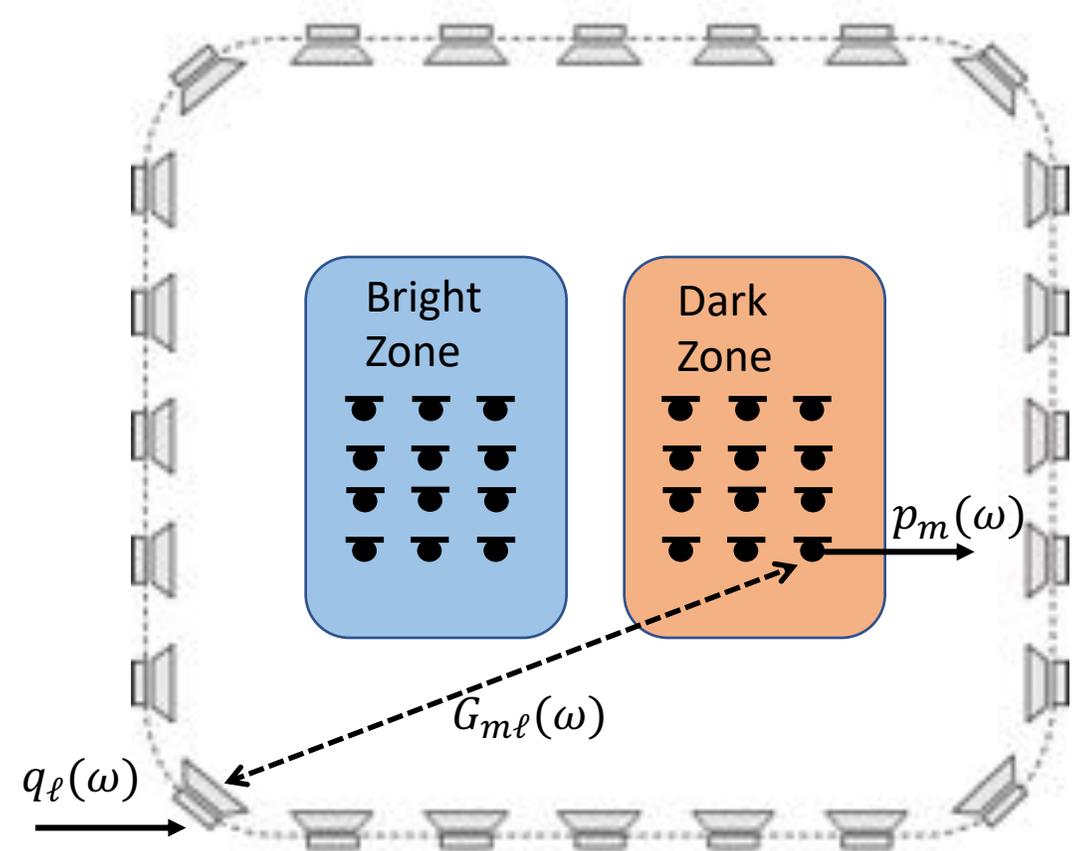
Ill-conditioning

$$\mathbf{q}_{opt} = \mathbf{G}^\dagger \mathbf{p}_T \quad J = \|\mathbf{G} \mathbf{q} - \mathbf{p}_T\|^2$$

Matrix \mathbf{G} can be ill-conditioned, especially at low frequencies (columns of \mathbf{G} are almost identical).

Array effort $E(\omega) \propto \|\mathbf{q}\|^2 = \sum_{\ell=1}^L |q_\ell(\omega)|^2$

Effort may be very large if \mathbf{G} is ill-conditioned, leading to unstable solutions.



$$\mathbf{p} = \mathbf{G} \mathbf{q}$$

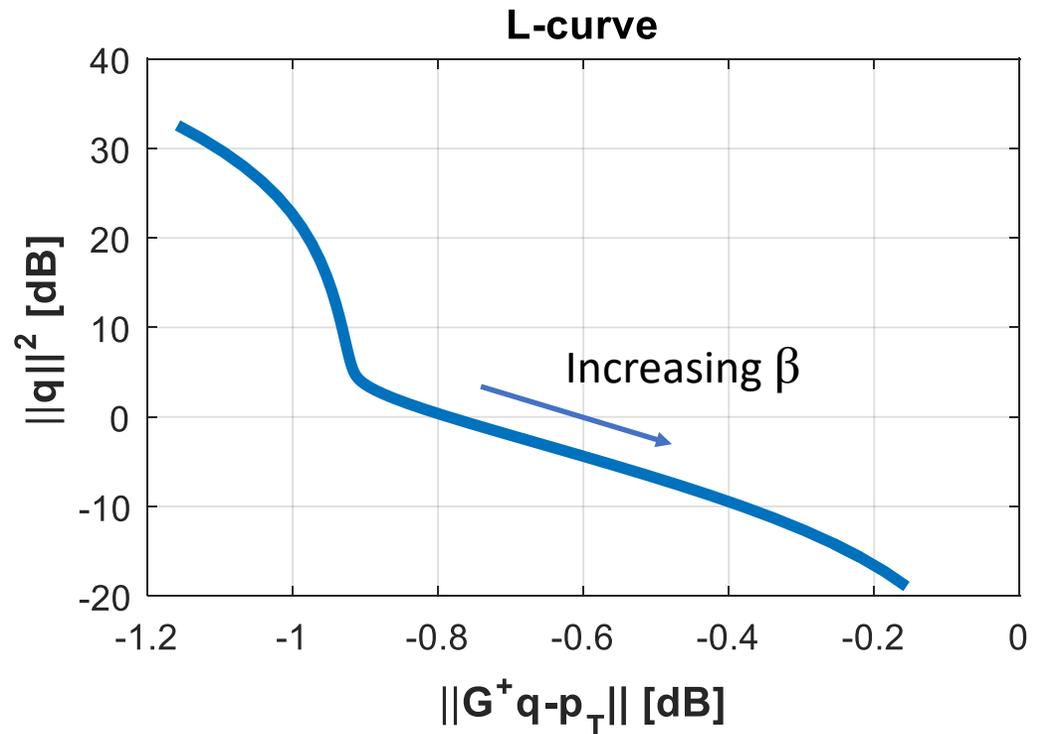
Tikhonov Regularization

Cost function with Tikhonov regularization

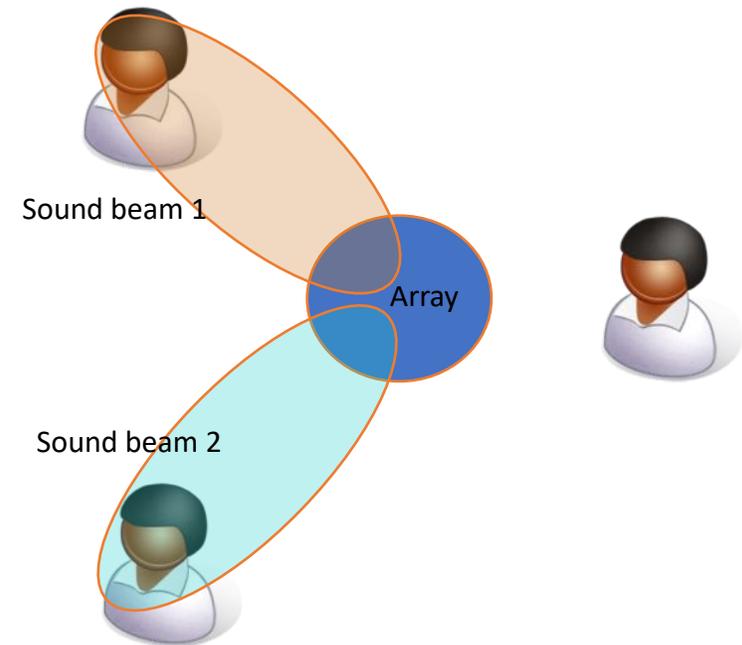
$$J = \|\mathbf{G} \mathbf{q} - \mathbf{p}_T\|^2 + \beta \|\mathbf{q}\|^2$$

$$\mathbf{q}_{opt} = (\mathbf{G}^H \mathbf{G} + \beta \mathbf{I})^{-1} \mathbf{G}^H \mathbf{p}_T$$

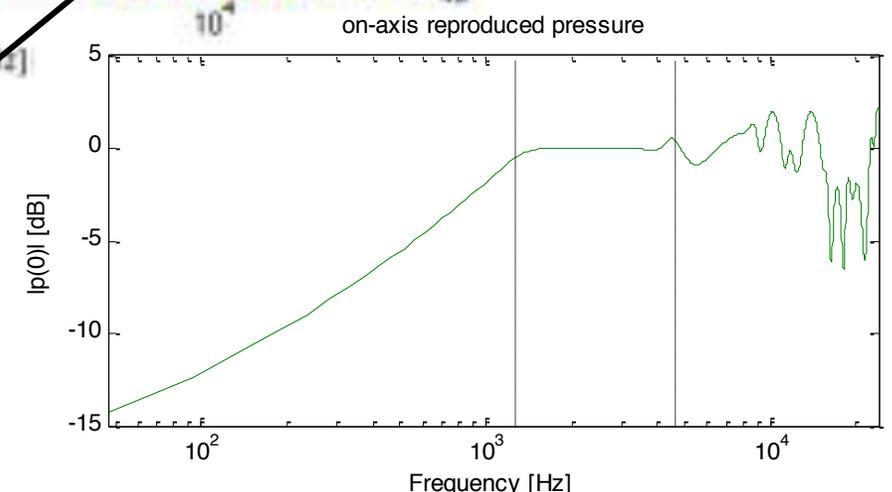
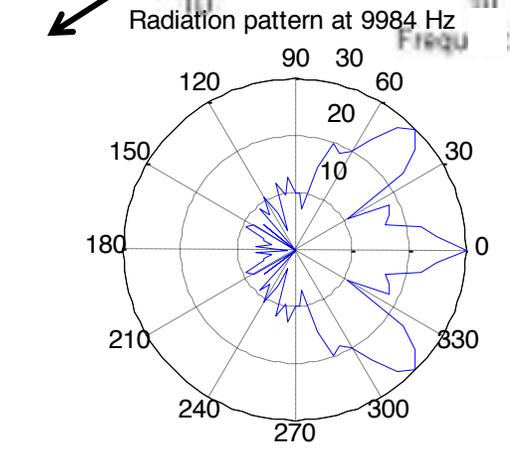
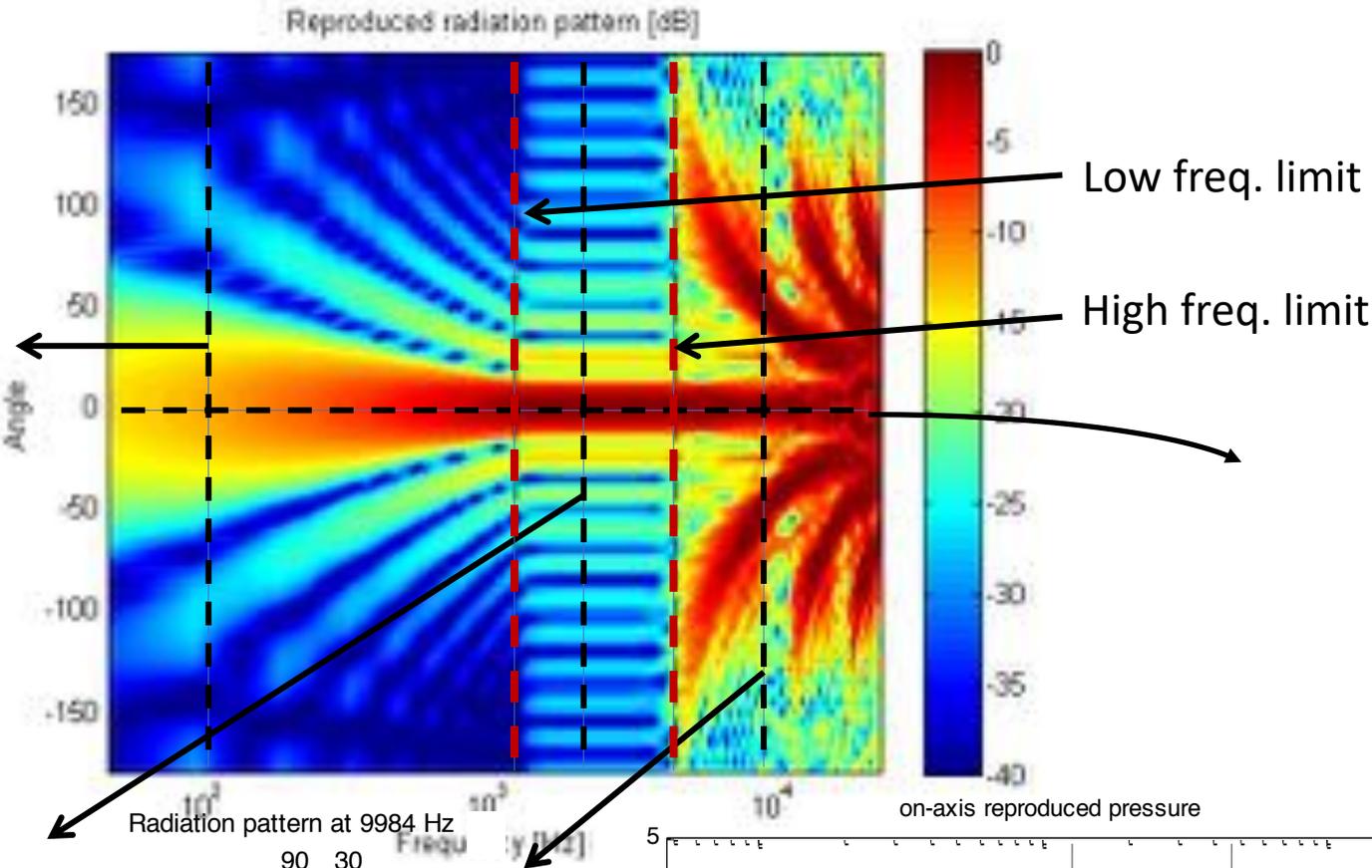
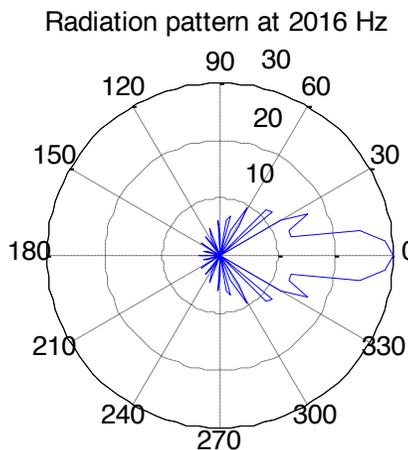
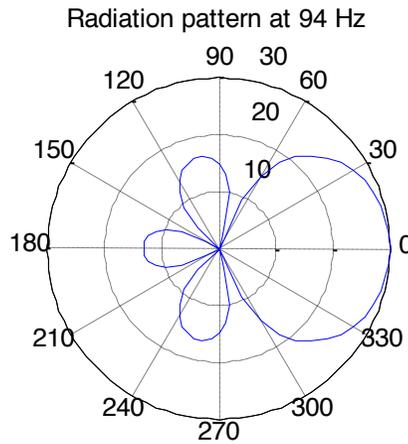
Reduces of array effort, but increases error



Example: multi-zone with cylindrical array



Reproduced radiation pattern



Bright and dark zones

$$\mathbf{q} = [q_1(\omega), q_2(\omega), \dots, q_L(\omega)]^T$$

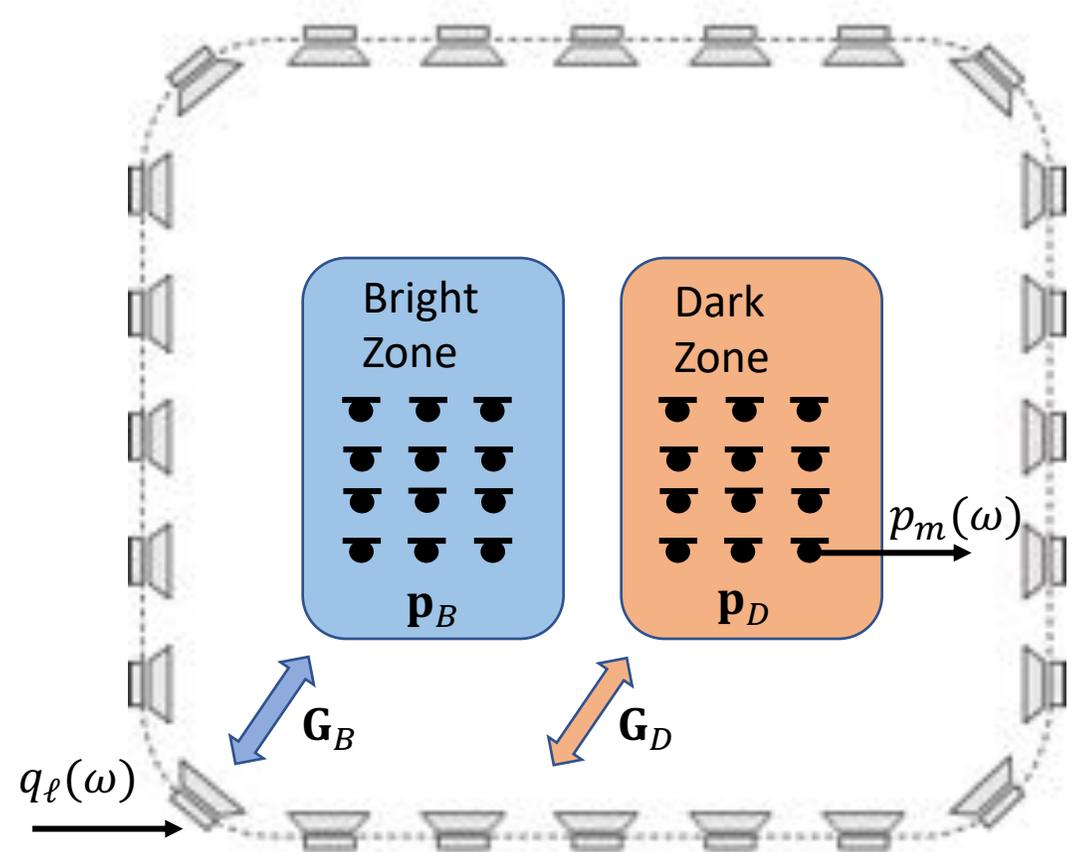
$$\mathbf{p}_B = [p_1^{(B)}(\omega), p_2^{(B)}(\omega), \dots, p_{N_B}^{(B)}(\omega)]^T$$

$$\mathbf{p}_D = [p_1^{(D)}(\omega), p_2^{(D)}(\omega), \dots, p_{N_D}^{(D)}(\omega)]^T$$

$$\mathbf{p}_B = \mathbf{G}_B \mathbf{q}$$

$$\mathbf{p}_D = \mathbf{G}_D \mathbf{q}$$

$$\mathbf{p} = \begin{vmatrix} \mathbf{p}_B \\ \dots \\ \mathbf{p}_D \end{vmatrix}, \quad \mathbf{G} = \begin{vmatrix} \mathbf{G}_B \\ \dots \\ \mathbf{G}_D \end{vmatrix}$$

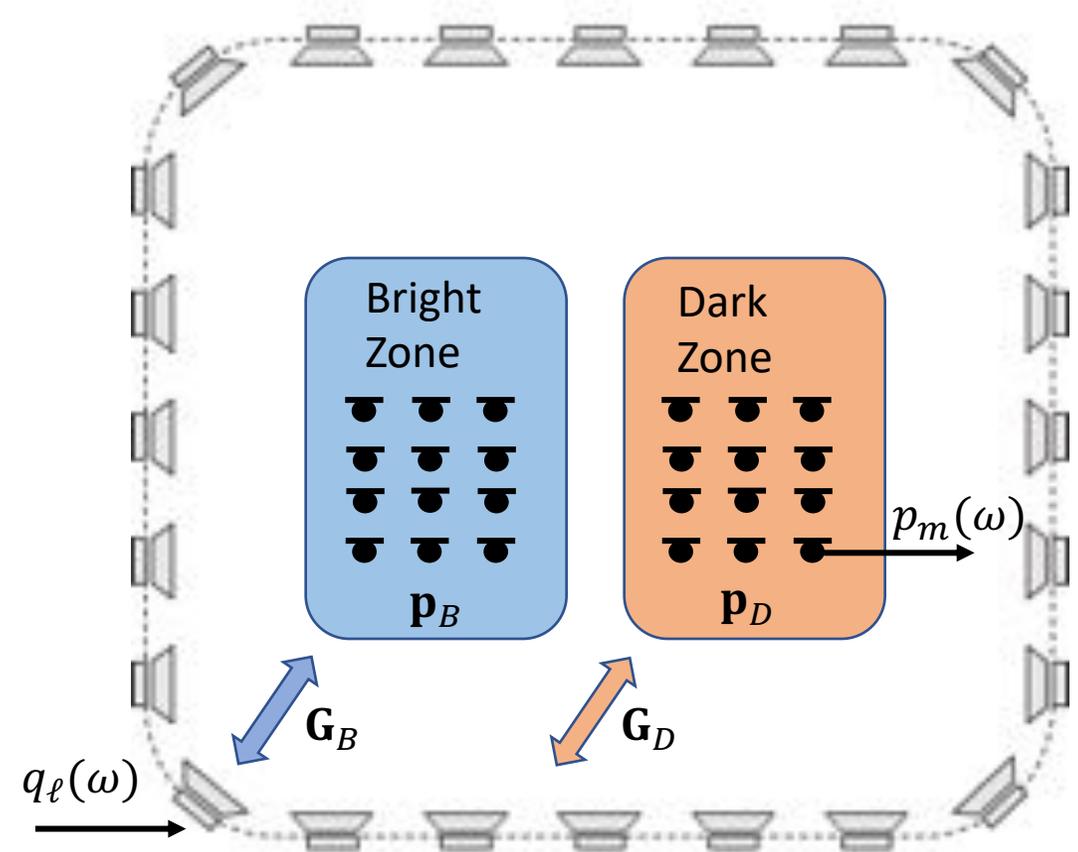


Acoustic contrast

It is the ratio of the average acoustic potential energy in two zones

$$C = \frac{\langle E_B \rangle}{\langle E_D \rangle} \approx \frac{\|\mathbf{p}_B\|^2 / N_B}{\|\mathbf{p}_D\|^2 / N_D}$$

$$= \frac{N_D \mathbf{q}^H \mathbf{G}_B^H \mathbf{G}_B \mathbf{q}}{N_B \mathbf{q}^H \mathbf{G}_D^H \mathbf{G}_D \mathbf{q}}$$



$$\mathbf{p}_{B/D} = \mathbf{G}_{B/D} \mathbf{q}$$

Acoustic contrast maximisation

- Direct formulation: maximises the energy in the bright zone while keeping the energy in the dark zone to a constant value D and the effort below a given value E

$$\text{Maximise } \|\mathbf{p}_B\|^2 \text{ s.t. } \|\mathbf{p}_D\|^2 = D \text{ and } \|\mathbf{q}\|^2 \leq E$$

- Indirect formulation: minimise the energy in the dark zone while keeping the energy in the bright zone to a constant value B and the effort below a given value E

$$\text{Minimise } \|\mathbf{p}_D\|^2 \text{ s.t. } \|\mathbf{p}_B\|^2 = B \text{ and } \|\mathbf{q}\|^2 \leq E$$

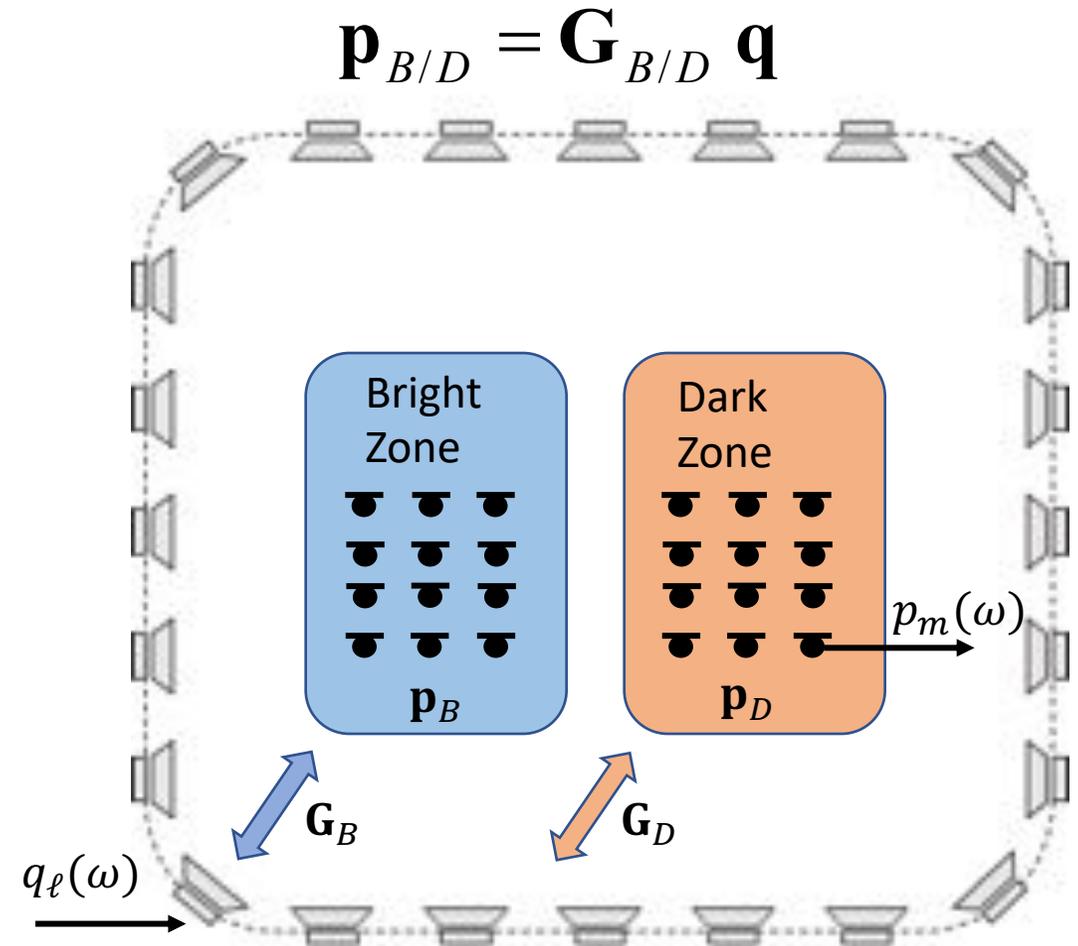
Acoustic contrast maximisation (direct)

- Maximise $\|\mathbf{p}_B\|^2$ s.t. $\|\mathbf{p}_D\|^2 = D$ and $\|\mathbf{q}\|^2 \leq E$

$$L = \mathbf{q}^H \mathbf{G}_B^H \mathbf{G}_B \mathbf{q} - \lambda_1 (\mathbf{q}^H \mathbf{G}_D^H \mathbf{G}_D \mathbf{q} - D) - \lambda_2 (\mathbf{q}^H \mathbf{q} - E)$$

$$\lambda_1 \mathbf{q}_{opt} = \left(\mathbf{G}_D^H \mathbf{G}_D \right)^{-1} \left(\mathbf{G}_B^H \mathbf{G}_B - \lambda_2 \mathbf{I} \right) \mathbf{q}_{opt}$$

- The optimal solution \mathbf{q}_{opt} is the eigenvector associated with the largest eigenvalue λ_1



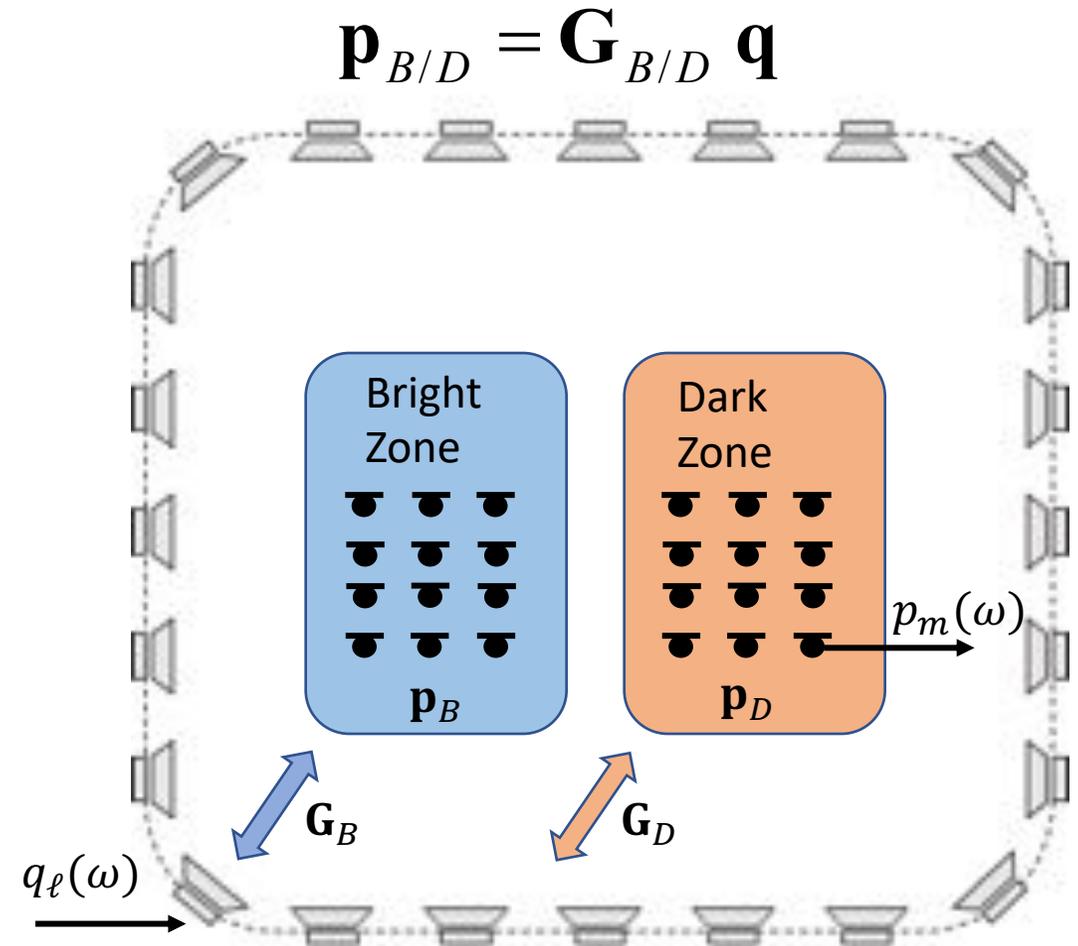
Acoustic contrast maximisation (indirect)

- Minimise $\|\mathbf{p}_D\|^2$ s.t. $\|\mathbf{p}_B\|^2 = B$ and $\|\mathbf{q}\|^2 \leq E$

$$L = \mathbf{q}^H \mathbf{G}_D^H \mathbf{G}_D \mathbf{q} + \lambda_1 (\mathbf{q}^H \mathbf{G}_B^H \mathbf{G}_B \mathbf{q} - B) + \lambda_2 (\mathbf{q}^H \mathbf{q} - E)$$

$$\lambda \mathbf{q}_{opt} = (\mathbf{G}_D^H \mathbf{G}_D + \lambda_2 \mathbf{I})^{-1} (\mathbf{G}_B^H \mathbf{G}_B) \mathbf{q}_{opt}$$

- The optimal solution \mathbf{q}_{opt} is the eigenvector associated with the largest eigenvalue λ



Energy difference maximisation

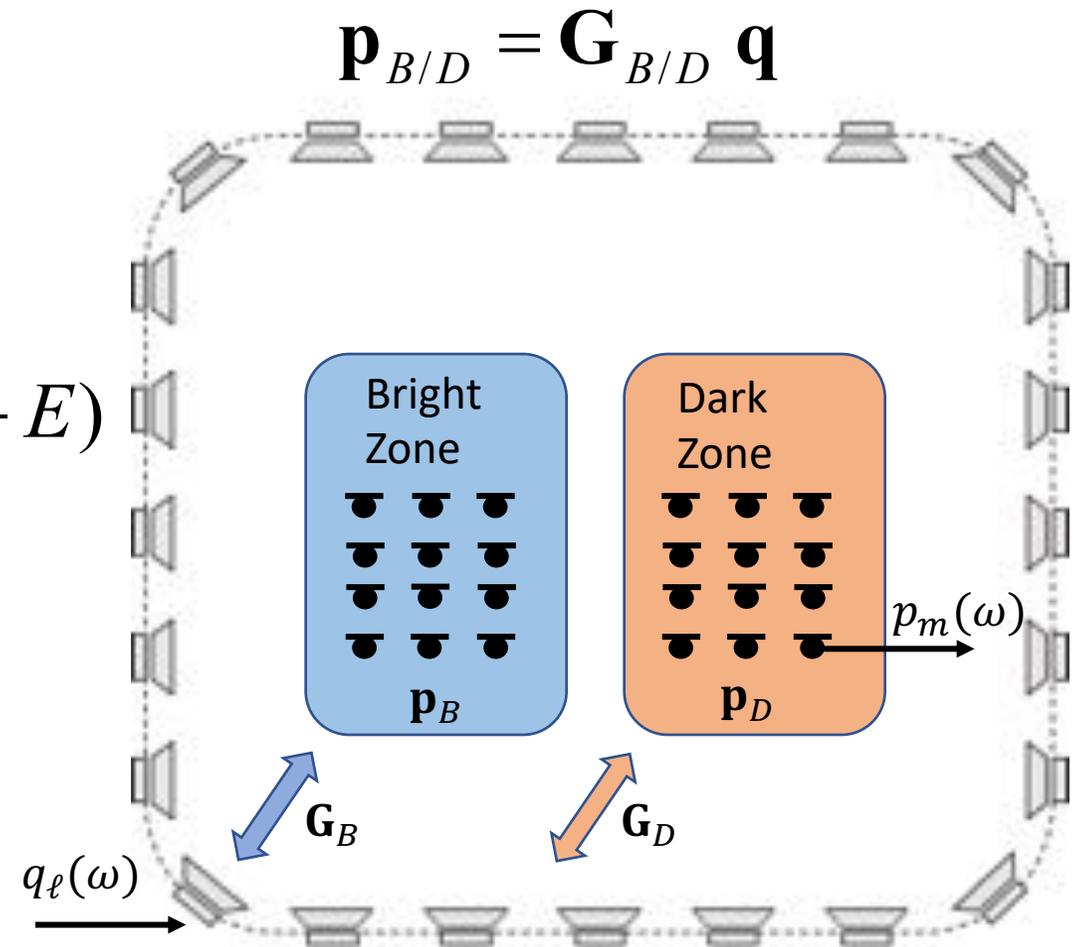
- Maximise energy difference

$$\|\mathbf{p}_B\|^2 - \alpha \|\mathbf{p}_D\|^2 \quad \text{s.t.} \quad \|\mathbf{q}\|^2 \leq E$$

$$L = \mathbf{q}^H \mathbf{G}_B^H \mathbf{G}_B \mathbf{q} - \alpha \mathbf{q}^H \mathbf{G}_D^H \mathbf{G}_D \mathbf{q} - \lambda_2 (\mathbf{q}^H \mathbf{q} - E)$$

$$\lambda_2 \mathbf{q}_{opt} = (\mathbf{G}_B^H \mathbf{G}_B - \alpha \mathbf{G}_D^H \mathbf{G}_D) \mathbf{q}_{opt}$$

- The optimal solution \mathbf{q}_{opt} is the eigenvector associated with the largest eigenvalue λ_2





Performance measures



Measures of personal sound quality

- Interference
 - From sounds other than the intended personal sound
- Robustness
 - For any practical implementation
- Timbral quality
 - Quality of the received personal sound
- Spatial quality
 - Reproduction artefacts or spatial distortion

Emiya et al., 2011. Subjective and objective quality assessment of audio source separation, TASLP

Francombe et al., 2014. Elicitation of attributes for the evaluation of audio-on-audio interference, JASA

Conetta et al., 2014. Spatial Audio Quality Perception (Part 2): A linear regression model, JAES

George et al., 2010. Development and validation of an unintrusive model for predicting..., JAES

Jackson et al., 2008. QESTRAL (Part 3): System and metrics for spatial quality prediction, AES

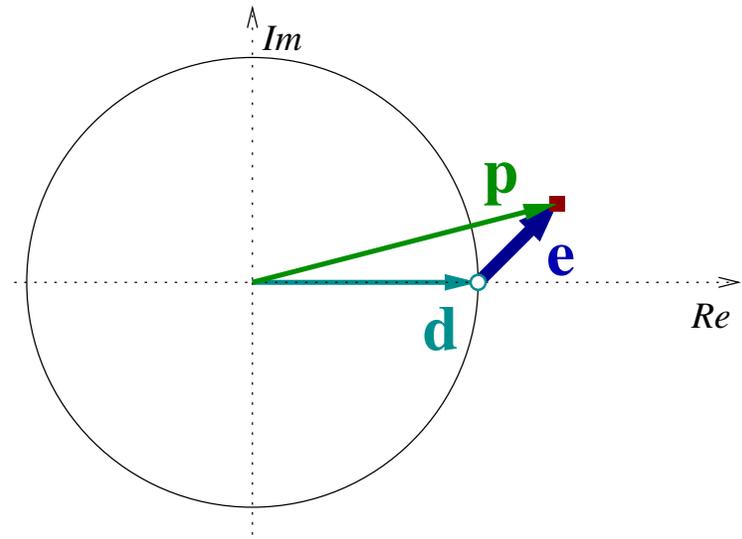
Goodness of fit: MMSE

- Target sound field
- Given a target sound field, $\mathbf{d}_A(f) = [d_{A,1}, \dots, d_{A,m}, \dots, d_{A,M}]^T$, the mean squared error is

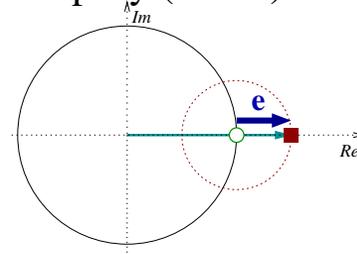
$$\text{MSE} = \frac{1}{M} \sum_{m=1}^M |p_{A,m} - d_{A,m}|^2$$

or

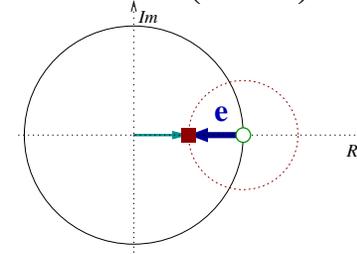
$$\text{MSE} = \frac{1}{M} (\mathbf{p}_A - \mathbf{d}_A)^H (\mathbf{p}_A - \mathbf{d}_A)$$



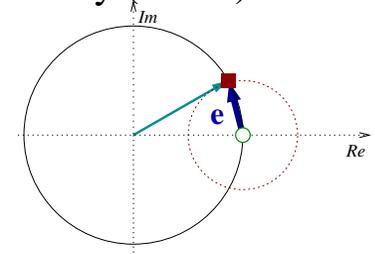
Amplify (100%)



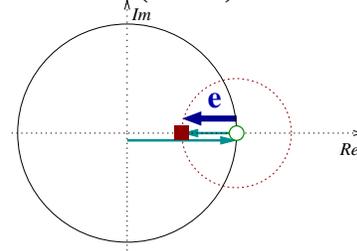
Attenuate (100%)



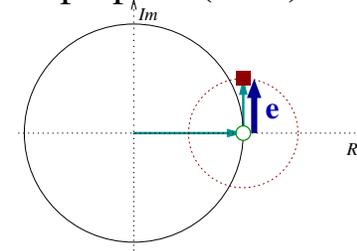
Delay (100%)



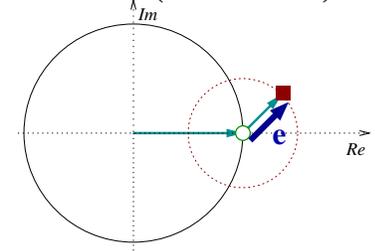
Cancel (60%)



Superpose (80%)



Disturb (60..100%)





Independent sampling points

Sources

$$\mathbf{q}(f)=[q_1,\dots,q_l,\dots,q_L]^T$$

Control mics

$$\mathbf{p}_A(f)=[p_{A,1},\dots,p_{A,n},\dots,p_{A,N}]^T$$

Monitor mics

$$\mathbf{o}_A(f)=[o_{A,1},\dots,o_{A,m},\dots,o_{A,M}]^T$$

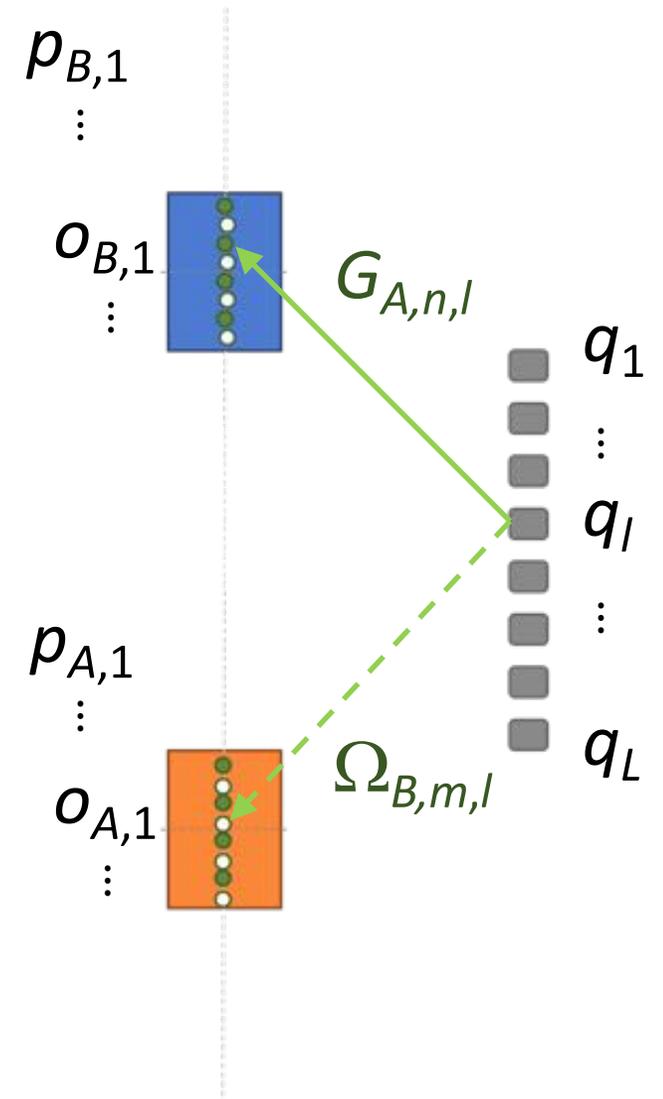
Transfer functions

$$\mathbf{G}_A(f) = \begin{bmatrix} G_{A,1,1} & \dots & G_{A,1,L} \\ \vdots & \ddots & \vdots \\ G_{A,N,1} & \dots & G_{A,N,L} \end{bmatrix}; \quad \mathbf{\Omega}_A(f) = \begin{bmatrix} \Omega_{A,1,1} & \dots & \Omega_{A,1,L} \\ \vdots & \ddots & \vdots \\ \Omega_{A,M,1} & \dots & \Omega_{A,M,L} \end{bmatrix}$$

Relations

$$\begin{bmatrix} \mathbf{p}_A \\ \mathbf{p}_B \end{bmatrix} = \begin{bmatrix} \mathbf{G}_A \\ \mathbf{G}_B \end{bmatrix} [\mathbf{q}]$$

$$\begin{bmatrix} \mathbf{o}_A \\ \mathbf{o}_B \end{bmatrix} = \begin{bmatrix} \mathbf{\Omega}_A \\ \mathbf{\Omega}_B \end{bmatrix} [\mathbf{q}]$$

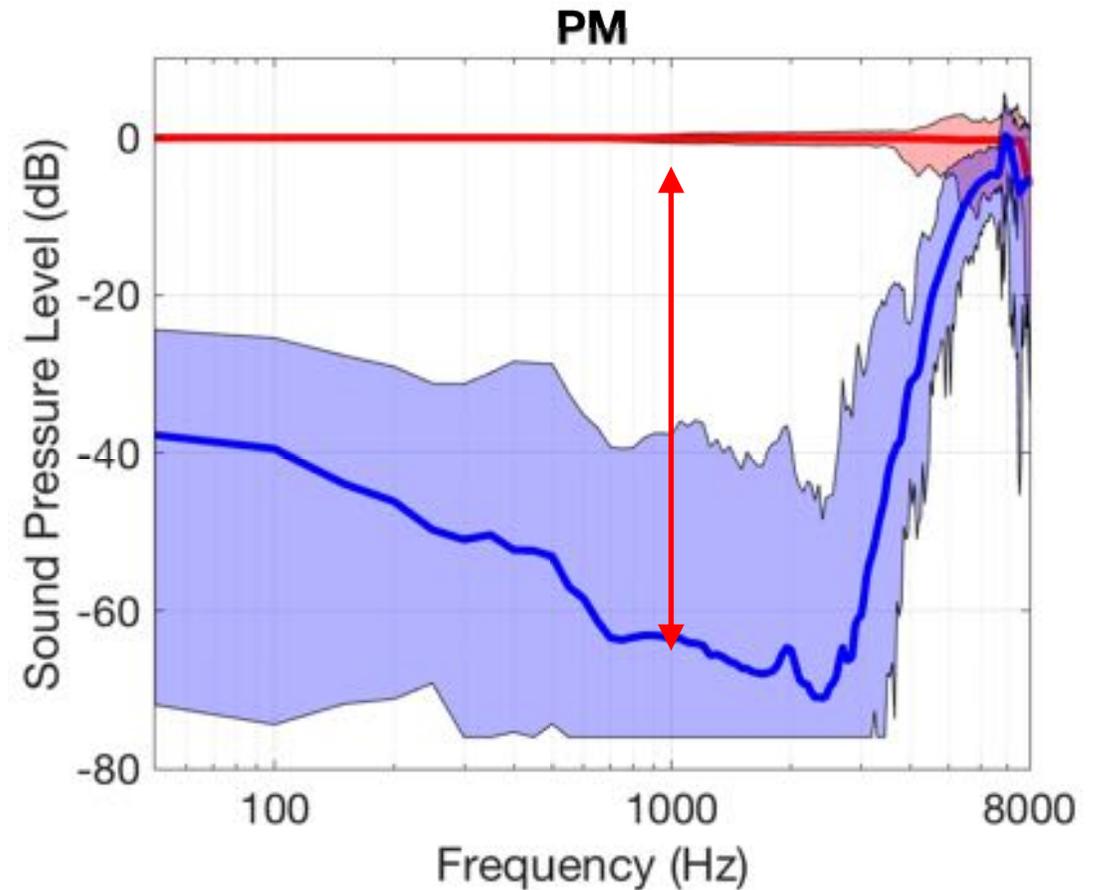


Contrast

- Contrast reflects the ratio of the sound pressure levels (SPLs) in the zones

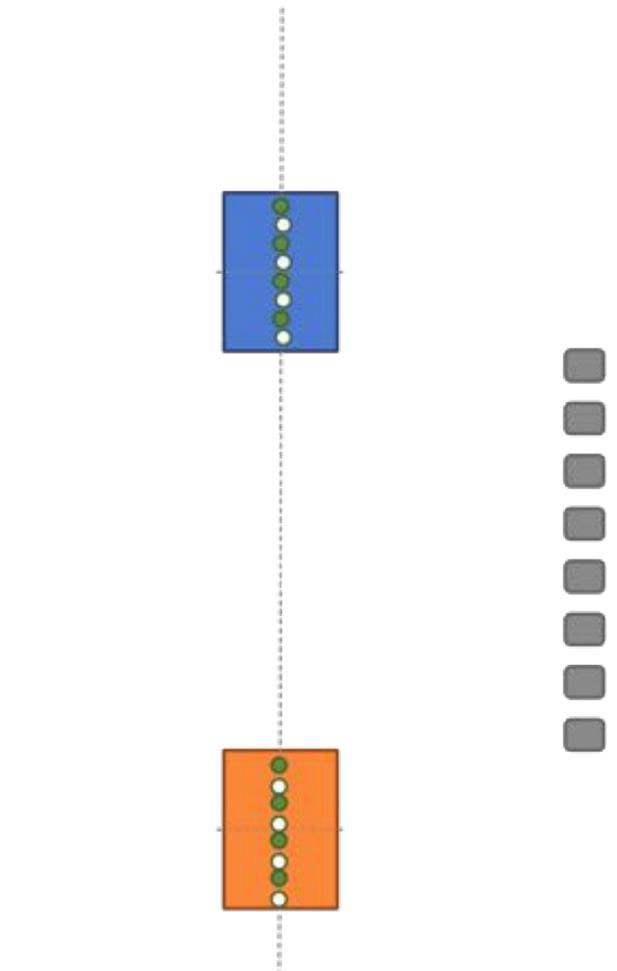
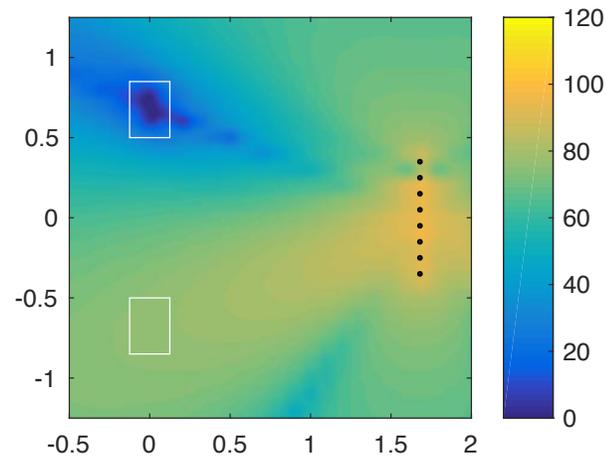
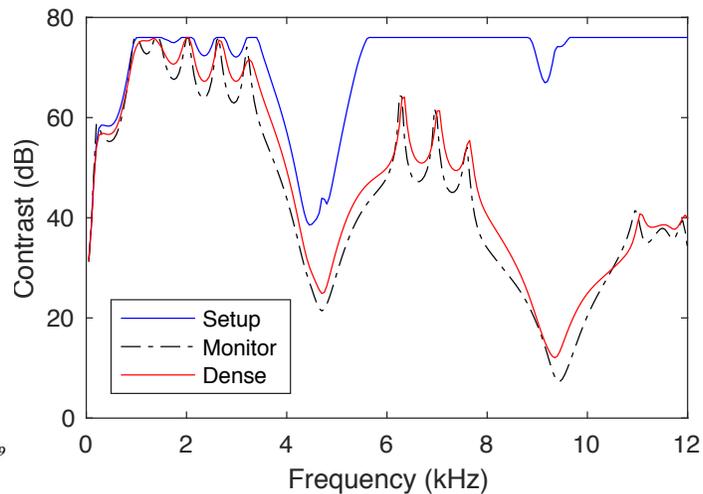
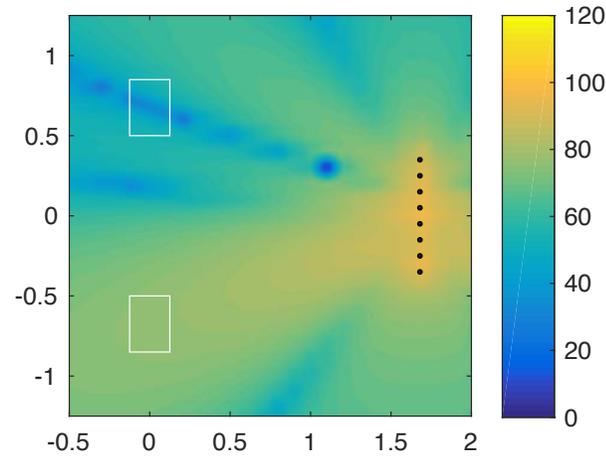
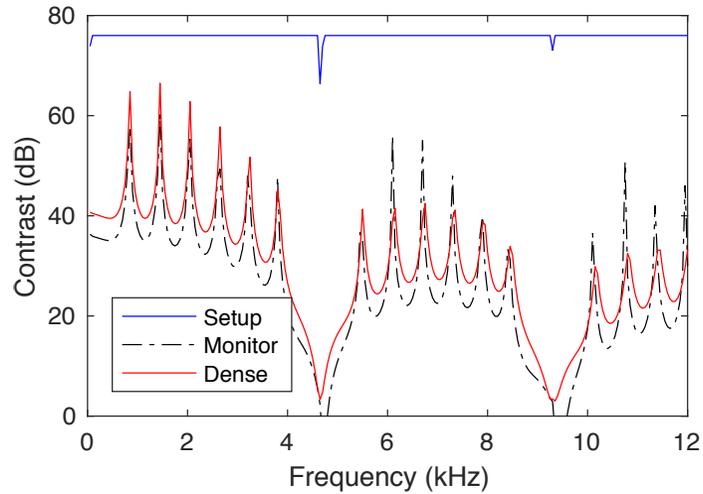
$$C = 10 \log_{10} \frac{N_B \mathbf{p}_A^H \mathbf{p}_A}{N_A \mathbf{p}_B^H \mathbf{p}_B}$$

$$C = 10 \log_{10} \frac{M_B \mathbf{o}_A^H \mathbf{o}_A}{M_A \mathbf{o}_B^H \mathbf{o}_B}$$





Calibration (control) and evaluation (monitor) microphones



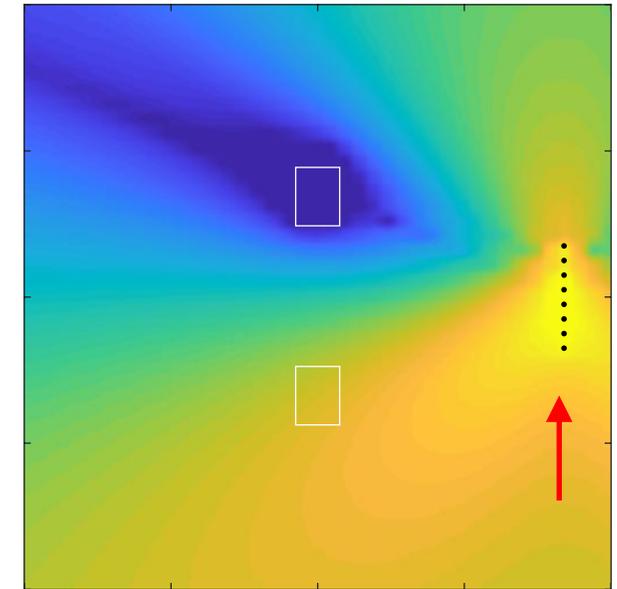
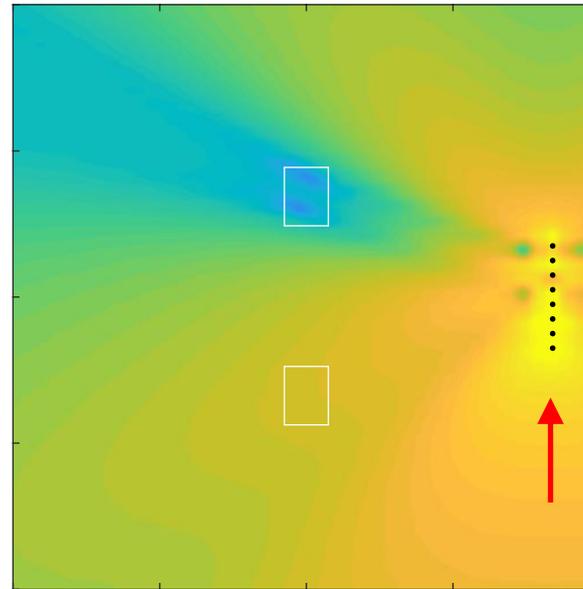
Jackson & Ross, 1996. Application of active noise control to corporate aircraft, ASME
Møller & Olsen, 2016. Sound zones: On performance prediction..., AES conf



Effort

- Efficiency
 - Total acoustical power
- Robustness
 - Errors in calibration
 - Influence of reflections
- Effort metric:

$$E = 10 \log_{10} \frac{\mathbf{q}^H \mathbf{q}}{|q_r|^2}$$

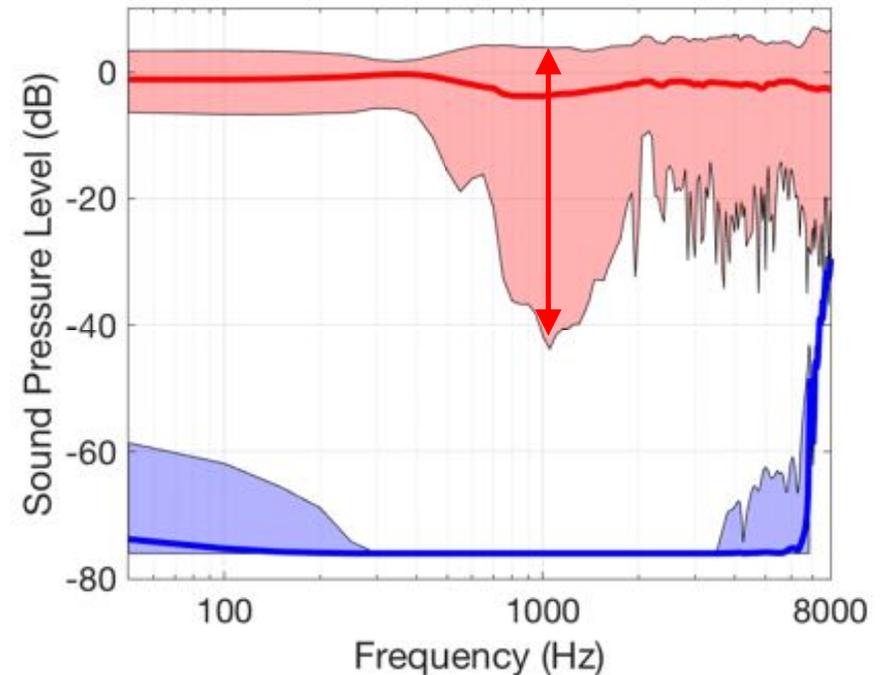
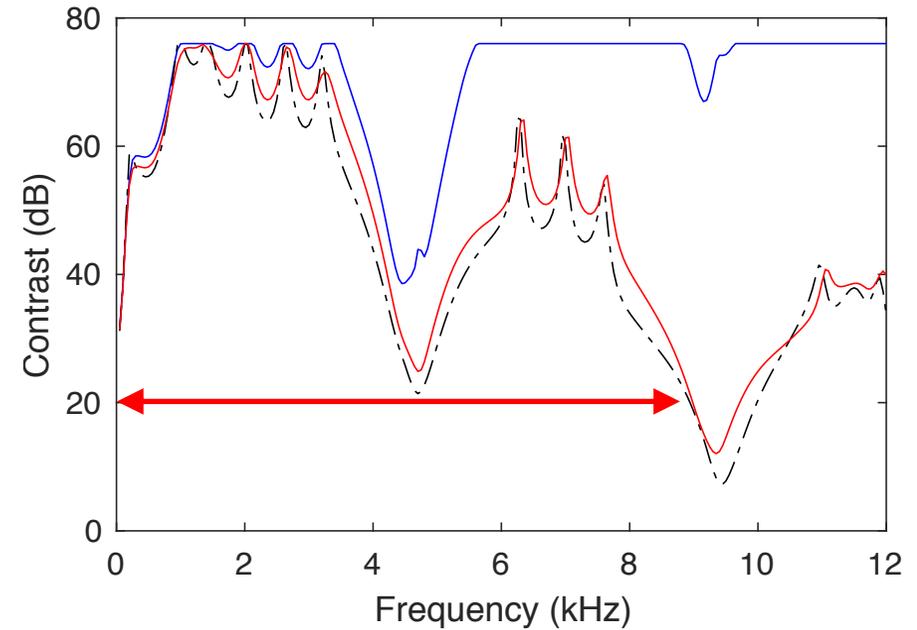


Frequency response

- Timbral quality
- Effective bandwidth
- Flatness across frequency
 - Tone, colouration
 - Comb filtering

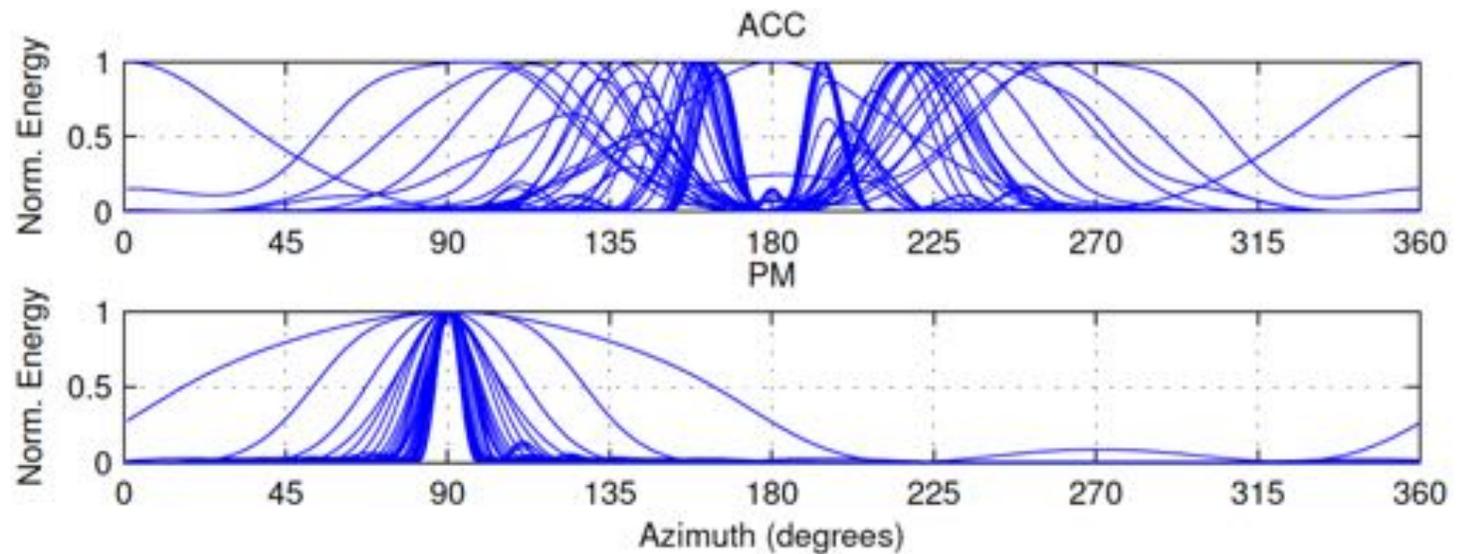
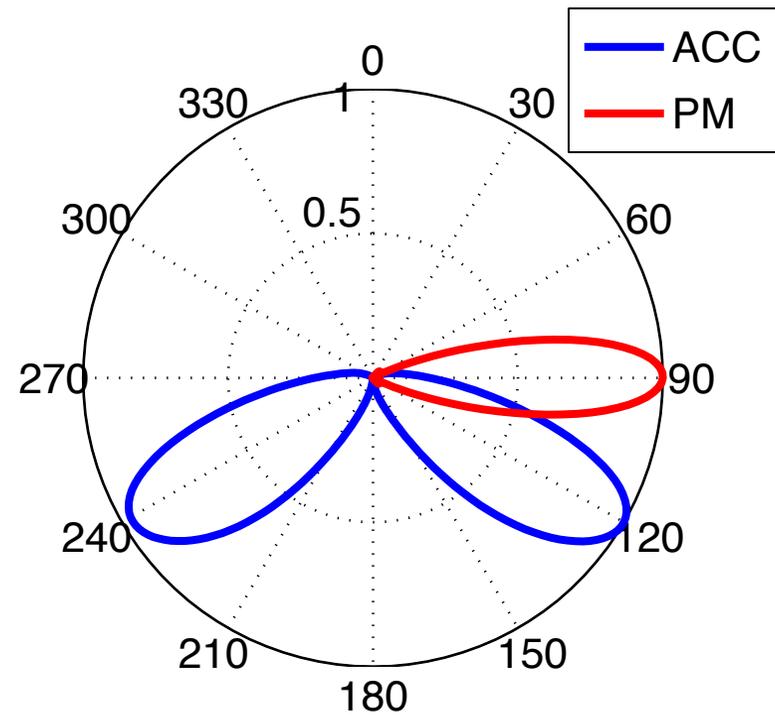
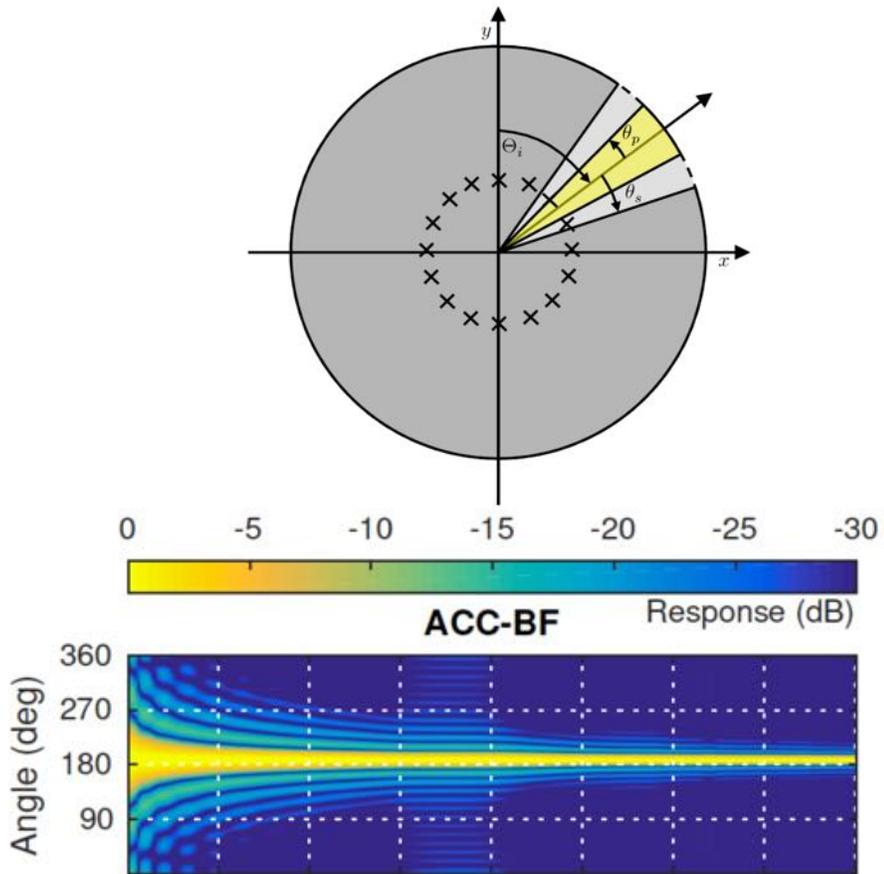
PEAQ

Baykaner et al., 2015. The relationship between target quality and interference in sound zone, JAES

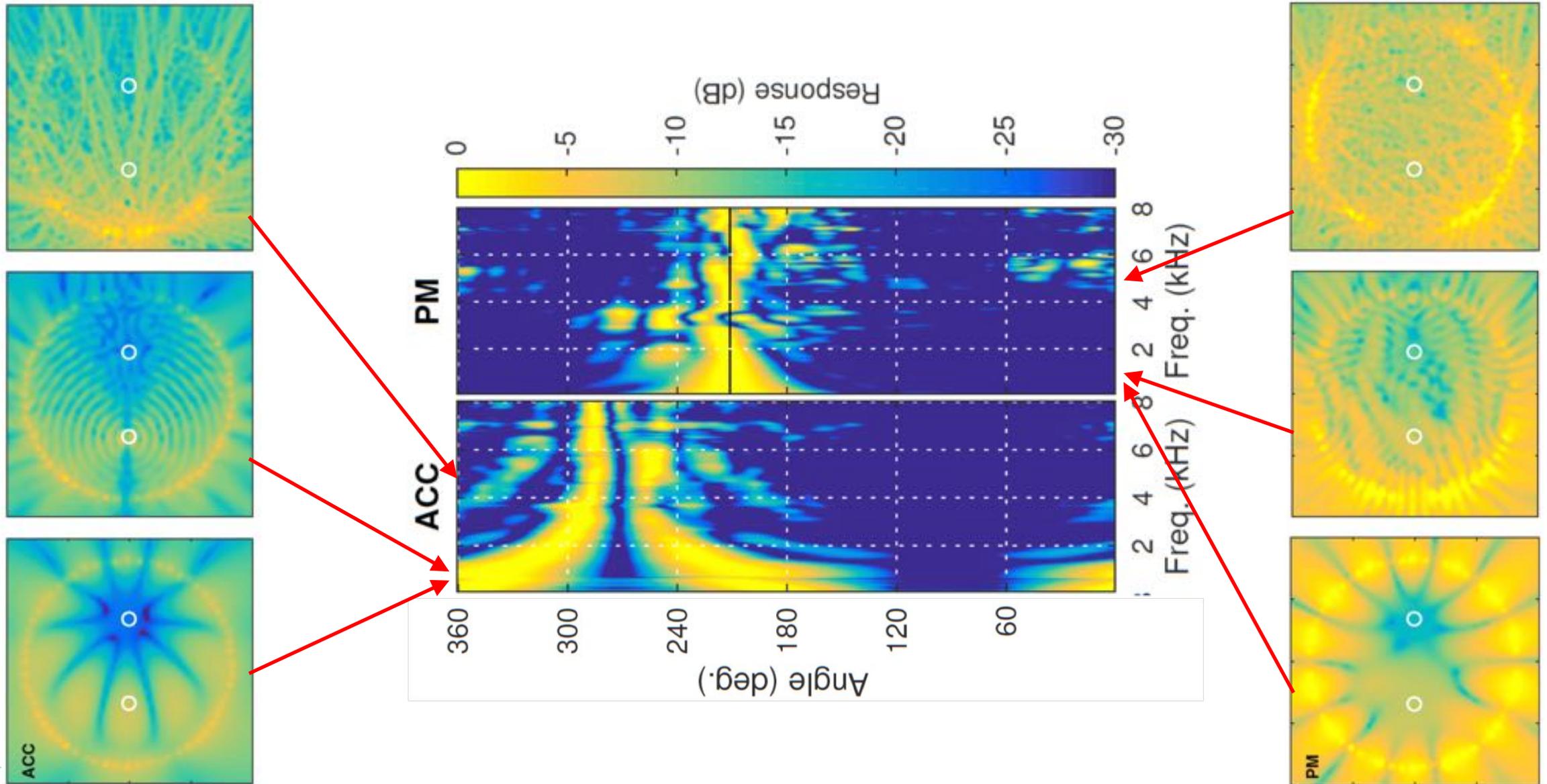


Spatial measures

- Estimated angular spectrum



Spatial examples



Coleman et al., 2014. Acoustic contrast, planarity and robustness of sound zone methods..., JASA

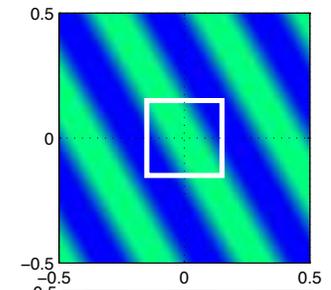
Planarity

- Summary measure of sound field homogeneity, i.e., vs. plane wave

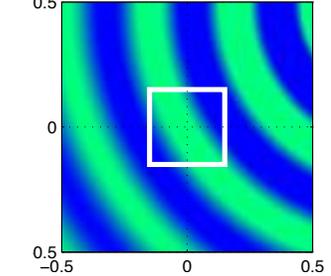
$$\eta = \frac{\sum_i w_i \mathbf{u}_i \cdot \mathbf{u}_\alpha}{\sum_i w_i}$$

where unit vector \mathbf{u}_i points in the i th look direction and has weight w_i , and principal unit vector \mathbf{u}_α is defined as $\alpha = \operatorname{argmax}_j w_j$.

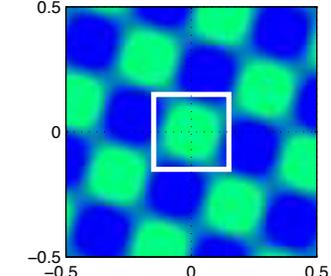
Plane wave
100%



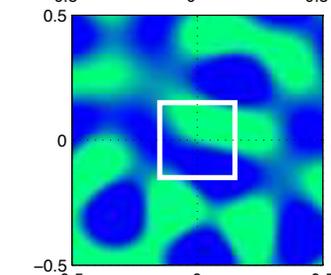
Point source
~90%



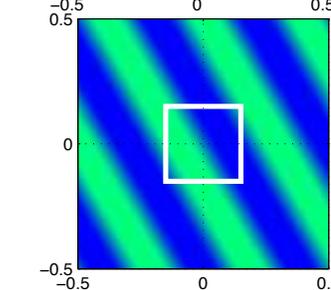
Orthogonal
50%



Diffuse
~0%



Standing wave
0%



Merimaa, 2007. Energetic sound field analysis of loudspeaker reproduction, AES
Jackson et al., 2013. Sound field planarity characterized by superdirective beamforming, POMA



Metrics summary

- Goodness of fit: **Mean squared error, MSE**
for least-squares optimization formulation
- Interference: **Contrast, C**
for assessing relative SPL or loudness
- Robustness: **Effort, E**
as a strong indicator for practical performance
- Bandwidth: **Frequency range, F**
for application-driven content coverage
- Directivity: **Planarity, η**
for spatial coherence at the listener





Fundamental Relationships under Ideal Conditions: Cost Function and System Setup



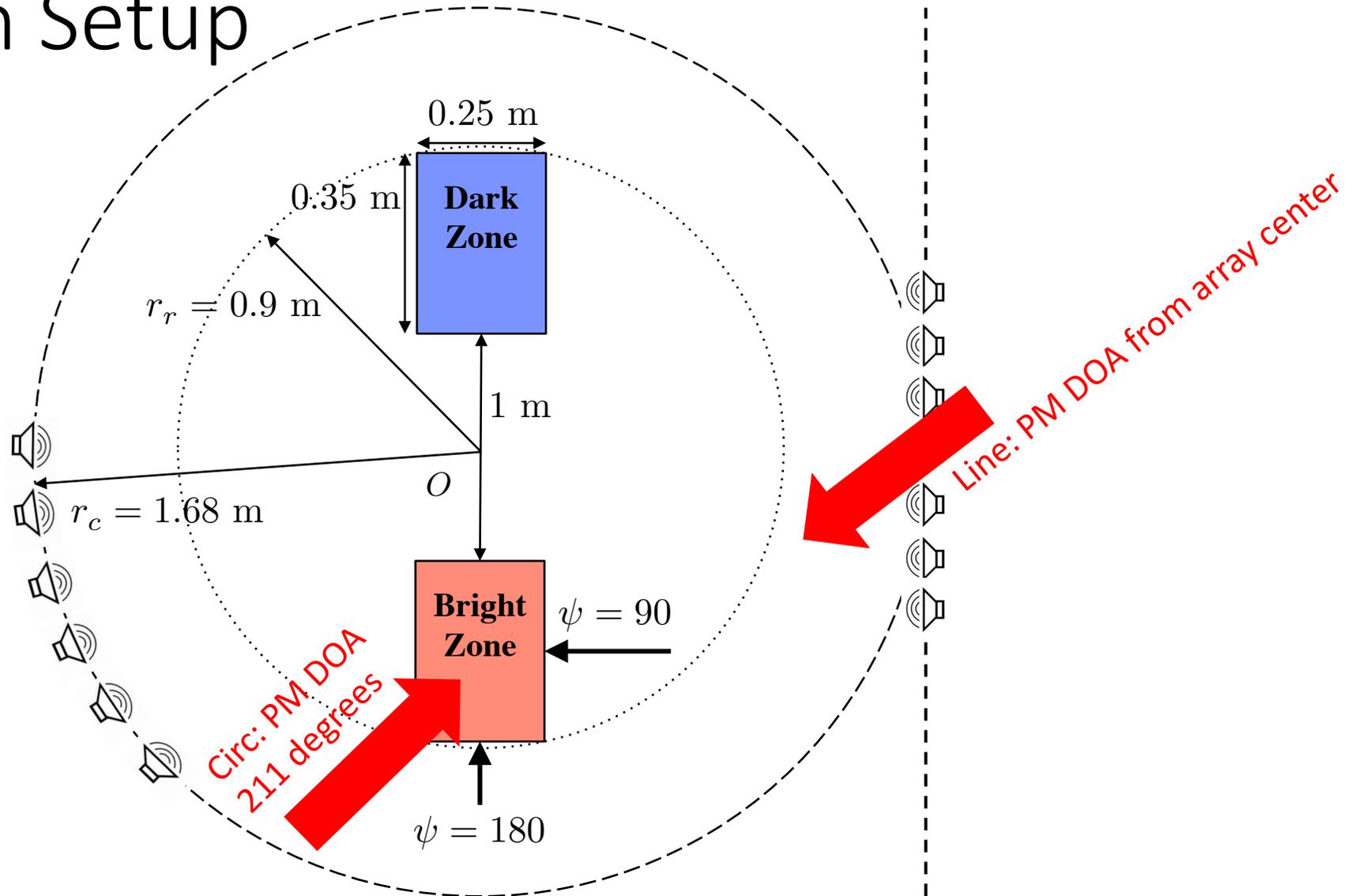


Aims

- Demonstrate the characteristics of ACC and PM:
 - Sound fields reproduced
 - Frequency range of operation
- Demonstrate the influence of loudspeaker array design:
 - Circular array (fully enclosed zones, relatively sparse loudspeaker spacing)
 - Line array (compact, relatively close loudspeaker spacing)



Simulation Setup





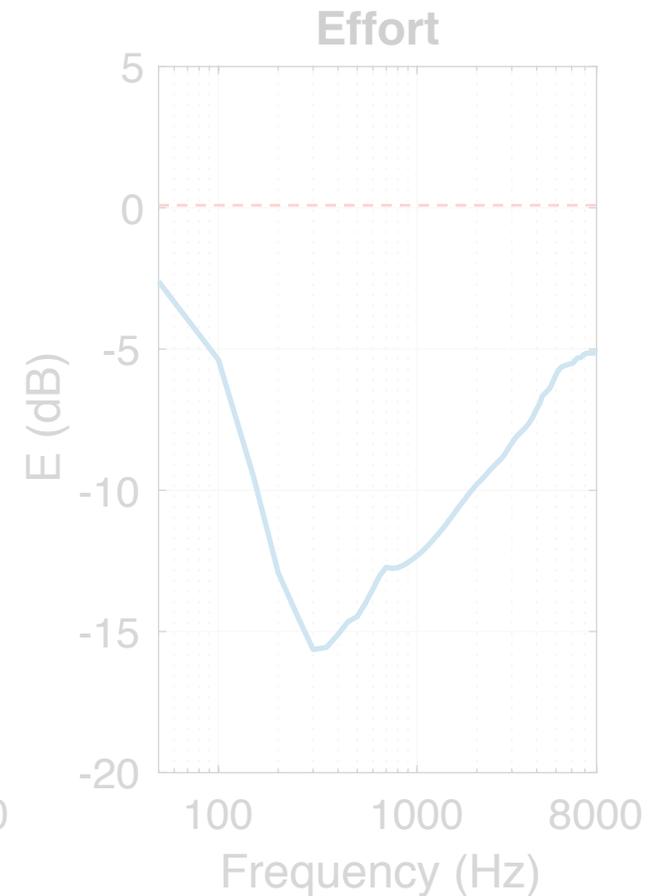
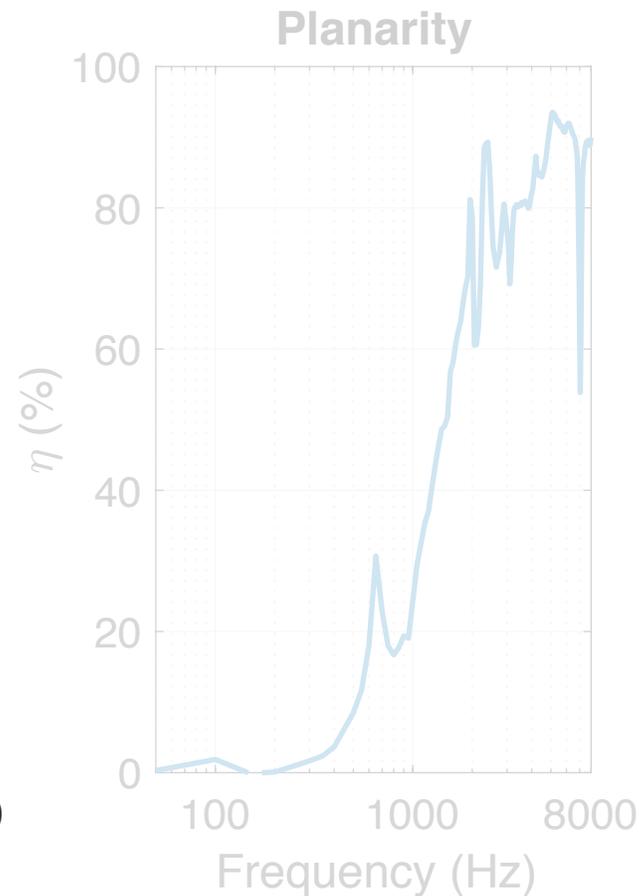
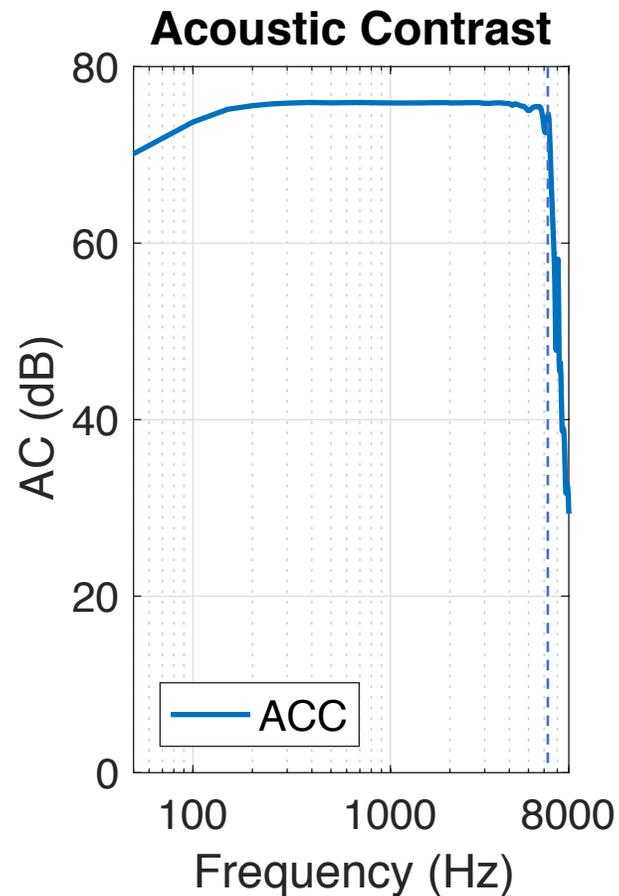
Simulation Setup

- Two arrays: line and circle
 - Circle: 60 loudspeakers, 1.68 m radius
 - Line: 8 loudspeakers, 10 cm spacing
- Two zones
- Regularization: Max matrix condition number of 10^{10} , Array Effort limit of 0 dB
- Listening level 76 dB SPL
- Acoustic contrast capped at 76 dB (i.e. no audible sound in dark zone)



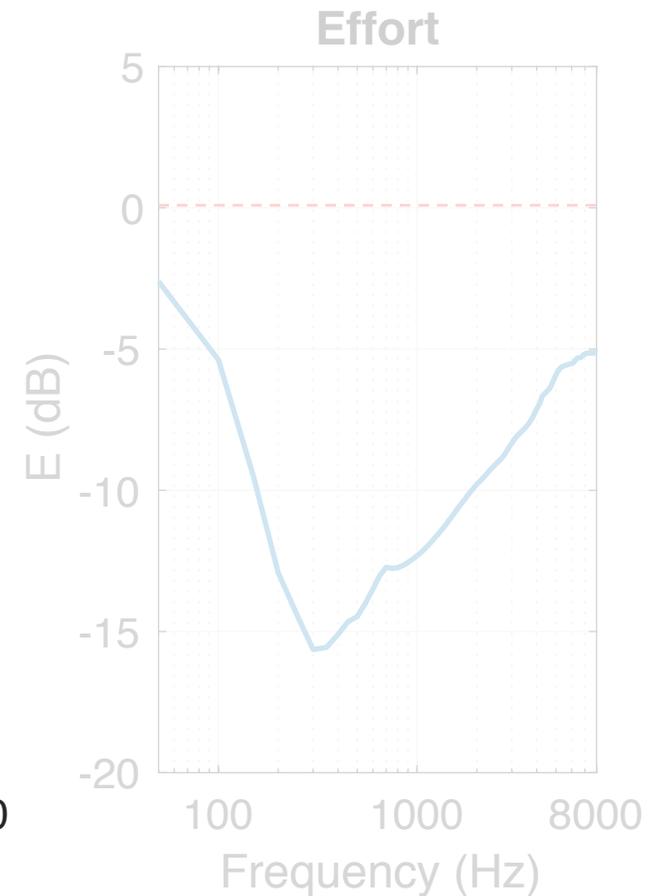
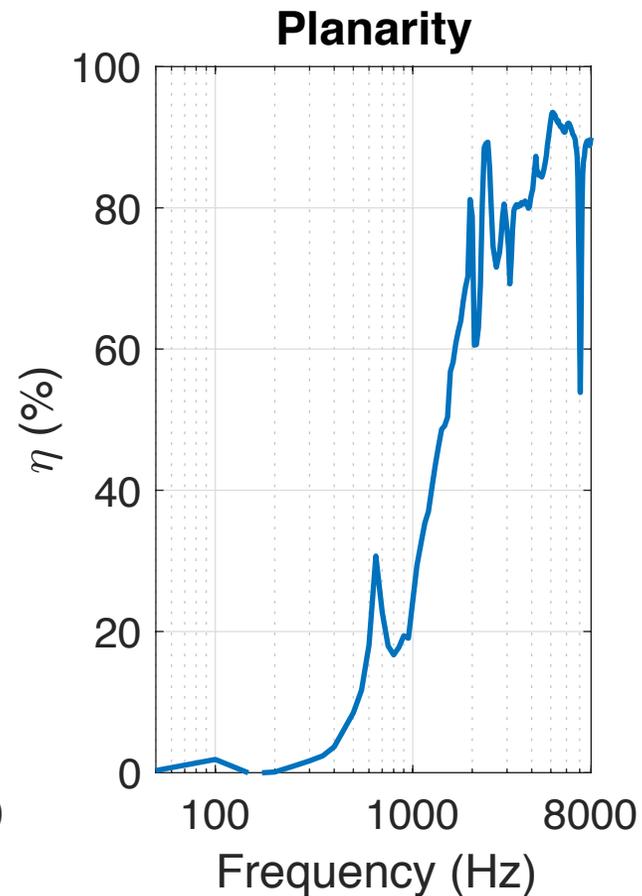
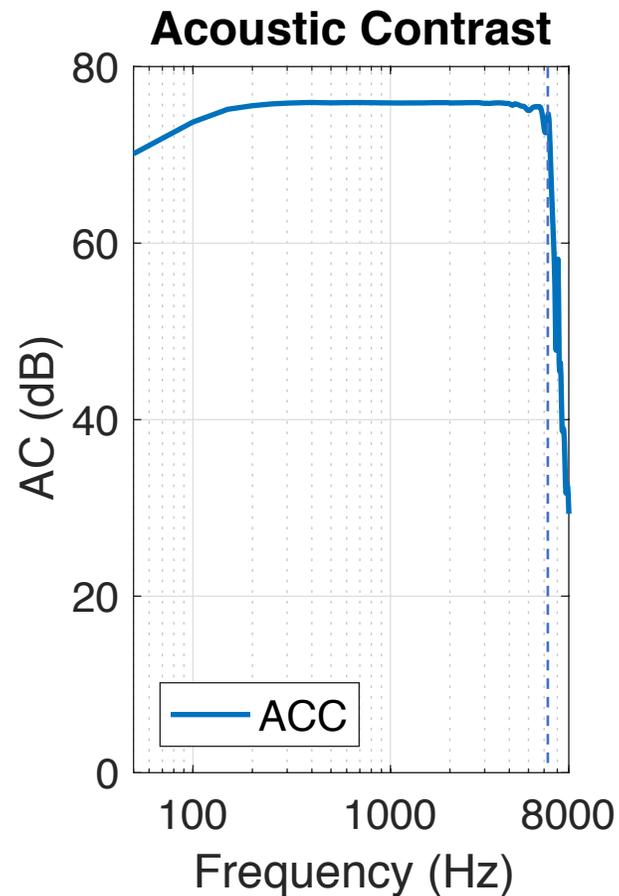
Acoustic Contrast Control: Circular Array

- Performance over frequency



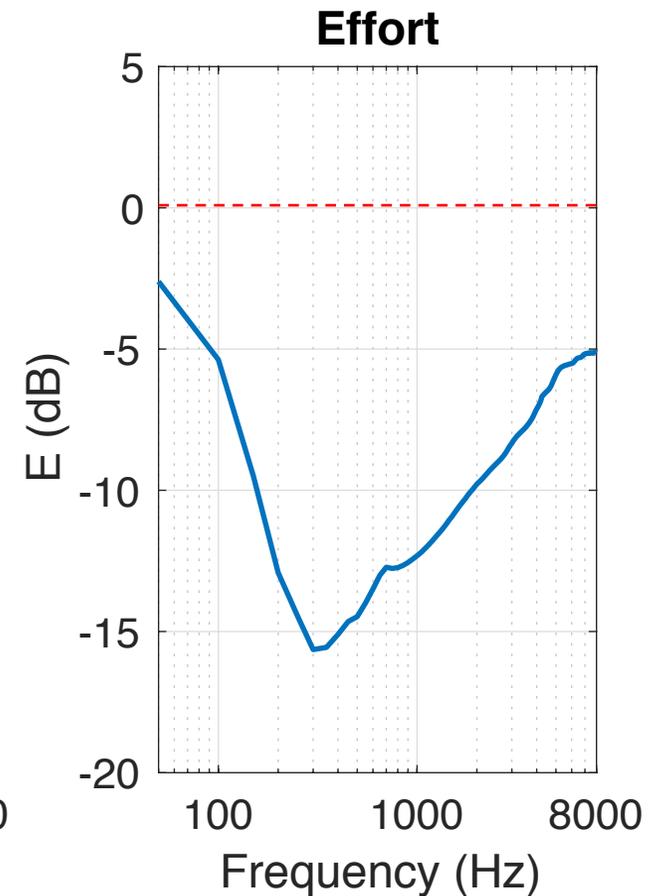
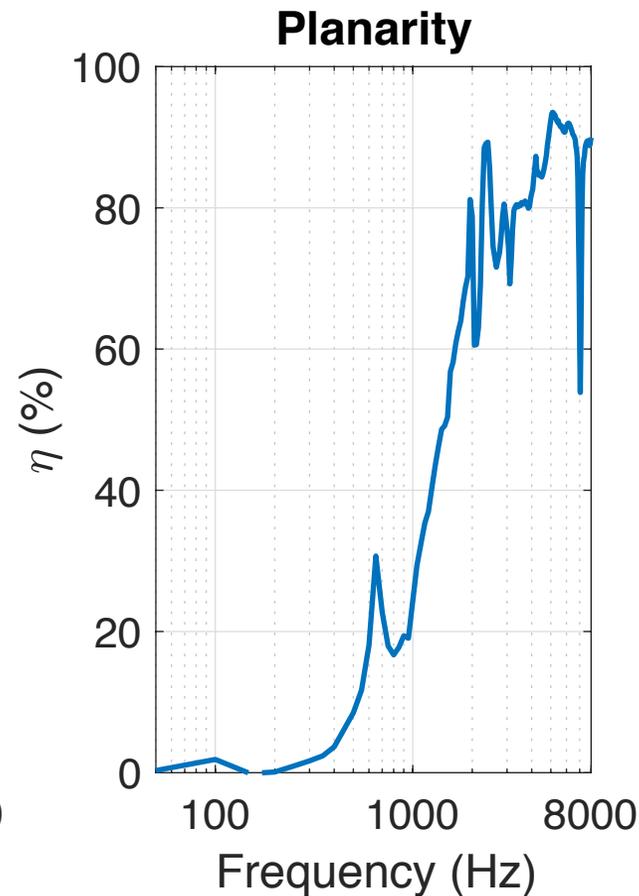
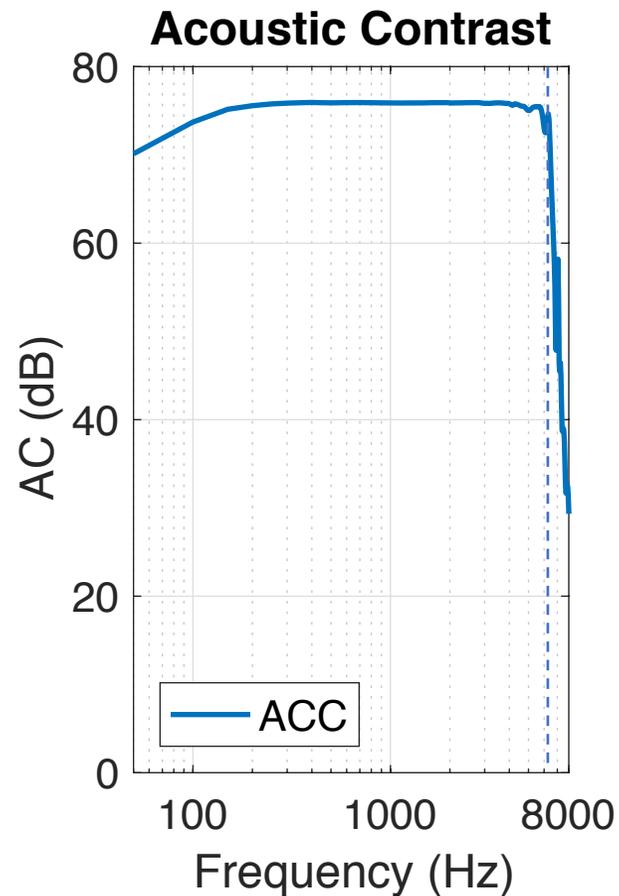
Acoustic Contrast Control: Circular Array

- Performance over frequency



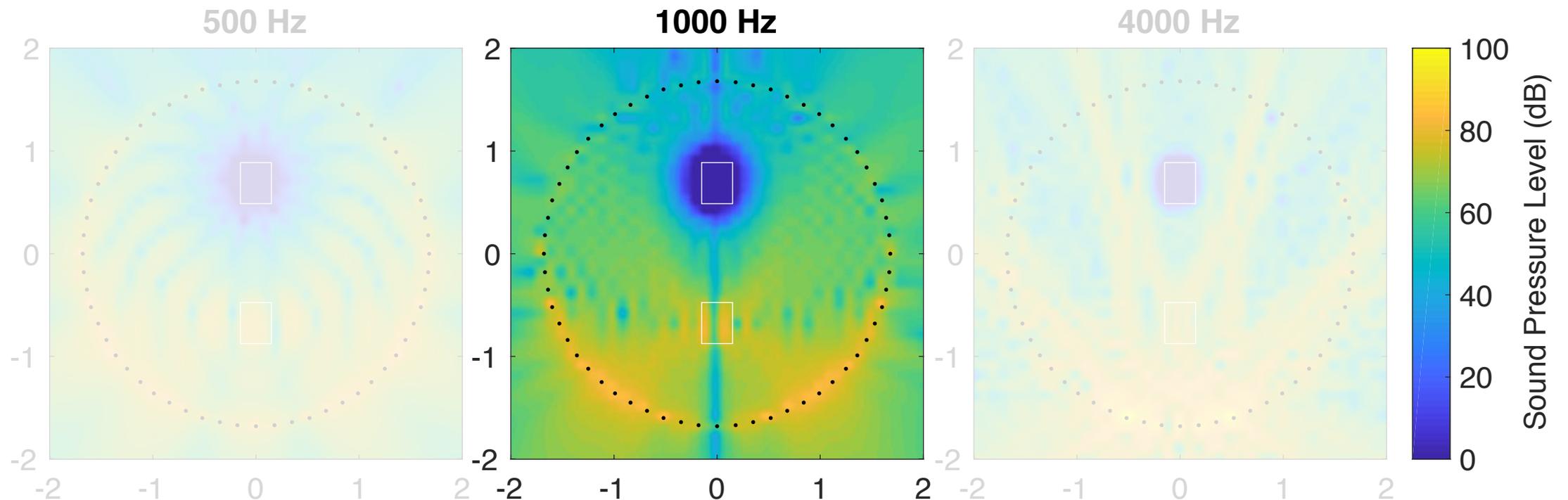
Acoustic Contrast Control: Circular Array

- Performance over frequency



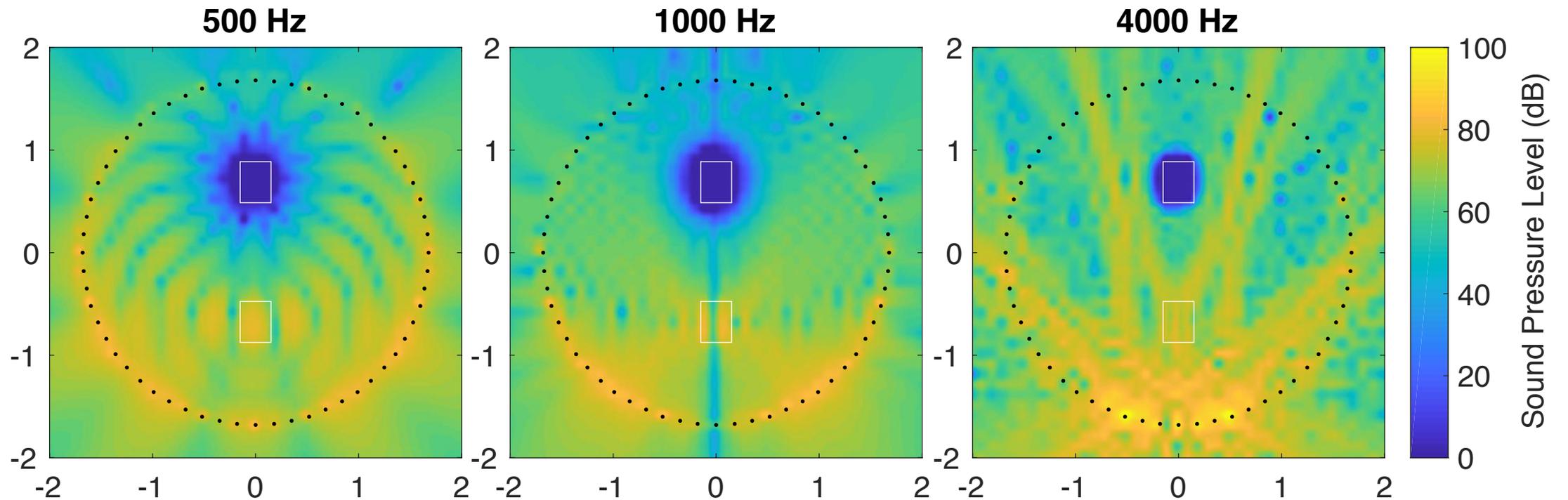
Acoustic Contrast Control: Circular Array

- Sound field: sound pressure level



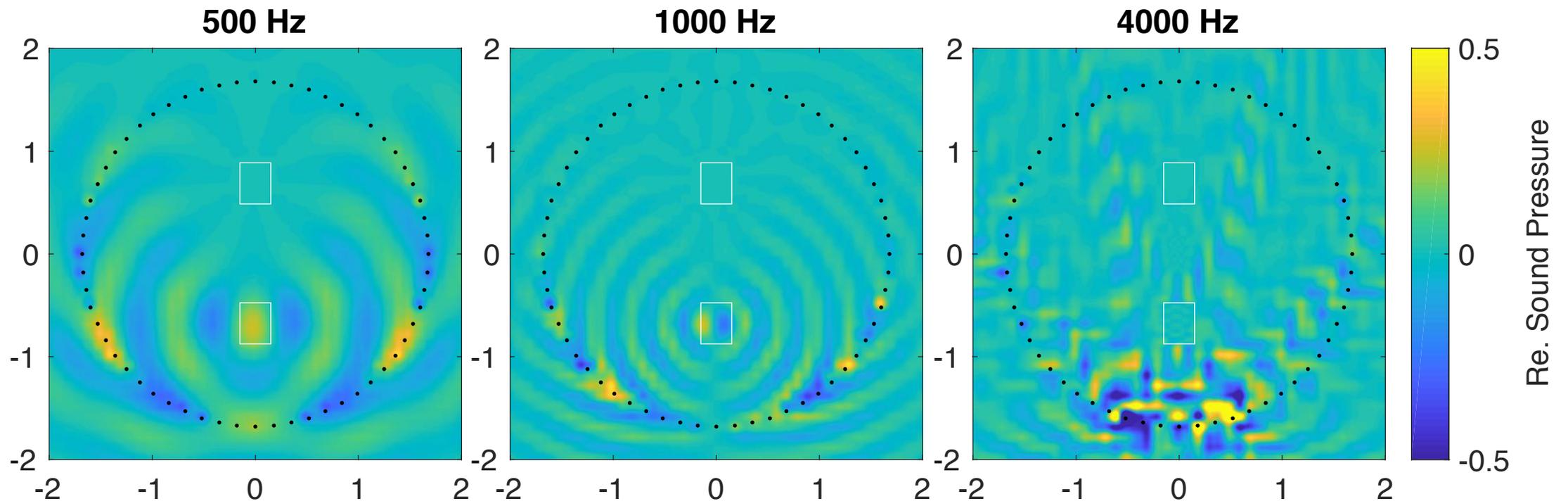
Acoustic Contrast Control: Circular Array

- Sound field: sound pressure level



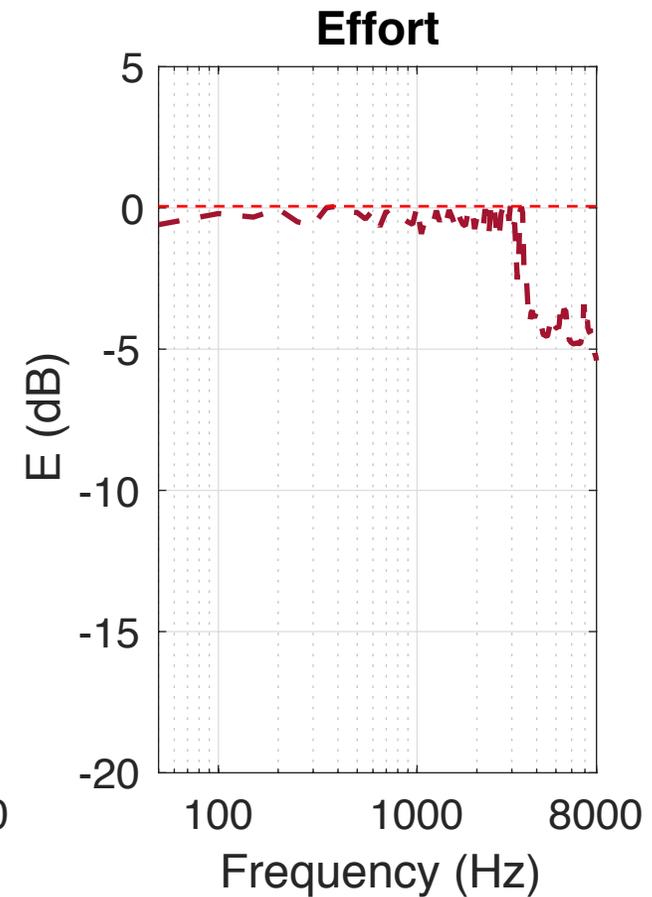
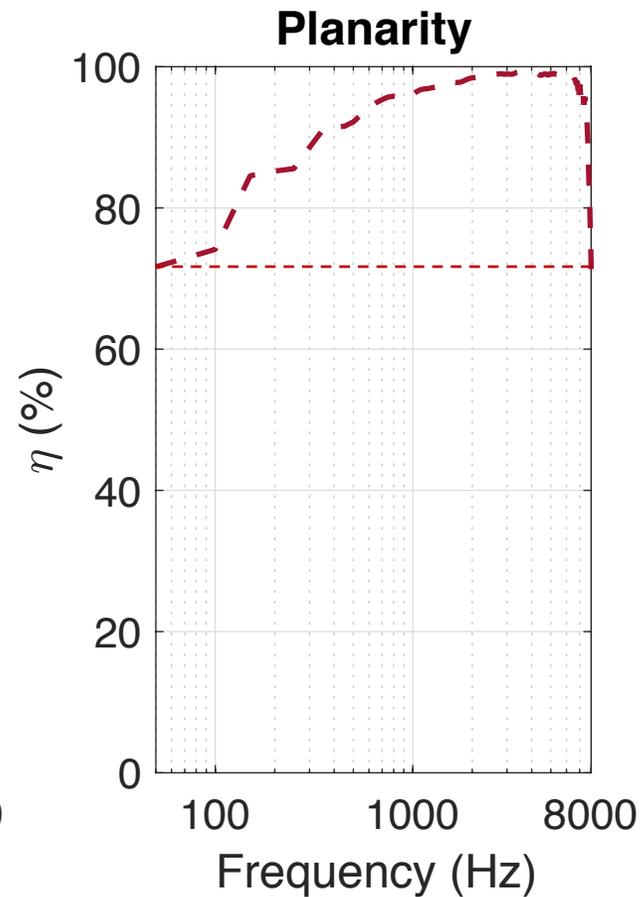
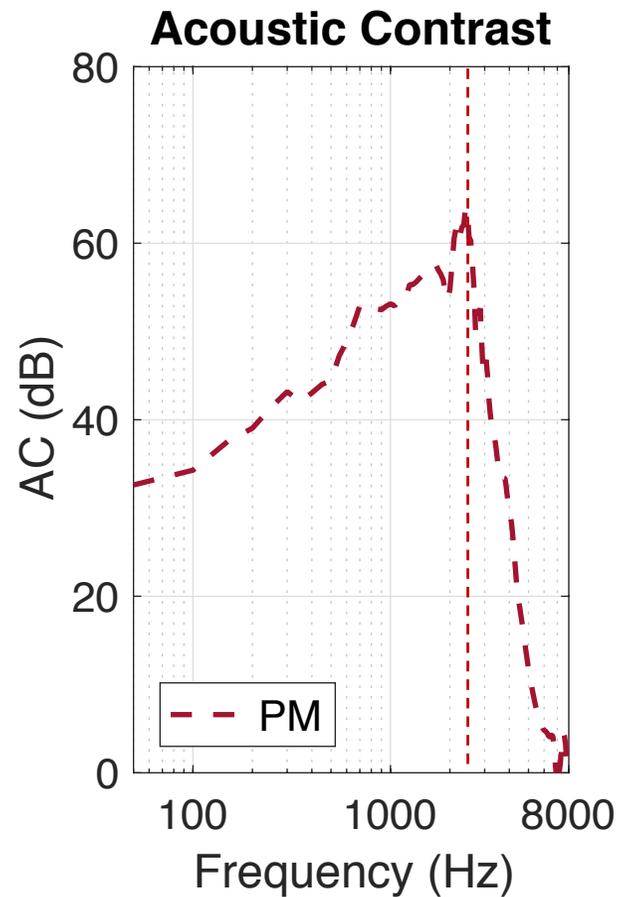
Acoustic Contrast Control: Circular Array

- Sound field: real part of the sound pressure



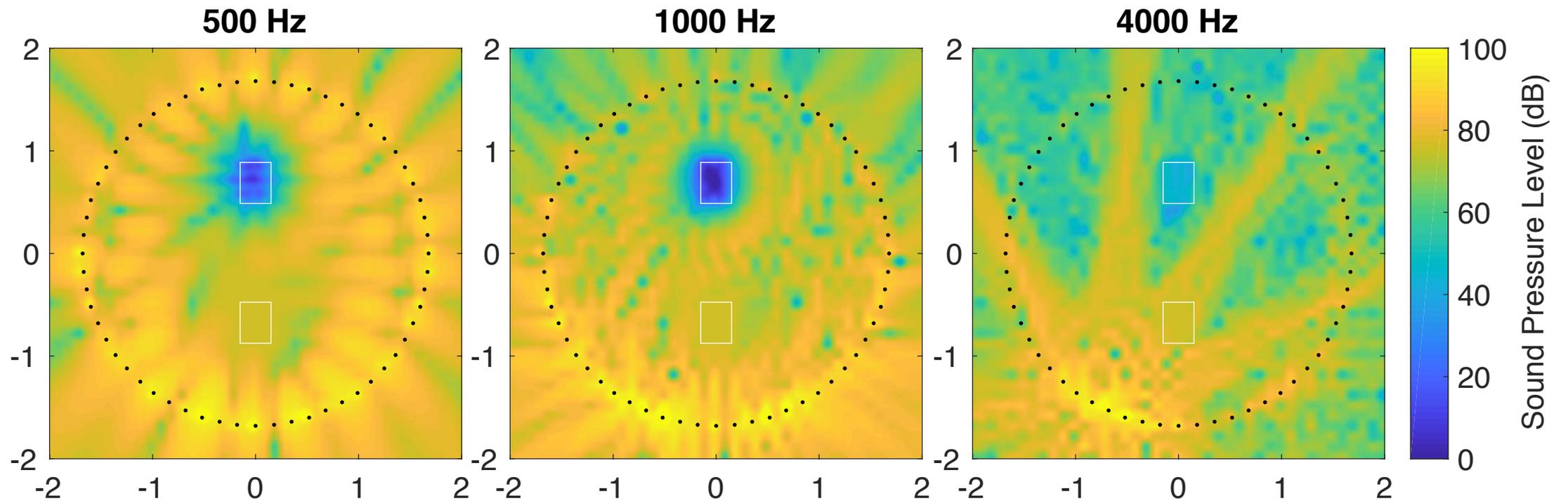
Pressure Matching: Circular Array

- Performance over frequency



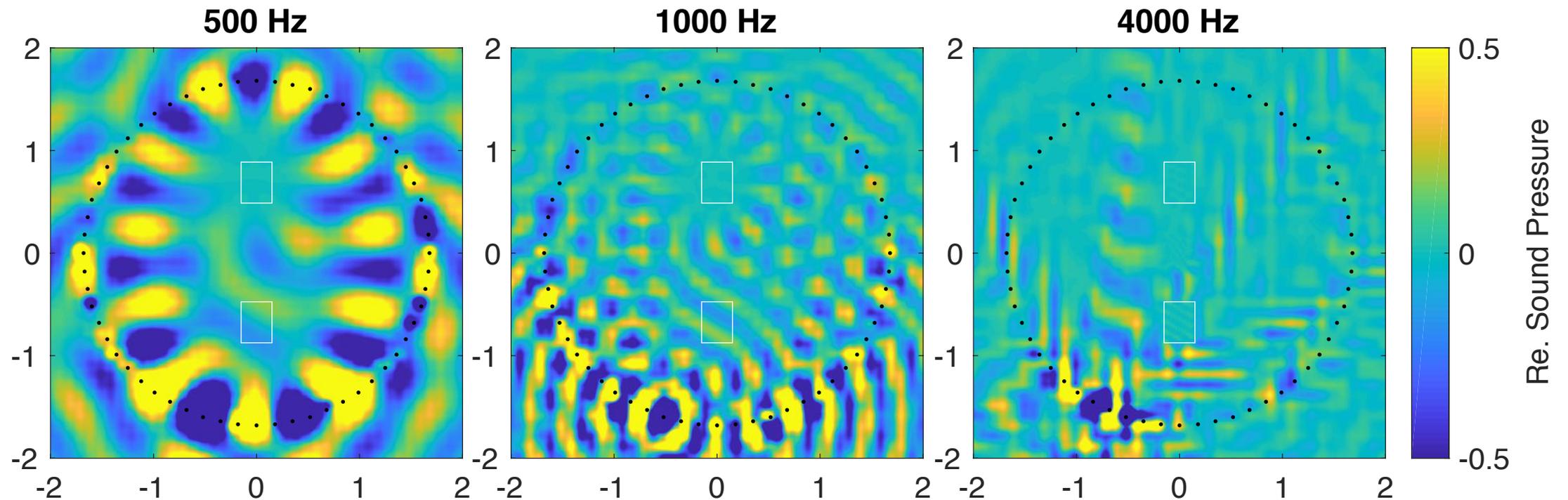
Pressure Matching: Circular Array

- Sound field: sound pressure level



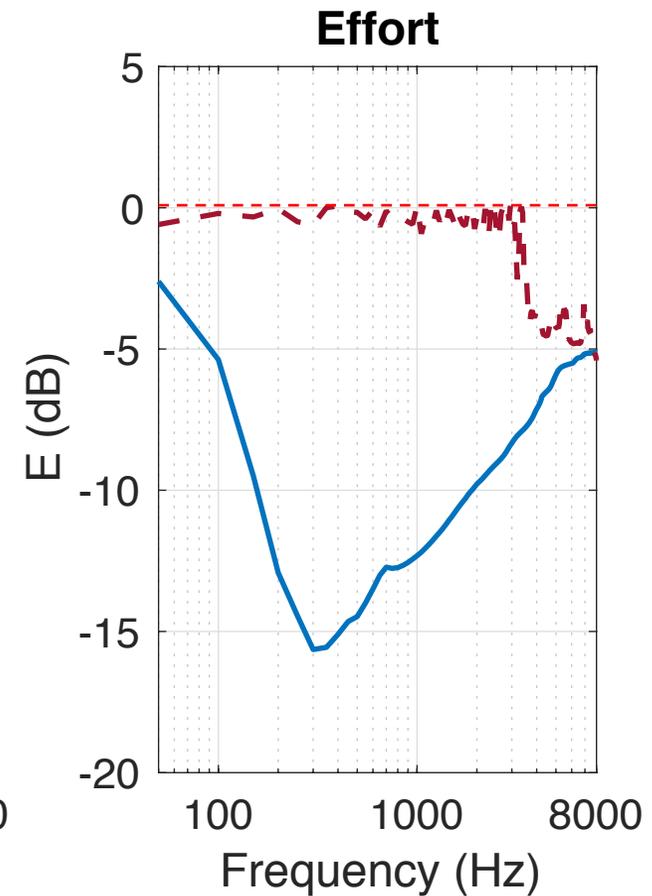
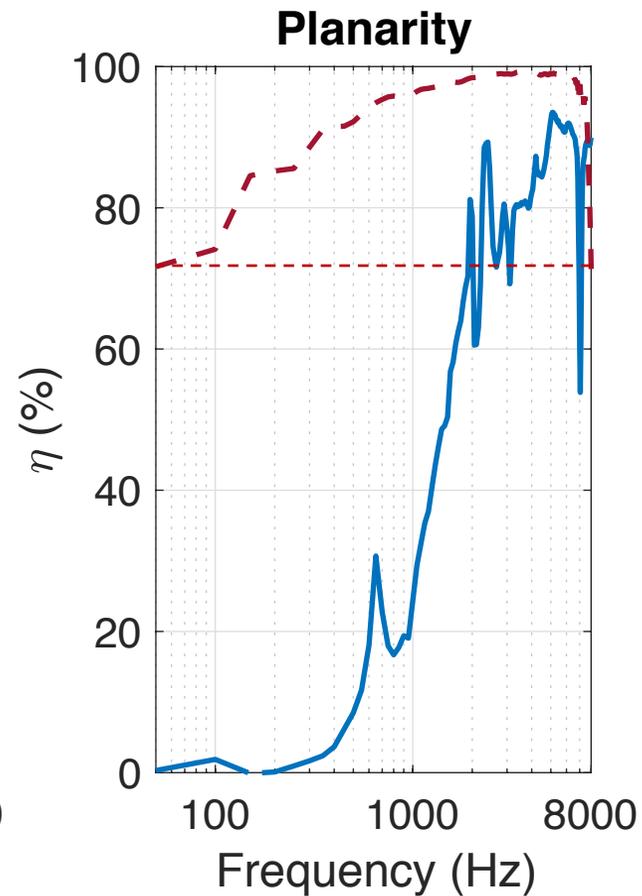
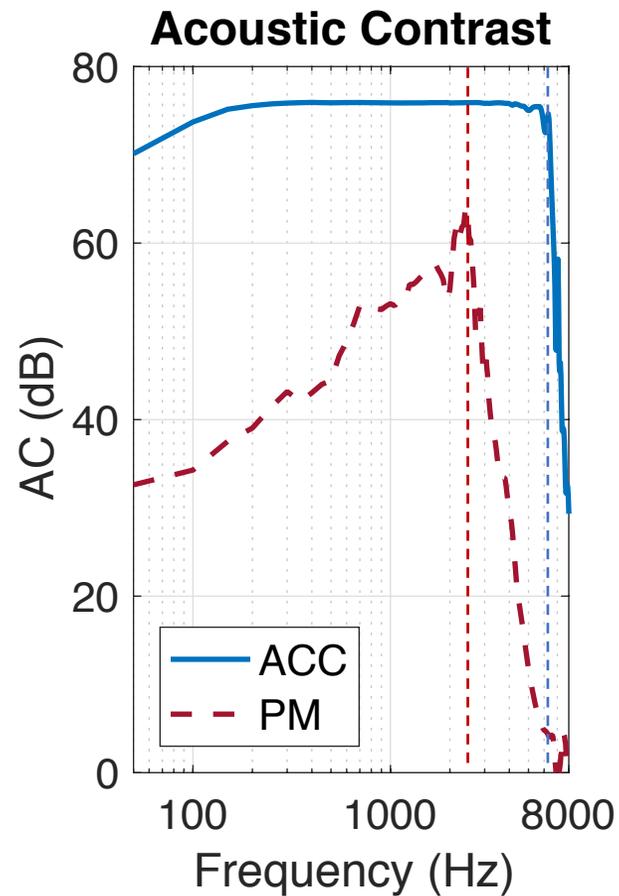
Pressure Matching: Circular Array

- Sound field: real part of the sound pressure



Performance Comparison: Circular Array

- Control of energy vs control of phase





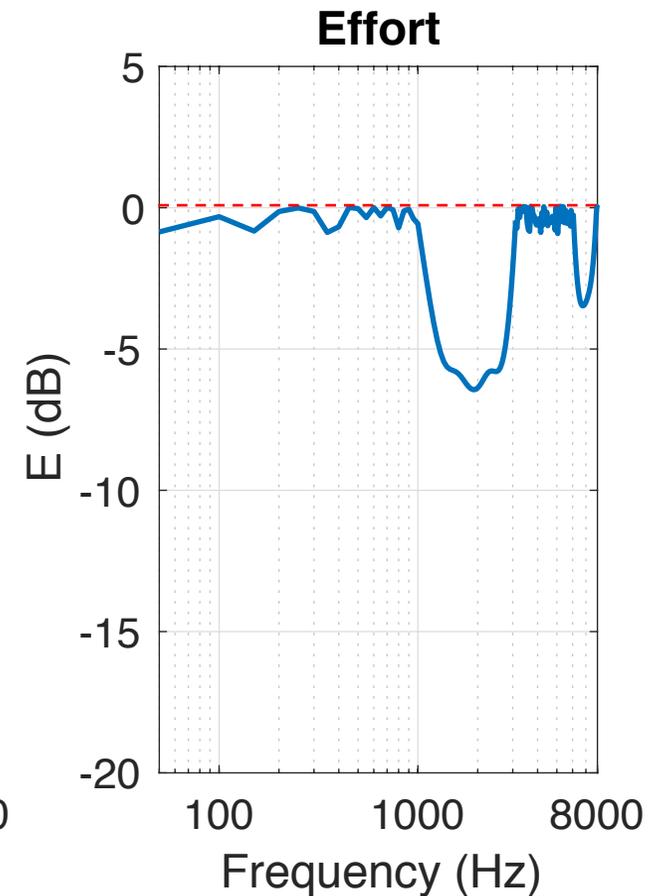
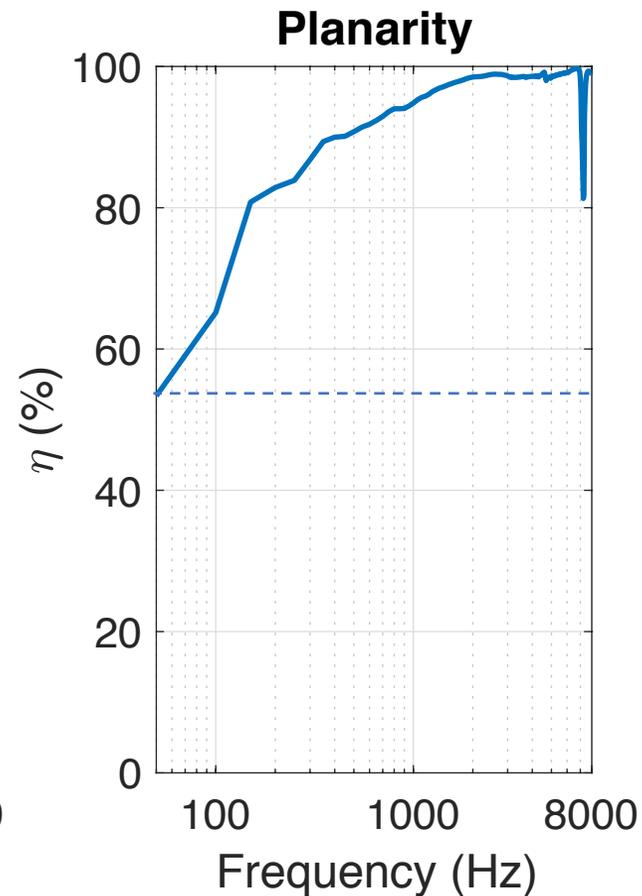
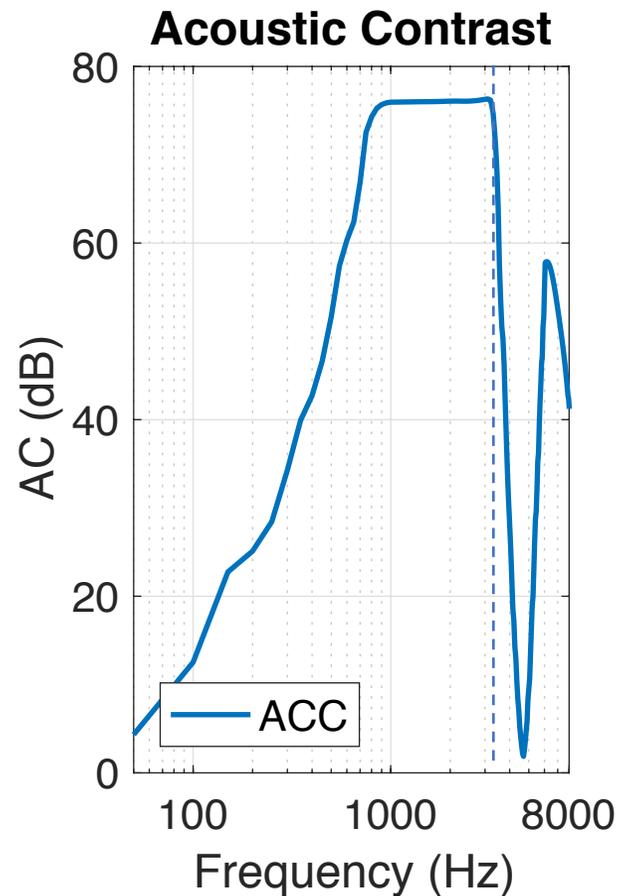
Summary: Circular Array

- ACC controls energy only; PM controls energy and phase
 - ACC gives excellent contrast
 - ACC controls contrast over a wide frequency range
 - PM gives excellent control of sound in the bright zone (homogenous, planar)
 - PM requires higher effort filters to achieve this control



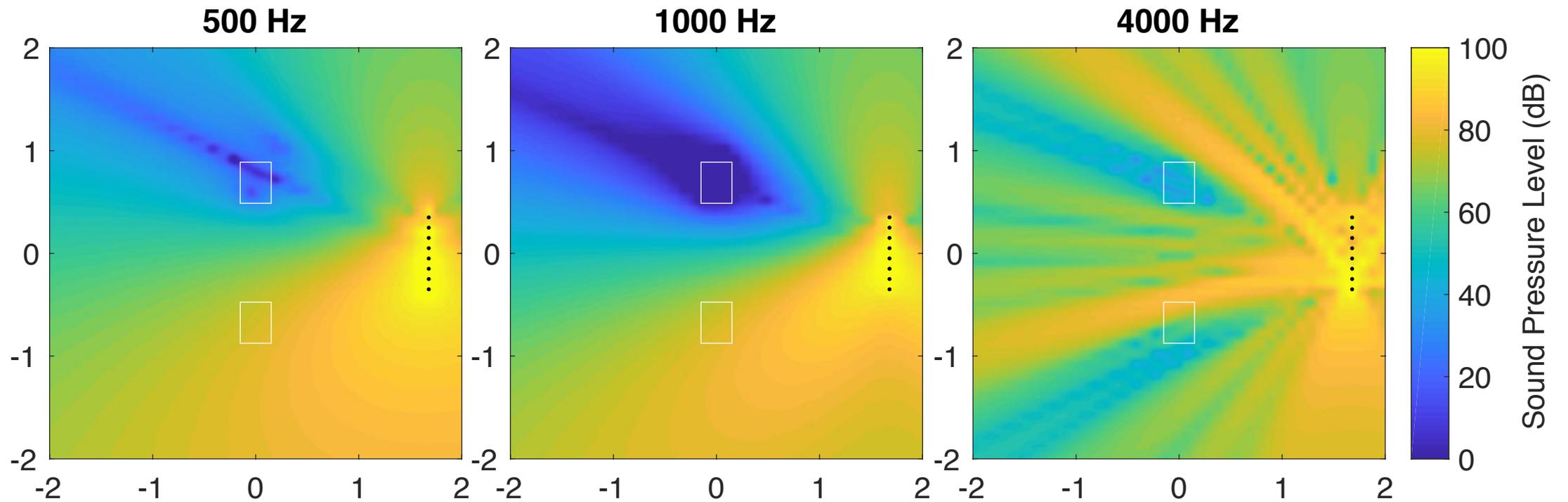
Acoustic Contrast Control: Line Array

- Performance over frequency



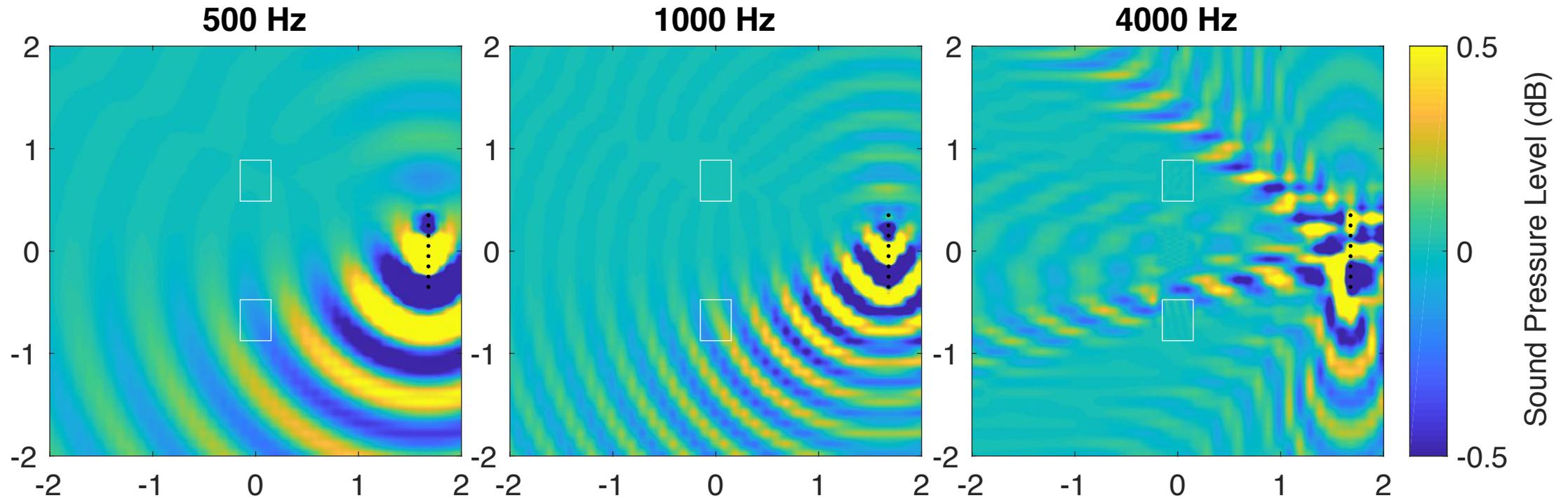
Acoustic Contrast Control: Line Array

- Sound field: sound pressure level



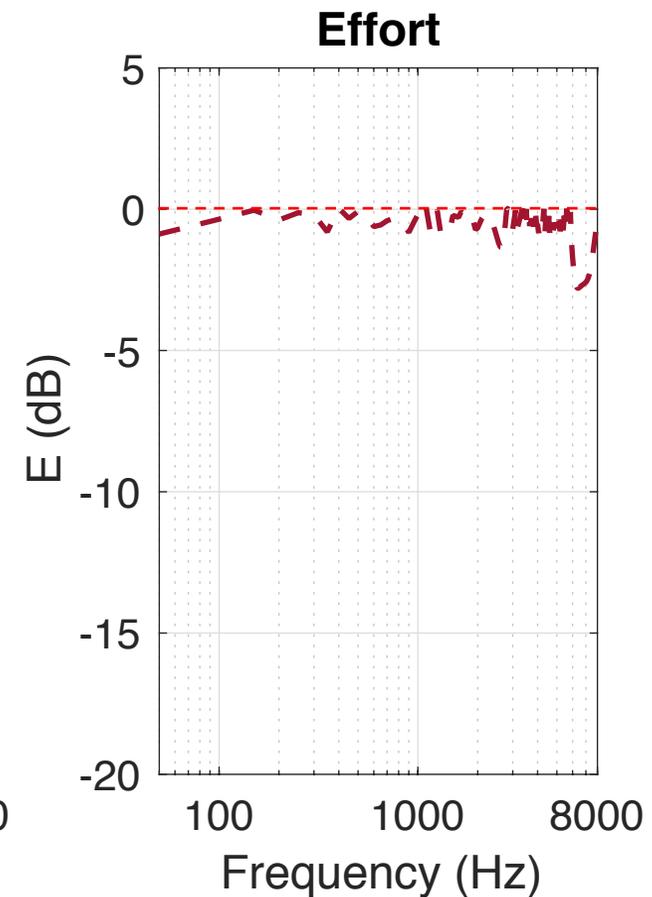
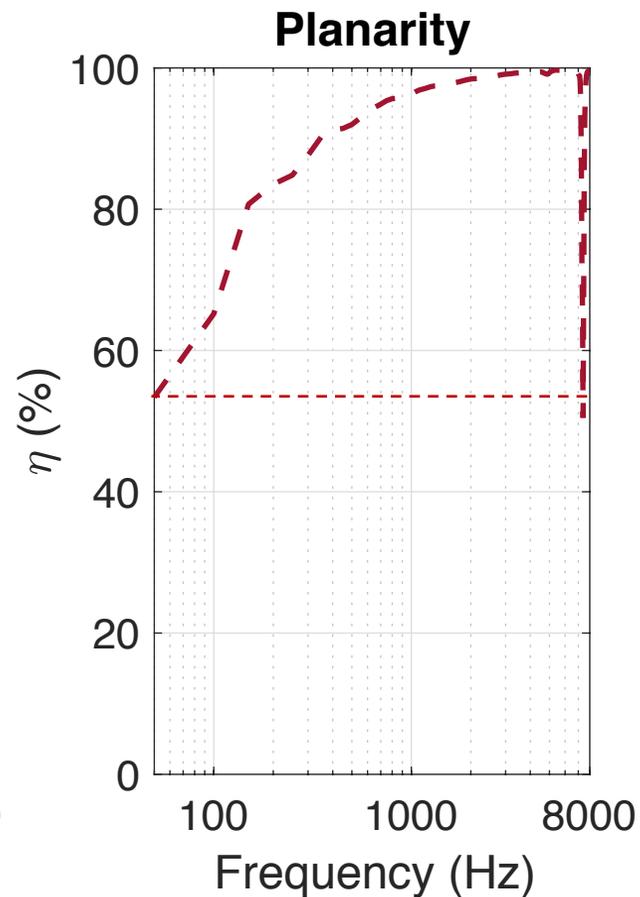
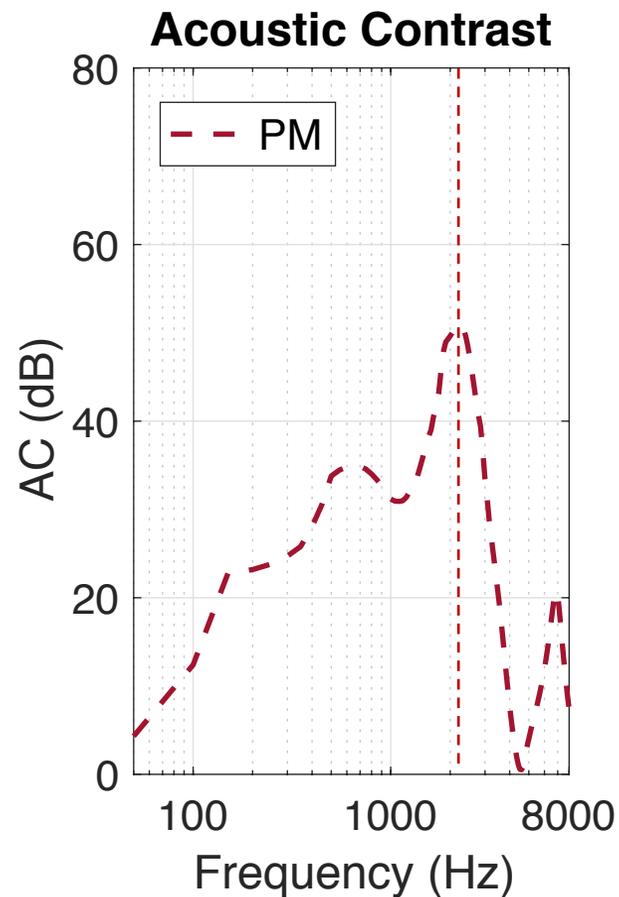
Acoustic Contrast Control: Line Array

- Sound field: real part of the sound pressure



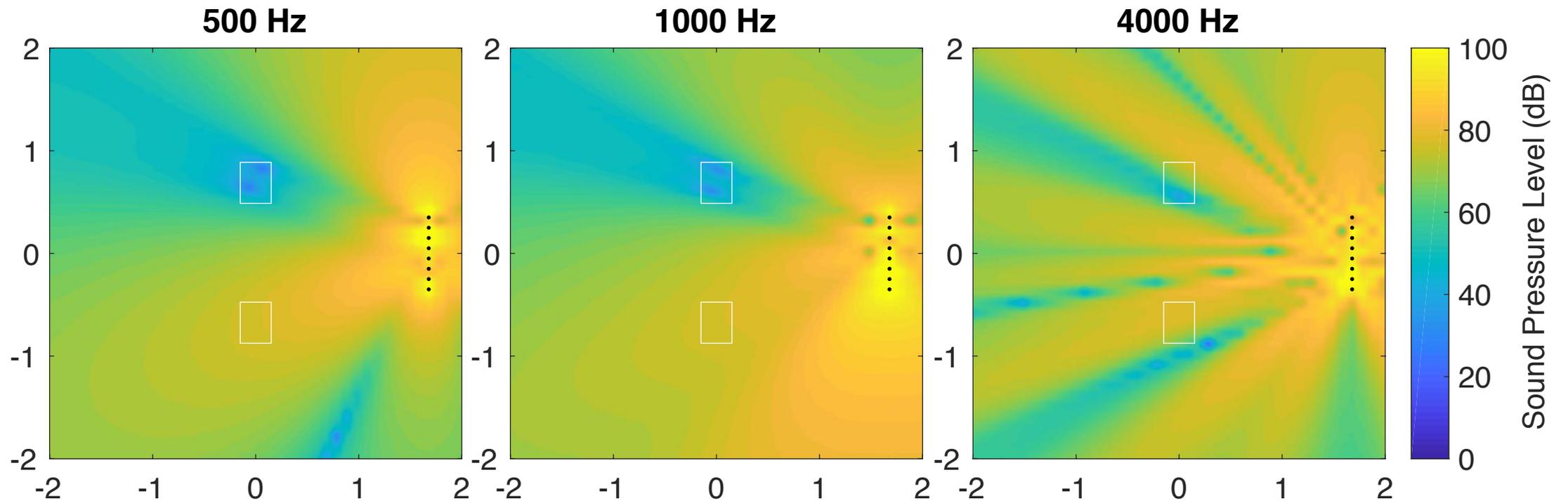
Pressure Matching: Line Array

- Performance over frequency



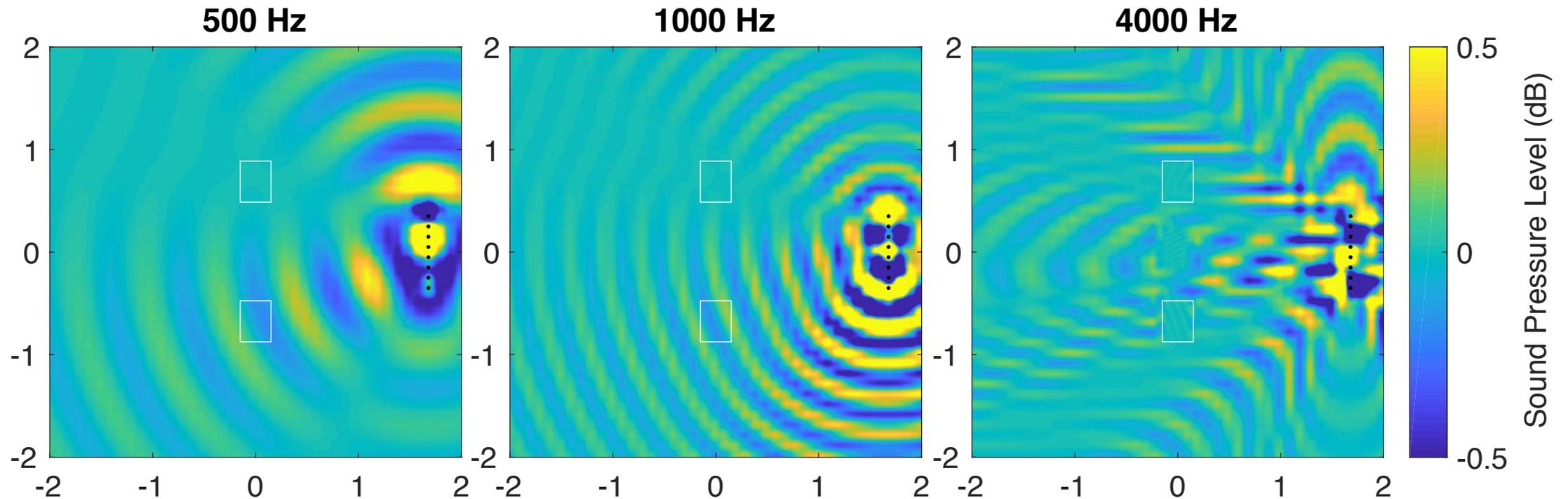
Pressure Matching: Line Array

- Sound field: sound pressure level



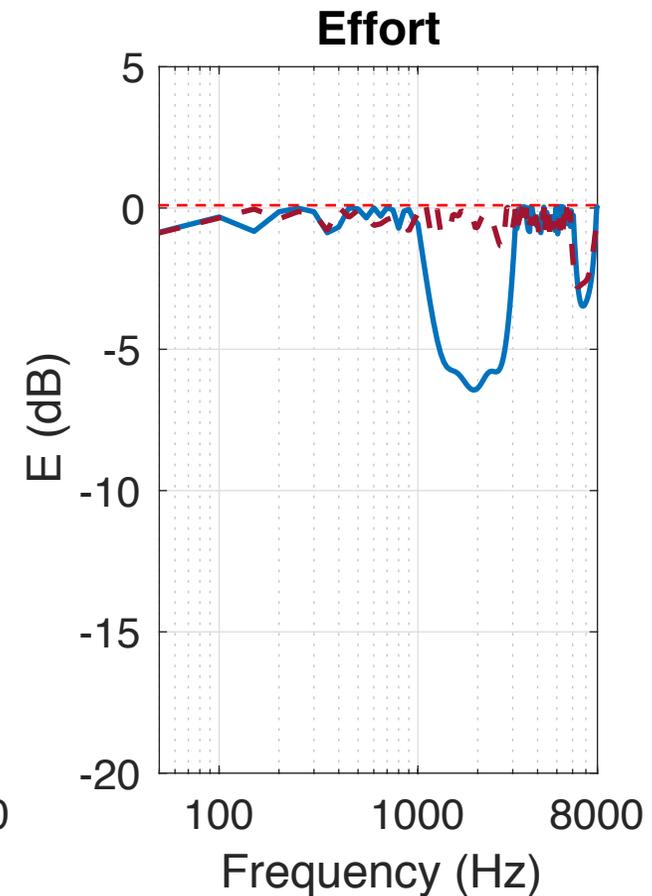
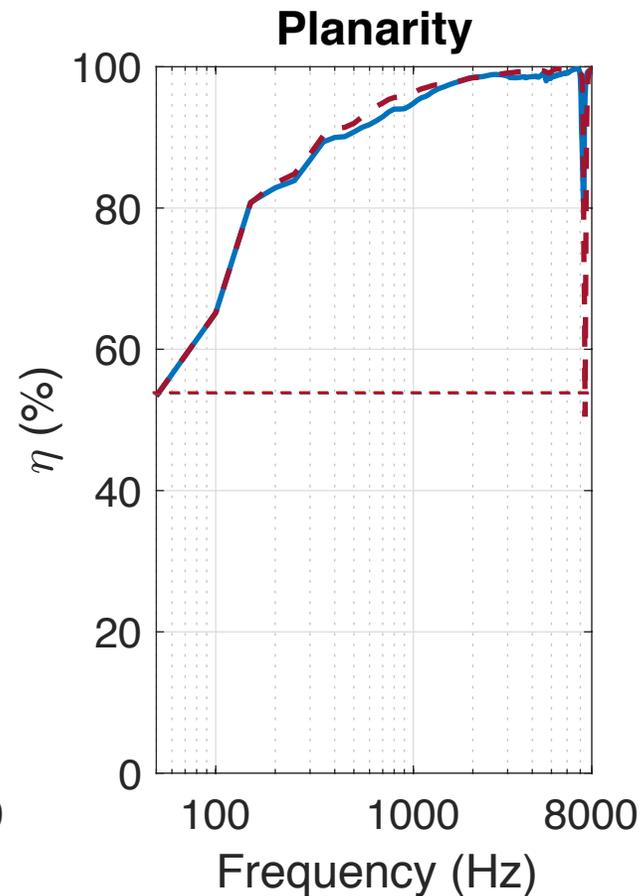
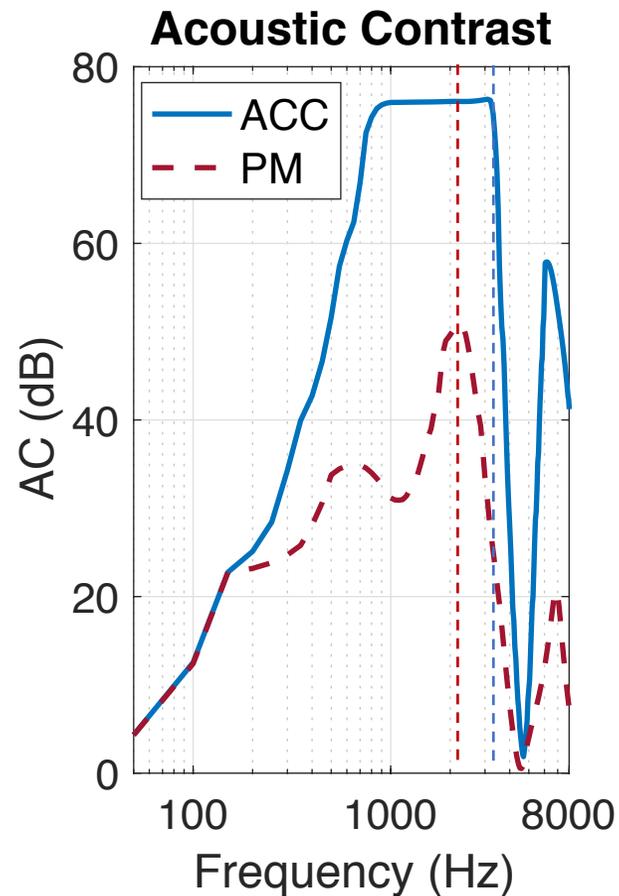
Pressure Matching: Line Array

- Sound field: real part of the sound pressure



Performance Comparison: Line Array

- Control of energy vs control of phase (within array capability)





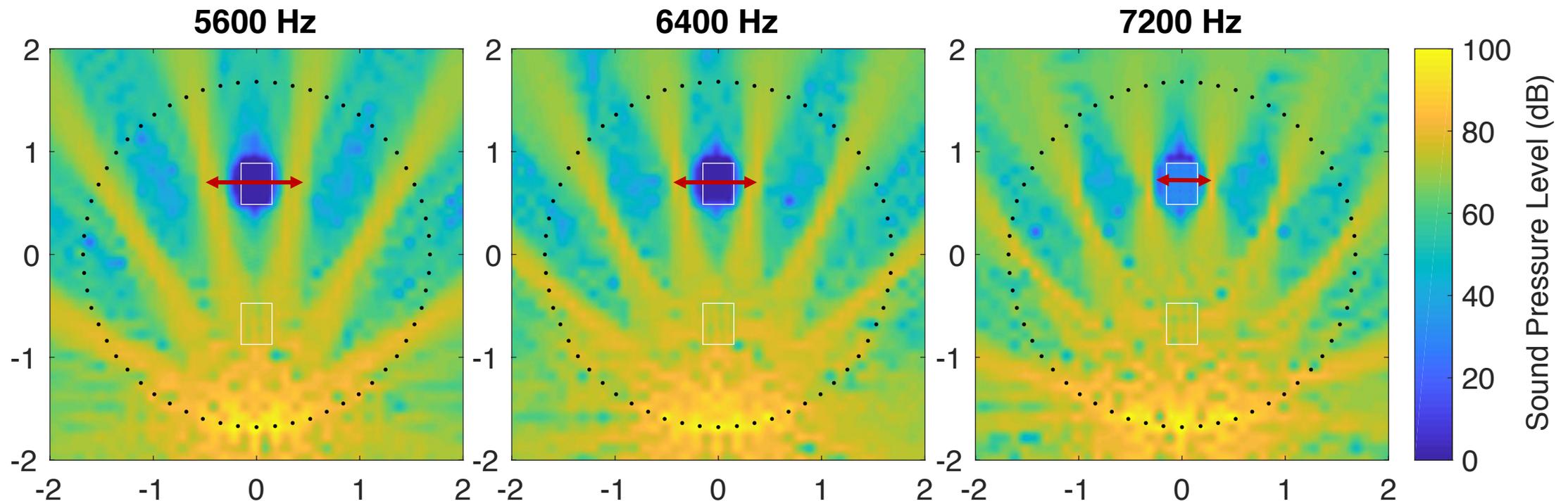
Summary: Line Array

- ACC controls energy only; PM controls energy and phase
 - ACC gives excellent contrast
 - But only reaches the maximum contrast over a relatively narrow frequency range
 - ACC controls contrast over a wide frequency range
 - Extra bandwidth compared to PM not as significant as for circular array
 - Low frequency contrast performance the same for both methods
 - PM gives excellent control of sound in the bright zone (homogenous, planar)
 - But ACC still planar due to limited possible DOAs into zone
 - PM requires higher effort filters to achieve this control
 - But ACC also high effort at most frequencies



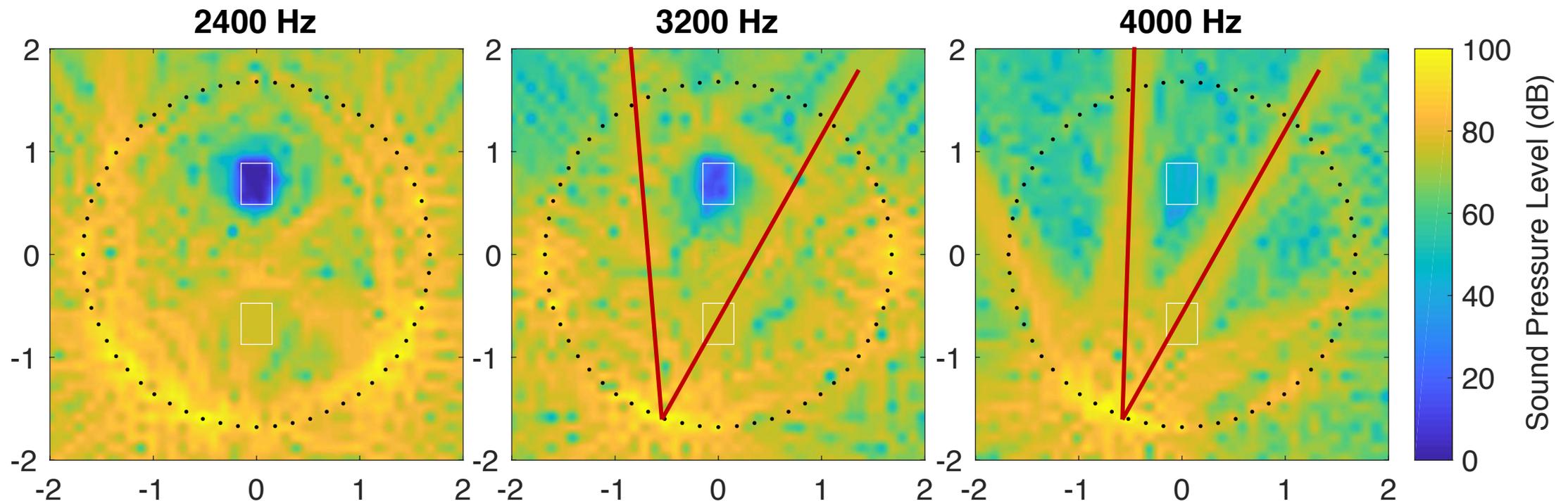
Loudspeaker Array Design

- Circular array degradation of acoustic contrast performance
 - ACC: onset when aliasing lobe separation equals dark zone diameter



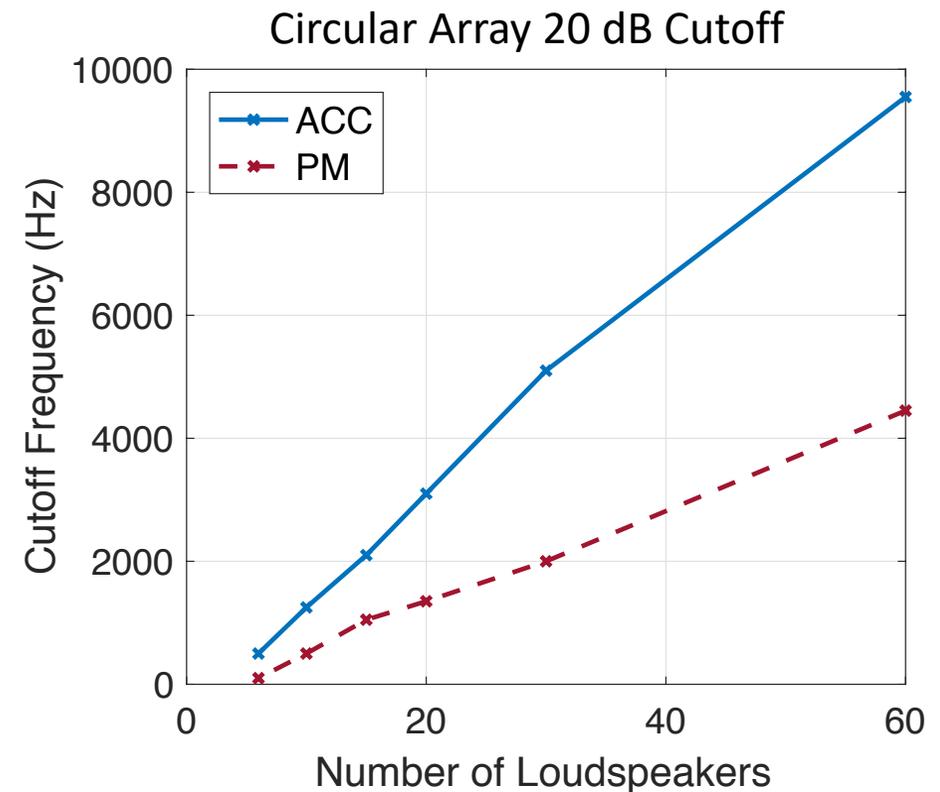
Loudspeaker Array Design

- Circular array degradation of acoustic contrast performance
 - **PM**: onset when aliasing lobe begins to impinge on dark zone



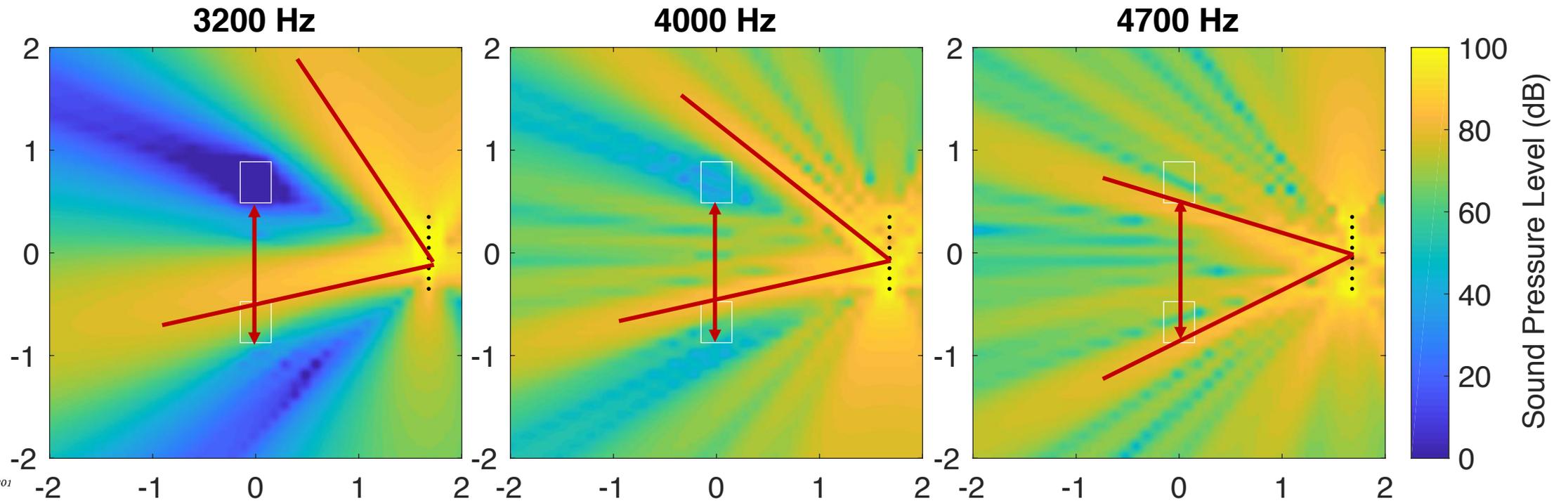
Loudspeaker Array Design

- Circular array:
 - ACC moves target beam direction to avoid aliasing for as long as possible
 - Aliasing depends on dark zone size
 - PM target direction specified in optimization
 - Aliasing depends on loudspeaker separation and overall control region size
 - To compare, consider the frequency at which contrast falls below 20 dB
- Recent work by Winter et al. to predict spatial aliasing for SFS



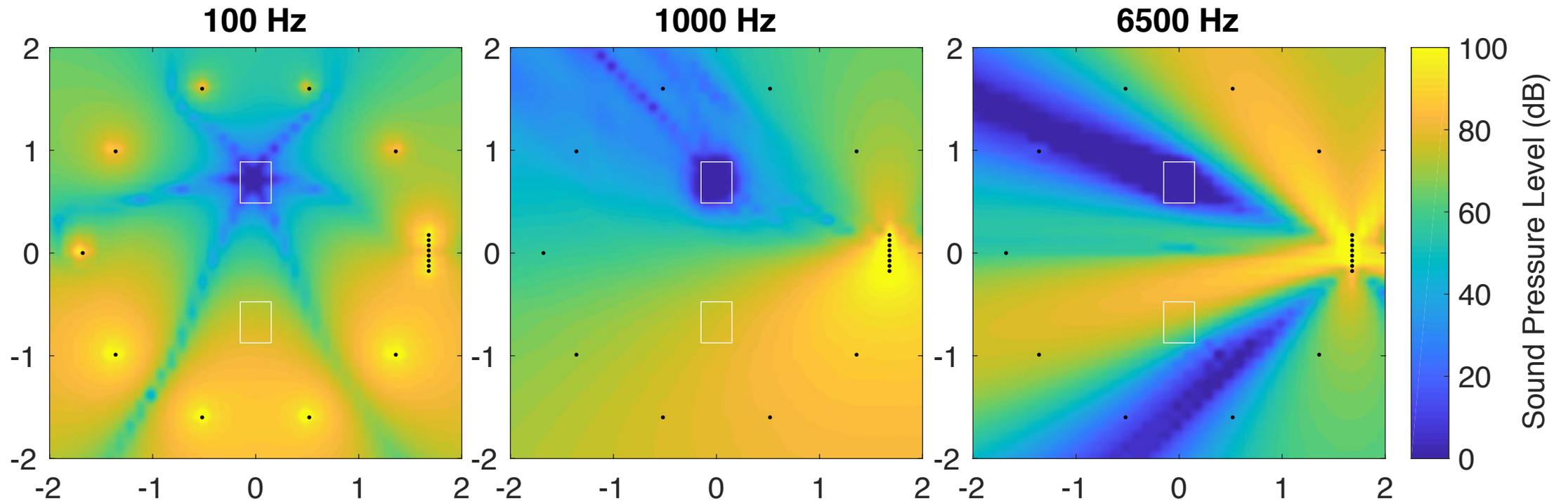
Loudspeaker Array Design

- Line array relationships with wavelength are well known:
 - Array aperture and regularization limit low frequency performance
 - Loudspeaker and zone spacing limit high frequency performance



Loudspeaker Array Design

- So we need a large aperture and close spacing 🤔



Summary: Performance Comparison

- **ACC**: creates maximal contrast between the zones, at the cost of having no control over the phase
 - High contrast
 - Low planarity (circular array)
 - Relatively efficient
- **PM**: aims to control the pressure amplitude and phase at each control microphone
 - Lower contrast
 - High planarity
 - Relatively inefficient
 - Weighted versions of PM have also been proposed
- Array geometry imposes fundamental limits on performance



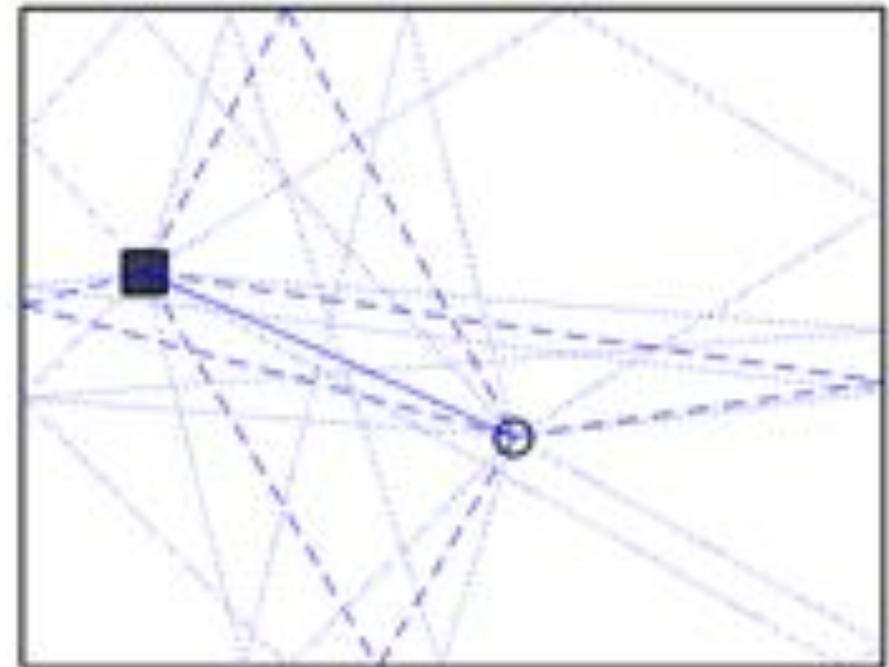
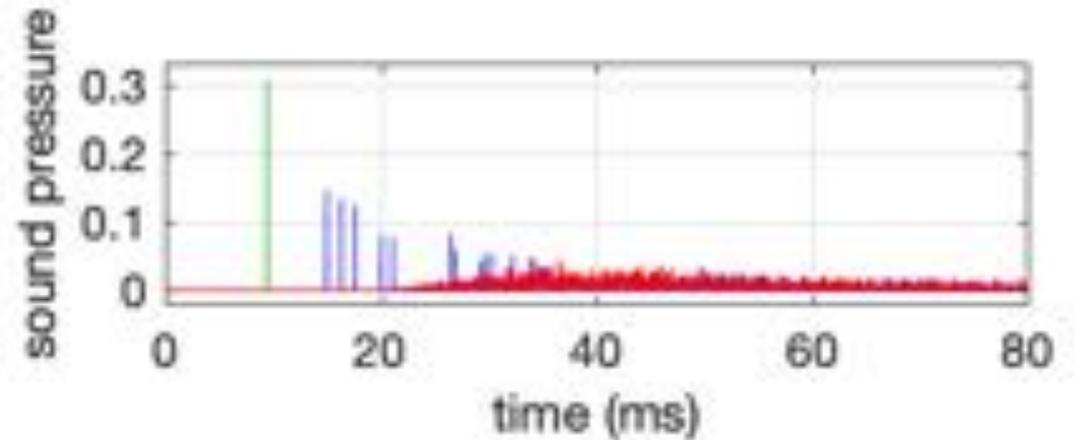
Dealing with room reverberation





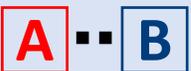
Structure of a room impulse response

- **Direct sound**
 - First, clean and narrow
 - From source direction
- **Early reflections**
 - Initially sparse, thickening
 - Absorbed and scattered
 - From all walls, up/down
- **Late reverberation**
 - Dense and diffuse
 - Directionless after mixing time



Ironing out the earlies

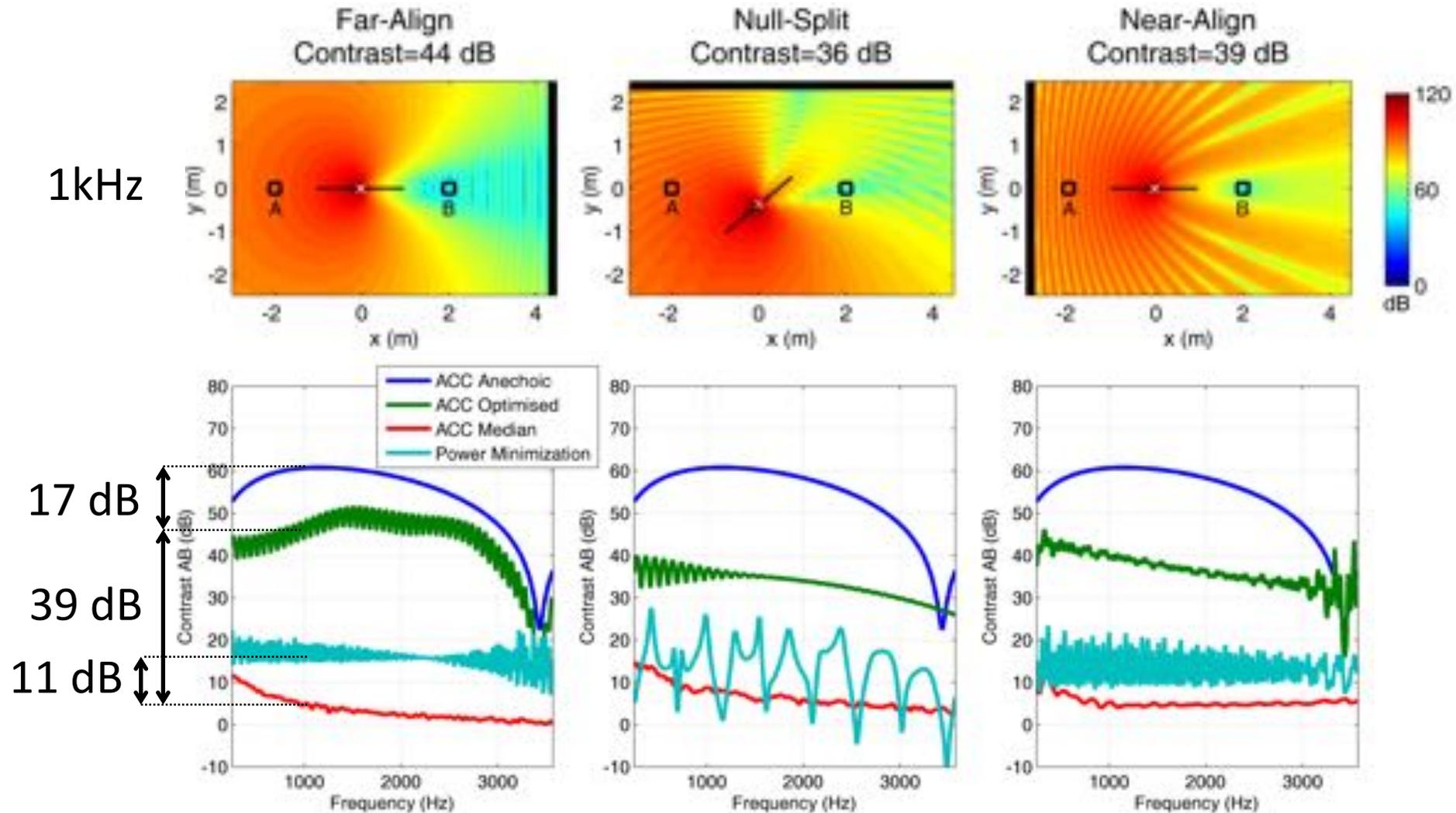
- Early reflections (coherent) colour the response
- Reflections add unwanted copies of loudspeakers as image sources

	Reduced control effort [Elliott et al., 2012, TASLP] or acoustic power [Jones & Elliott, 2008, JASA]
	Array close to zones [Chang et al., 2009, JASA]
	Directive sources [Simón Gálvez et al., 2012, JASA]
	Superdirective beamforming [Elliott et al., 2010, JASA]

- For strong low order reflections, it may be beneficial to
 - Reduce radiation in specific directions
 - Maximize effectiveness of room equalization for specific reflection paths
- This can be achieved by geometrical optimization of source positions.

Geometrical considerations

- Arrangement of loudspeakers in relation to reflectors
- An unwanted reflection can be avoided, averted or cancelled
- Higher-order reflections use source modes

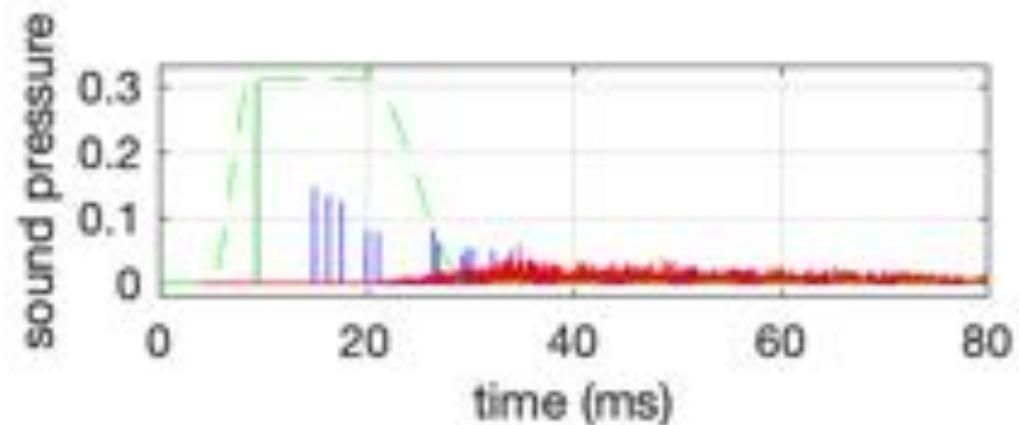


Late reverberation

There are two main strategies to mitigate effects of late reverberation:

- Truncation during calibration

By shortening the calibration RIRs used to calculate the personal sound zone filters



- Suppression through regularization

By applying regularization to make the filters less sensitive to convolutive noise during playback



Room effect summary

- Acoustic environments add early reflections and late reverberation
- Left untreated, effects can severely degrade system performance
- Mitigation strategies can be deployed for partial recovery:
 1. Acoustically treat/select the room
 2. Arrange your setup to avoid, avert or cancel reflections
 3. Truncate reverberation from the RIRs in calibration
 4. Apply regularization wisely...



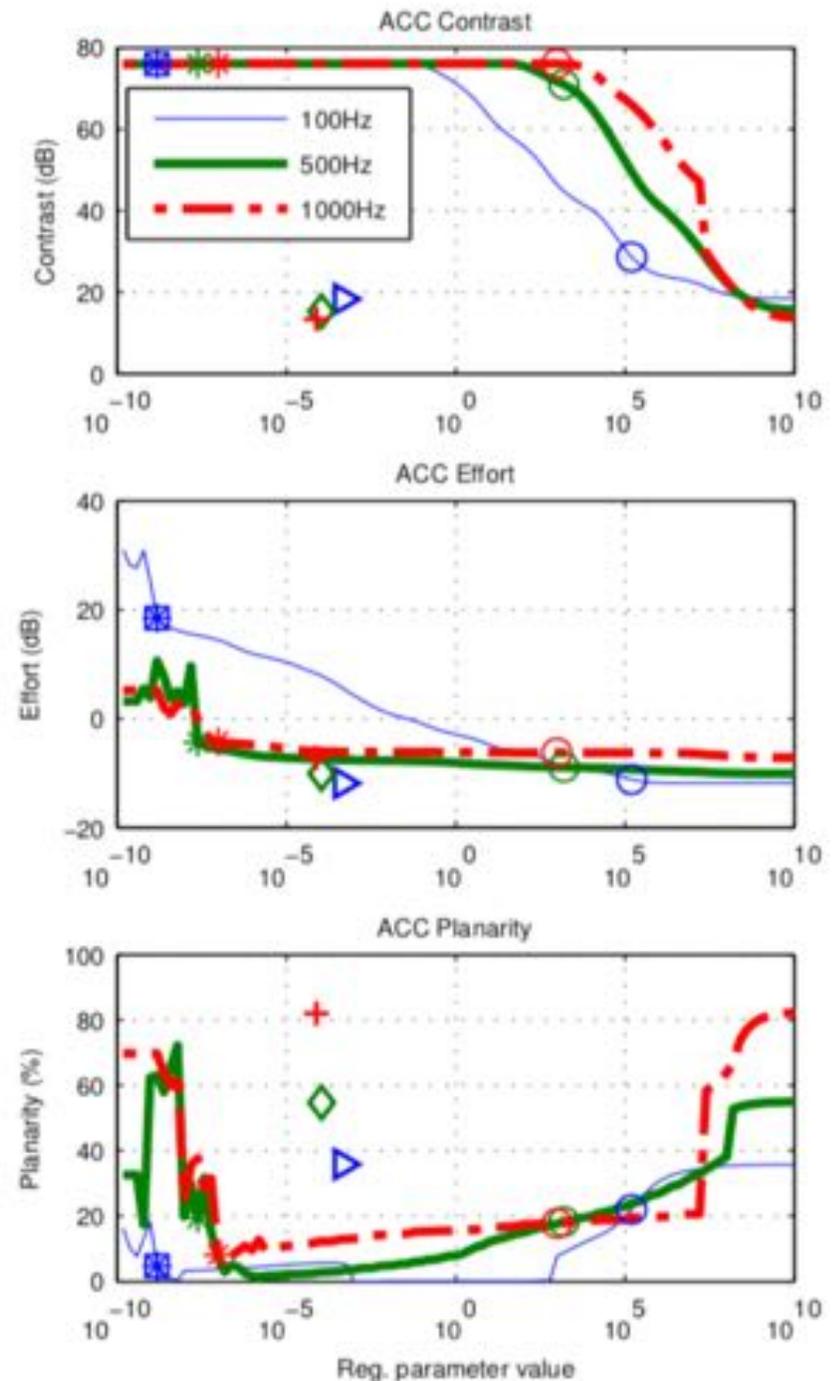


Fundamental Relationships under Non-Ideal Conditions: Robustness and Regularization



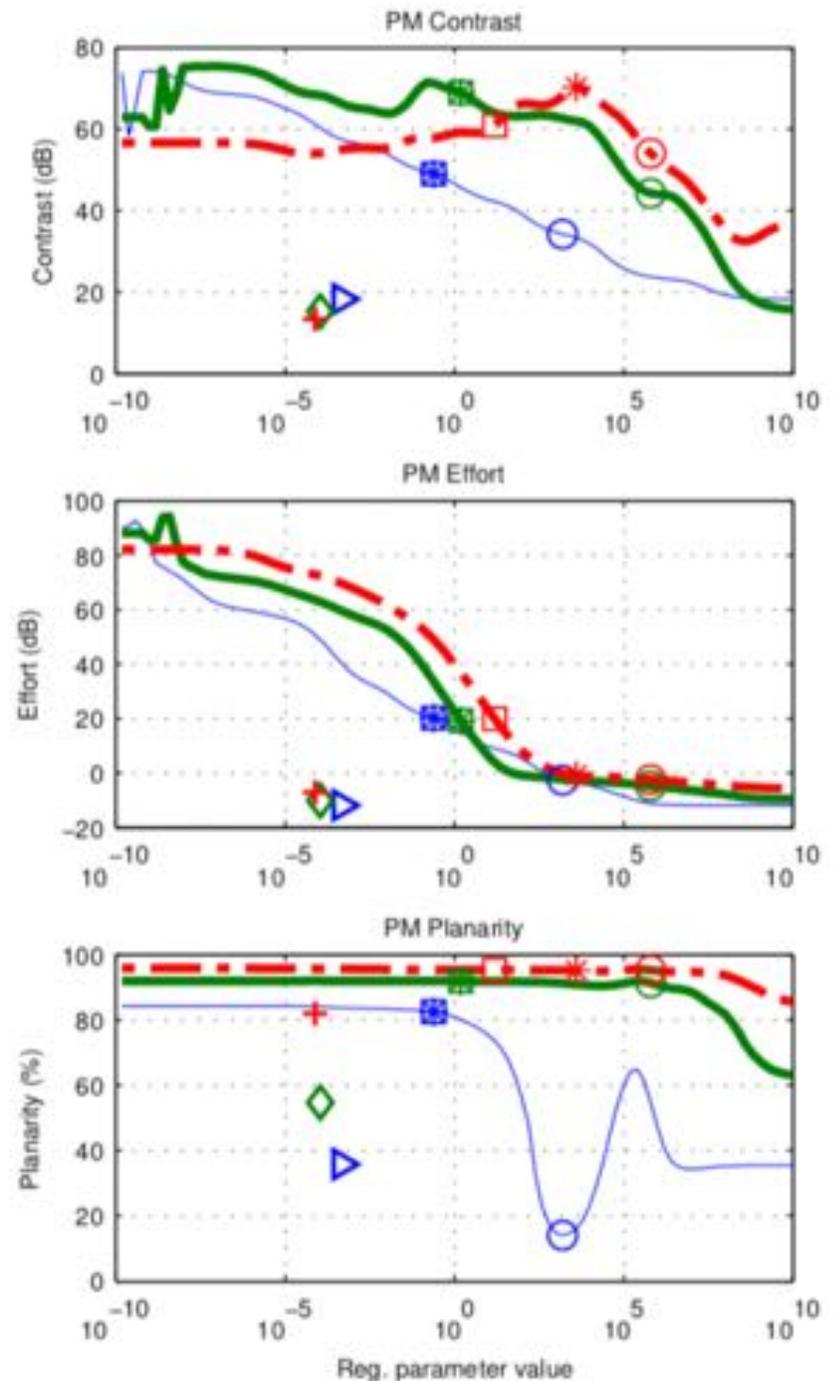
Regularization: Fundamentals

- Regularization parameter vs performance (ideal conditions)
- Acoustic Contrast Control:
 - **More regularization = lower effort**
 - Contrast and planarity fairly robust unless over-regularized



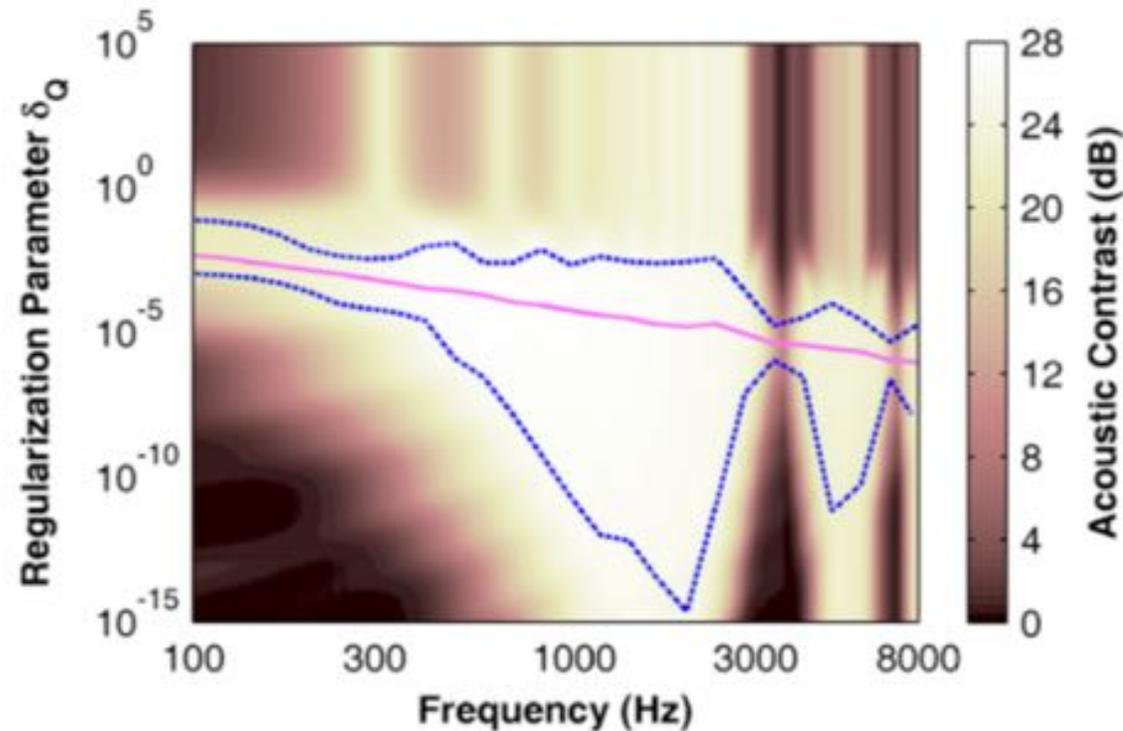
Regularization: Fundamentals

- Regularization parameter vs performance (ideal conditions)
- Pressure Matching:
 - **More regularization = lower effort**
 - Planarity fairly robust unless over-regularized
 - Some local optima in contrast



Regularization: Non-ideal Conditions

- Design challenge: choose optimal regularization parameter
- Some frequencies are more sensitive than others



Regularization: Non-ideal Conditions

- Design challenge: choose optimal regularization parameter

	Approach	Example references
Matrix Perspective	Choose diagonal loading regularization parameter based on condition number / singular values of matrix	Shin, M., et al., 2014. "Controlled sound field with a dual layer loudspeaker array". <i>J. Sound Vib.</i>
Array Effort Perspective	Choose diagonal loading regularization parameter to meet a maximum control effort constraint	Elliott, S.J., et al., 2012. "Robustness and regularization of personal audio systems". <i>TASLP</i> . Coleman, P., et al., 2014. "Acoustic contrast, planarity and robustness of sound zone methods using a circular loudspeaker array". <i>JASA</i> .
Error Modelling Perspective	Choose regularization parameters (matrix) based on predicted system errors	Doclo, S. & Moonen, M., 2003. "Design of broadband beamformers robust against gain and phase errors in the microphone array characteristics". <i>TSP</i> . Zhu, Q, et al., 2017, "Robust reproduction of sound zones with local sound orientation". <i>JASA-EL</i> . Zhu, Q, et al., 2017, "Robust Acoustic Contrast Control with Reduced In-situ Measurement by Acoustic Modelling". <i>JAES</i> .

Regularization: Non-ideal Conditions

- Design challenge: choose optimal regularization parameter

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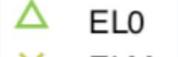
Regularization: Non-ideal Conditions

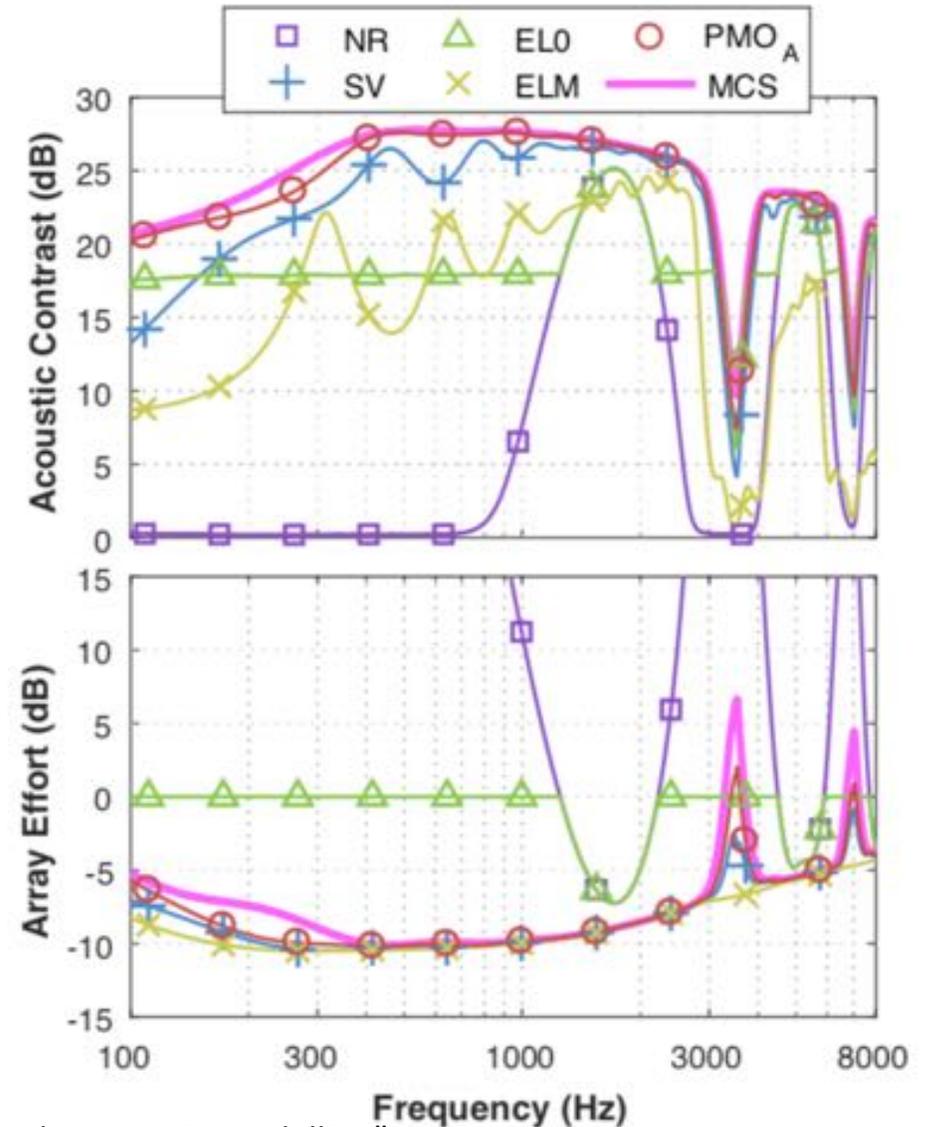
- Design challenge: choose optimal regularization parameter

	Approach	Example references
Matrix Perspective	Choose diagonal loading regularization parameter based on condition number / singular values of matrix	Shin, M., et al., 2014. "Controlled sound field with a dual layer loudspeaker array". <i>J. Sound Vib.</i>
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Regularization: Non-ideal Conditions

- Optimal regularization parameter?

	Approach
Matrix Perspective  SV	Choose diagonal loading regularization parameter based on condition number / singular values of matrix
Array Effort Perspective  ELO  ELM	Choose diagonal loading regularization parameter to meet a maximum control effort constraint
Error Modelling Perspective  PMO _A	Choose regularization parameters (matrix) based on predicted system errors





Regularization: Summary

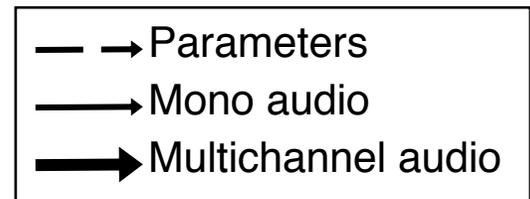
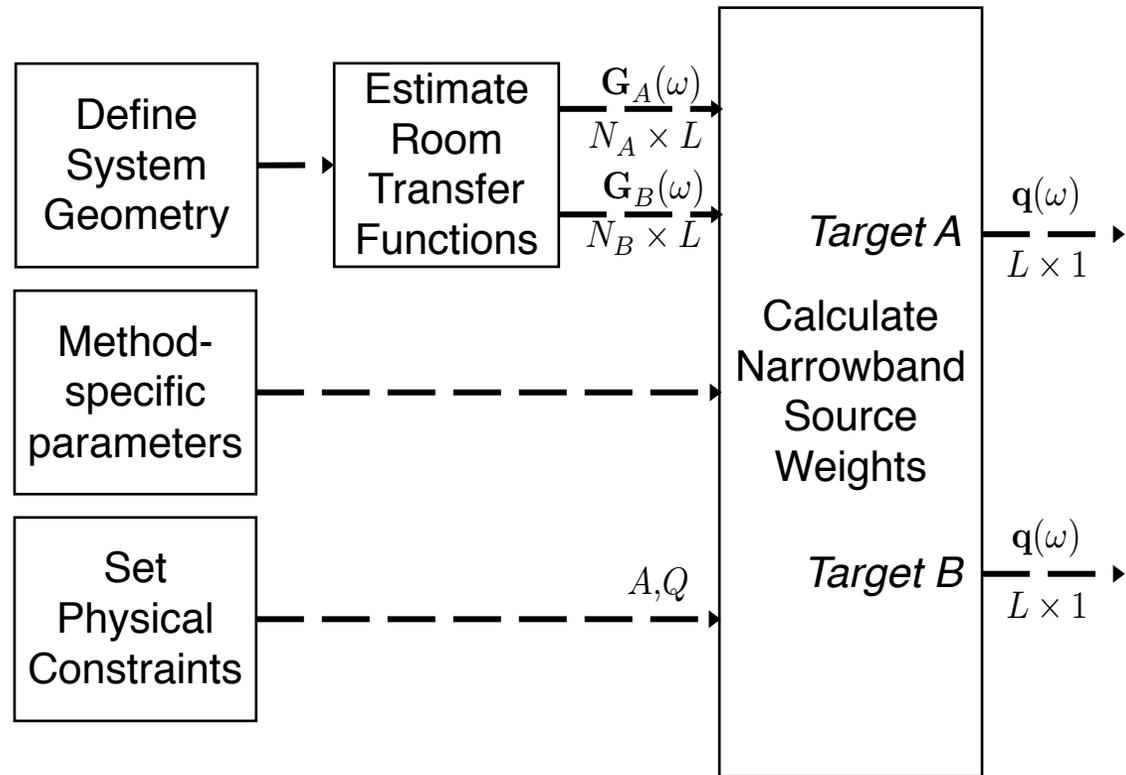
- Under real-world conditions, regularization is critical:
 - Too little regularization: not robust
 - Too much regularization: no contrast
- Three representative ways to find parameter(s) introduced:
 - Look at the matrix to be inverted
 - Fix the array effort
 - Consider the predicted kind/magnitude of errors in the whole system

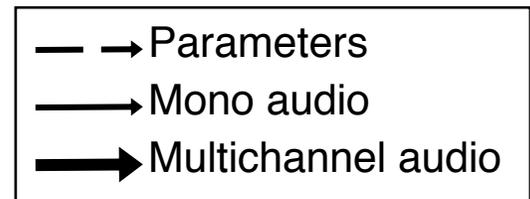
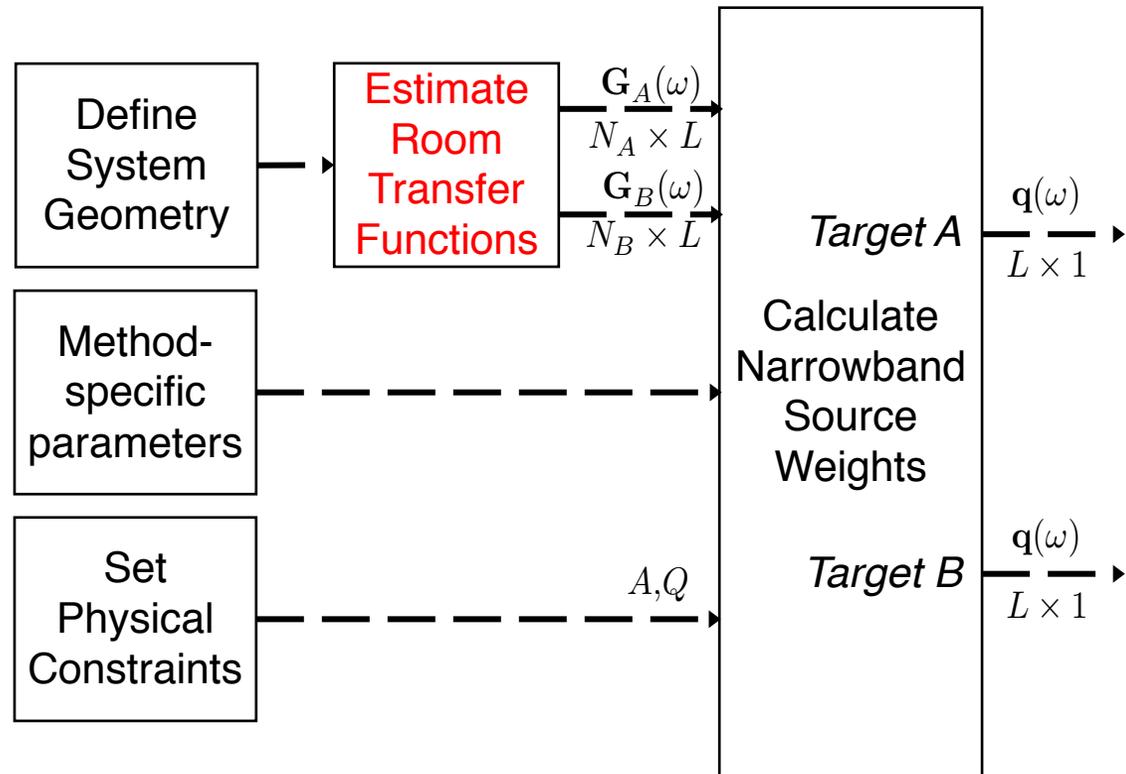


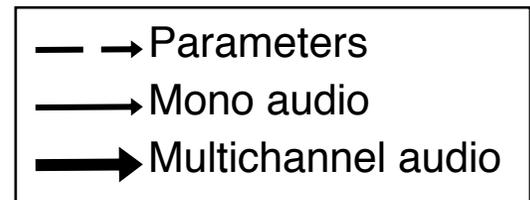
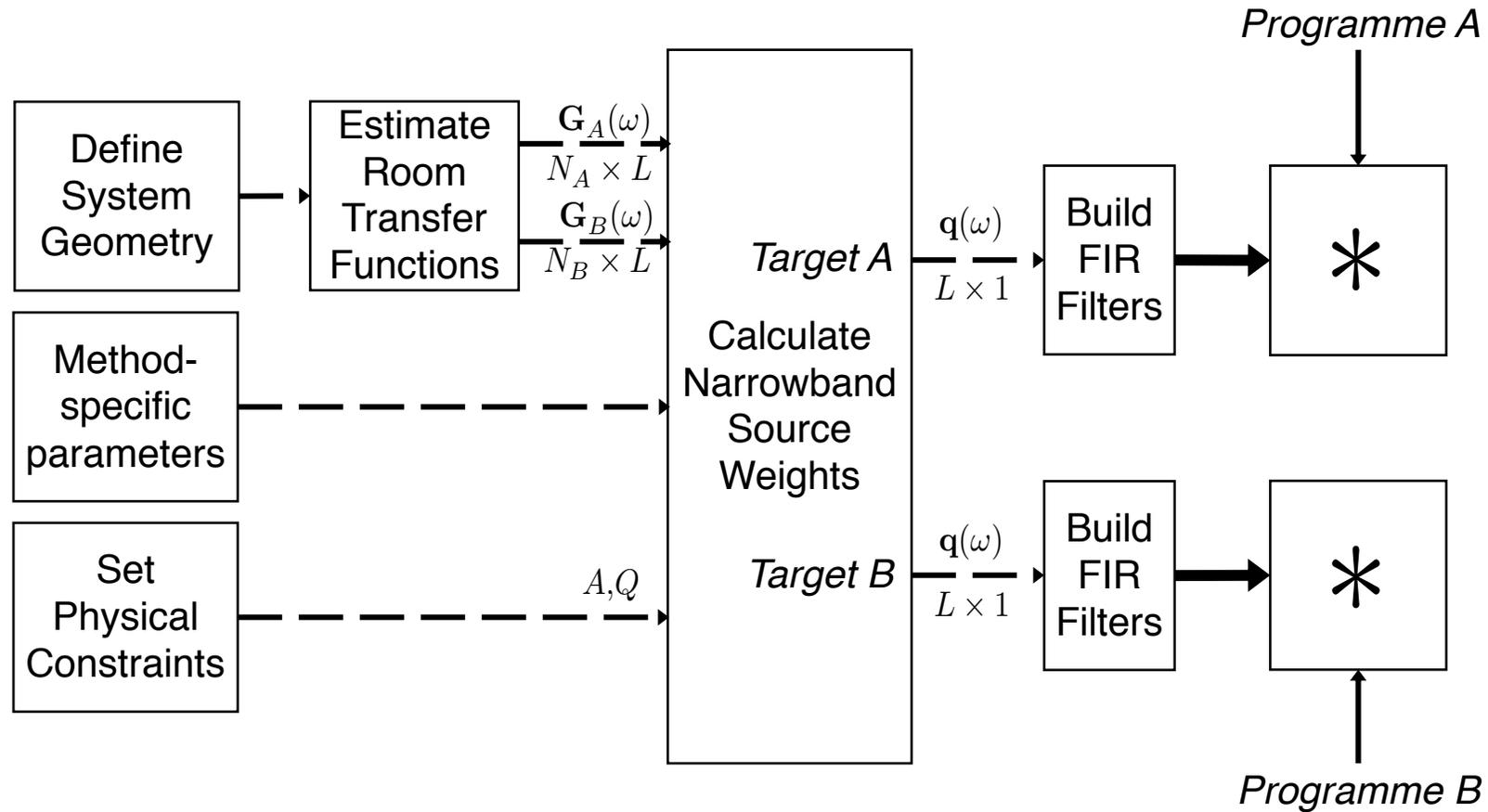


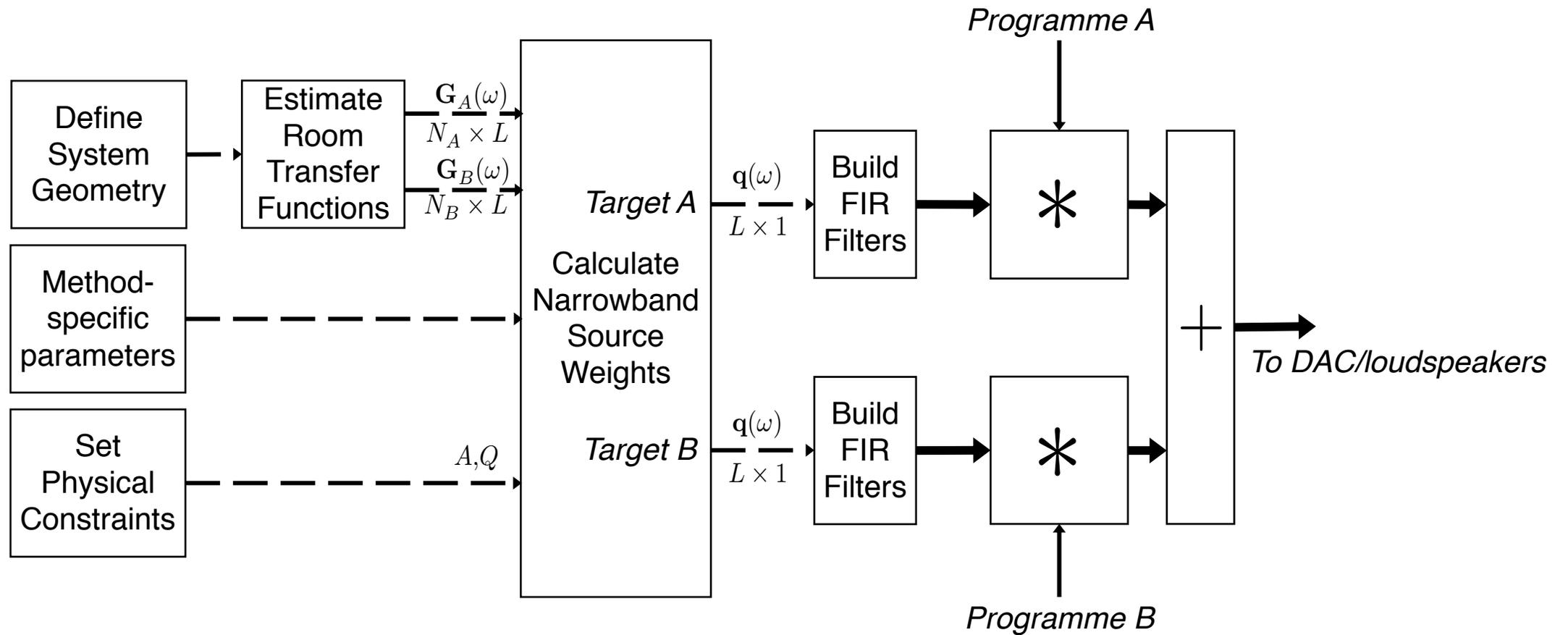
Making Sound Zone Filters



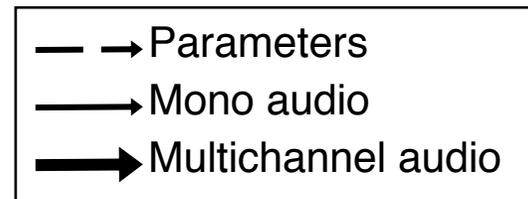
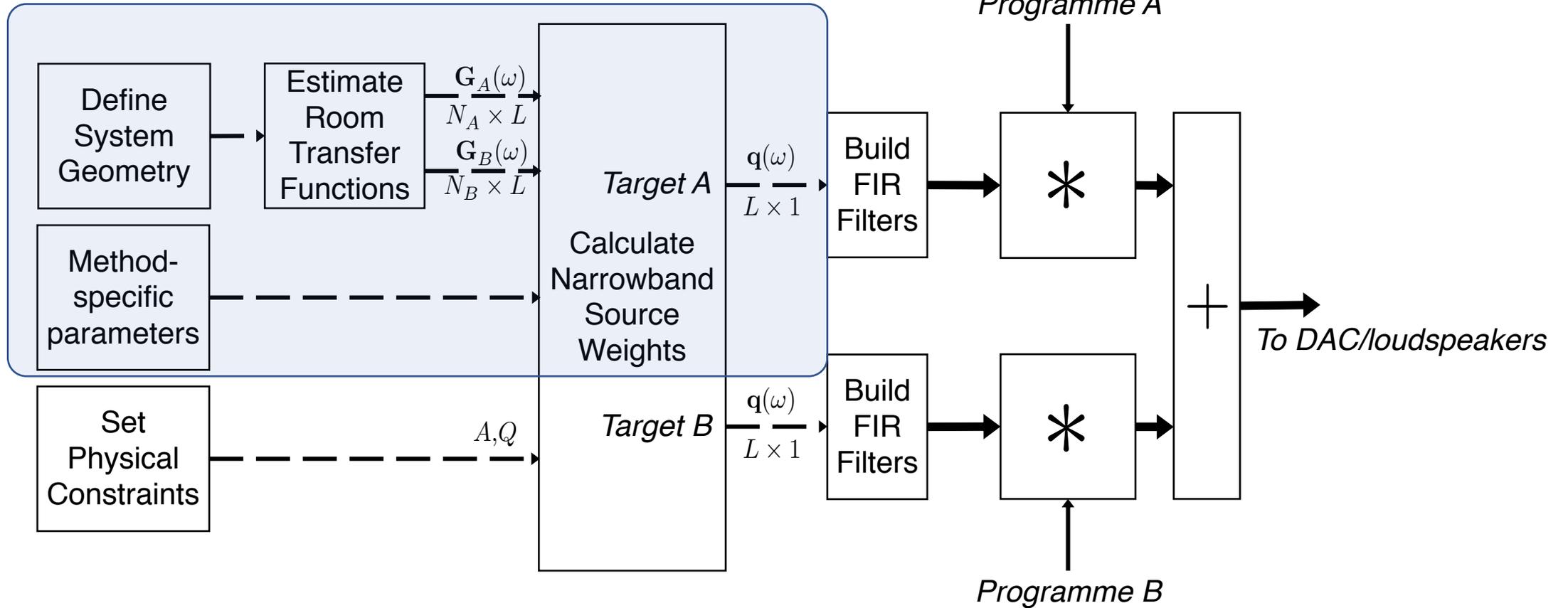








Demo





Demo: PyZones Toolbox for Sound Zone Filter Design

<https://github.com/loSR-Surrey/PyZones>



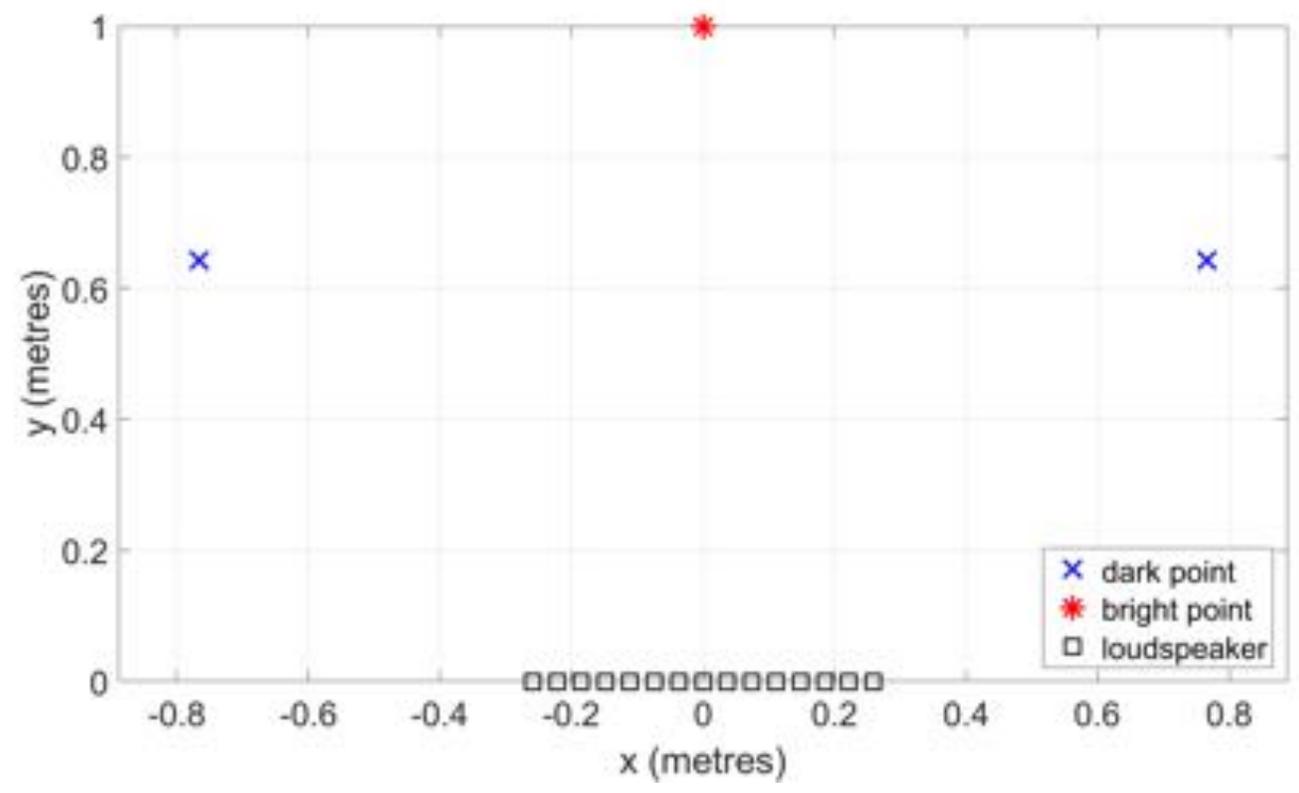


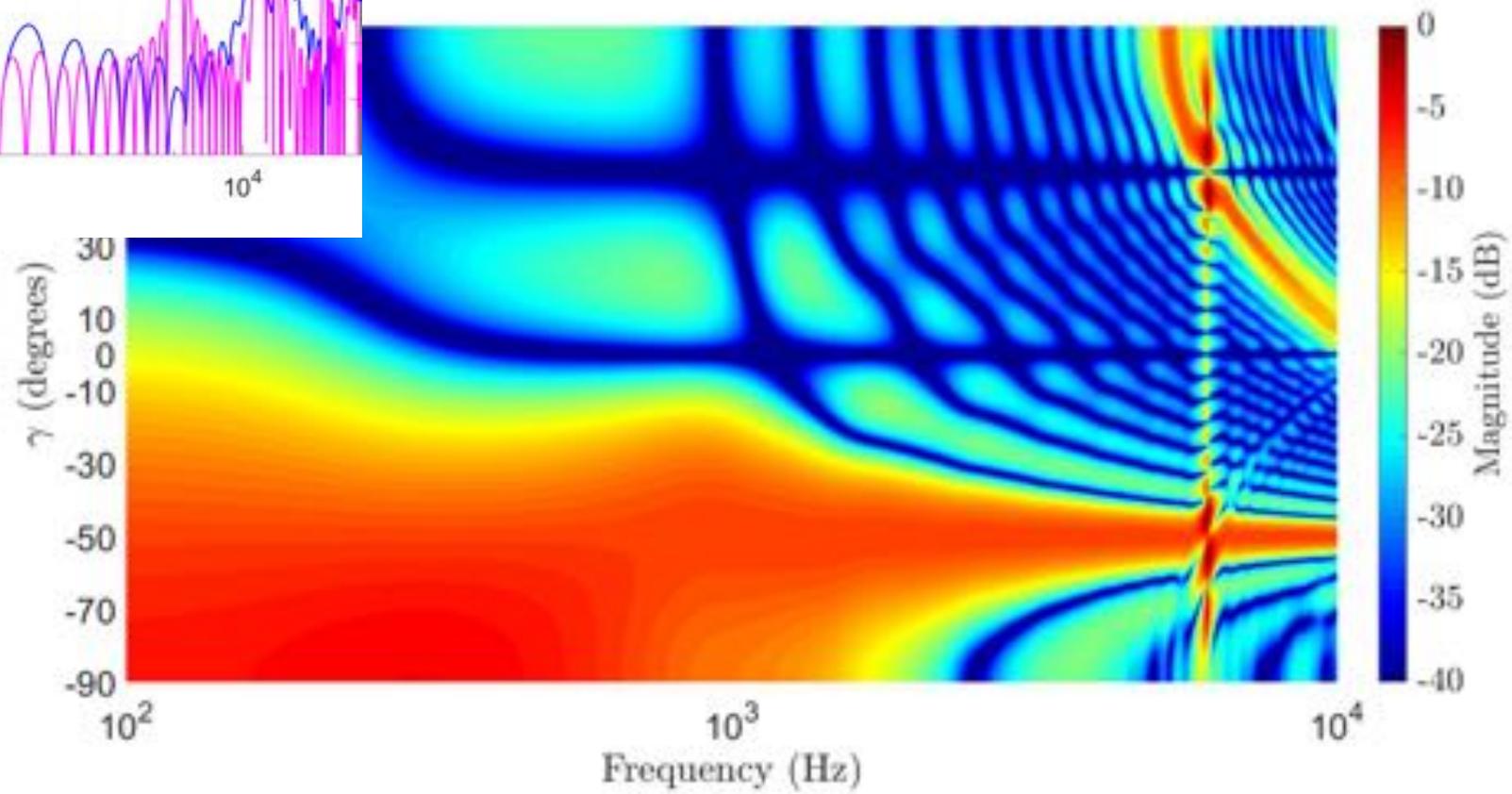
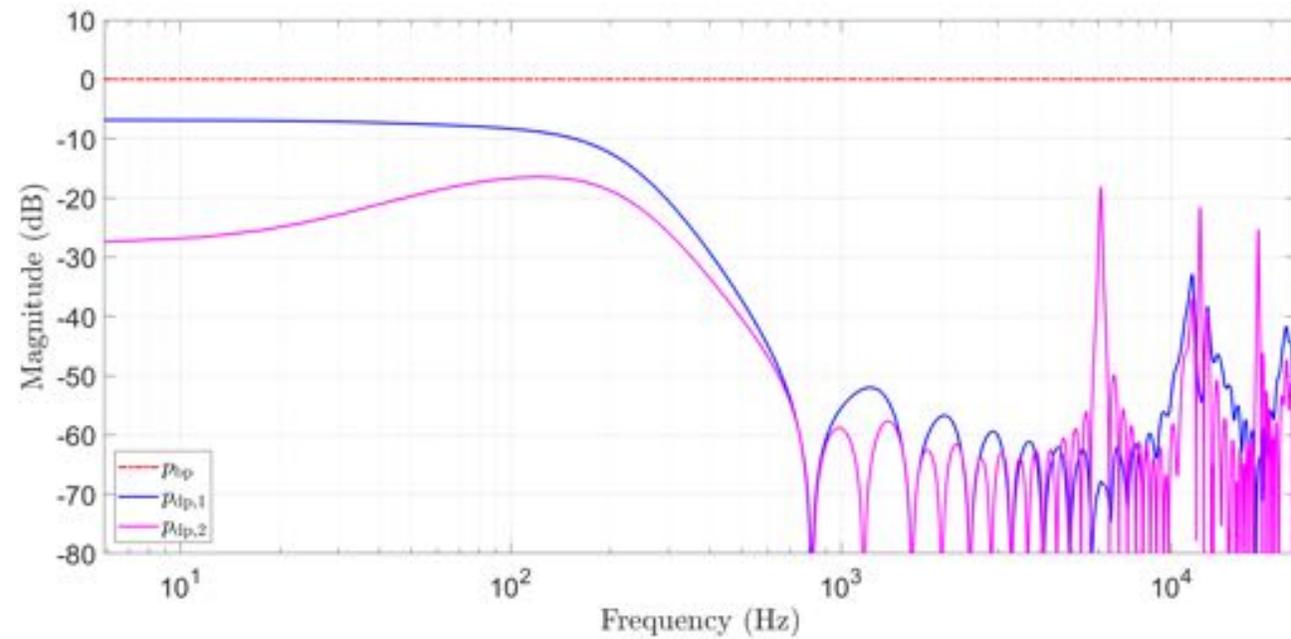
Demo: multi-zone audio delivery with a 15 channel linear loudspeaker array

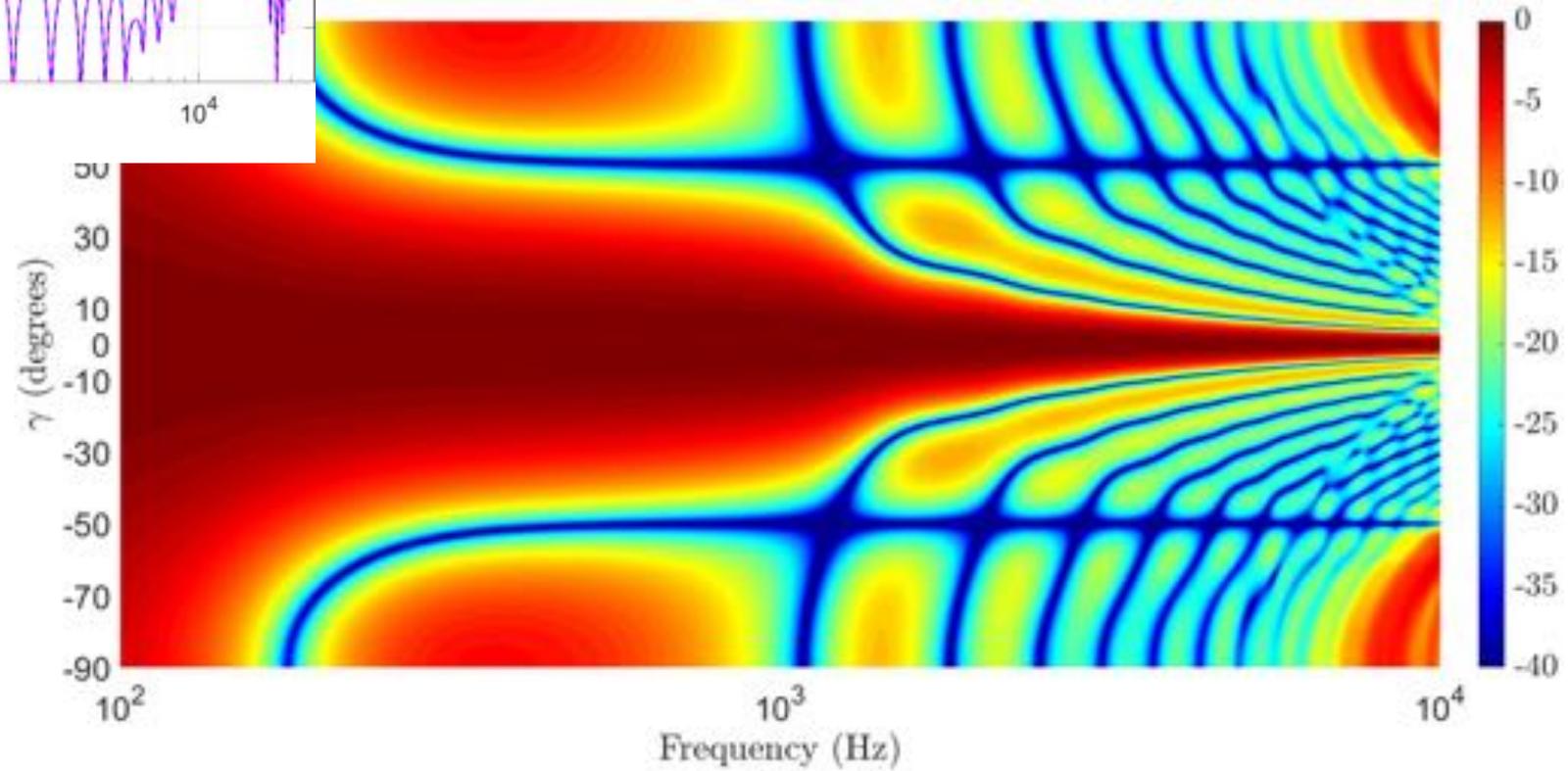
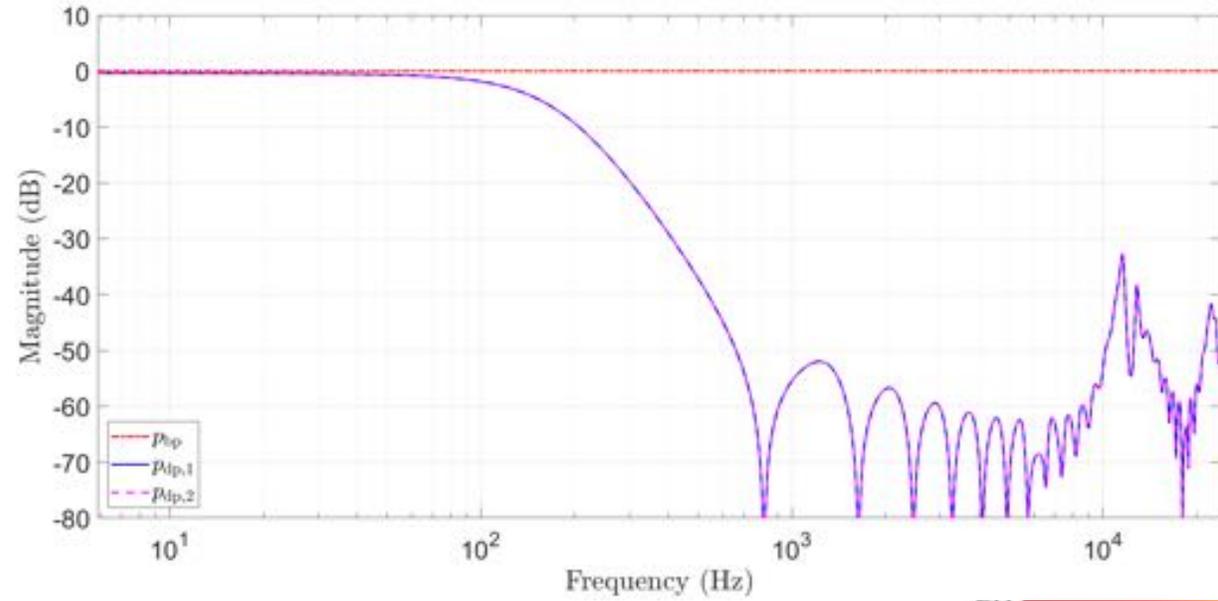
With contribution by Eric Hamdan



System geometry









Perceptual Measures of Sound Zone Performance





Does it sound any good then?

- Two main factors contribute:
 - Interference suppression
 - Target sound quality
- Interference from the other zone is affected by:
 - Acoustic contrast (cost function design)
 - Relative loudness / spectra of the programme content (not accounted for in design)
- Sound quality in the target zone is affected by:
 - Flatness of frequency response (constraint, cost function design)
 - Ringing in filters (regularization, cost function design)
 - Bright zone sound propagation (cost function design)





Interference

- Understand the perceptual effects of audio interference
- First need to describe interference in a perceptually relevant way
- Attribute elicitation process led to **distraction** as the most perceptually relevant descriptor

Francombe, J. et al., 2014. "Elicitation of attributes for the evaluation of audio-on-audio interference". *JASA*.



Modelling Distraction

- A regression model to predict distraction was developed at Surrey
- Main physical correlates: overall loudness, **loudness ratio**, interference rejection, and frequency content of the interferer
- Loudness ratio better correlation than (unweighted) target to interferer ratio / acoustic contrast
- Recent work has developed and validated a real-time implementation

Francombe, J., 2014. *Perceptual evaluation of audio-on-audio interference in a personal sound zone system* (Doctoral dissertation, University of Surrey).

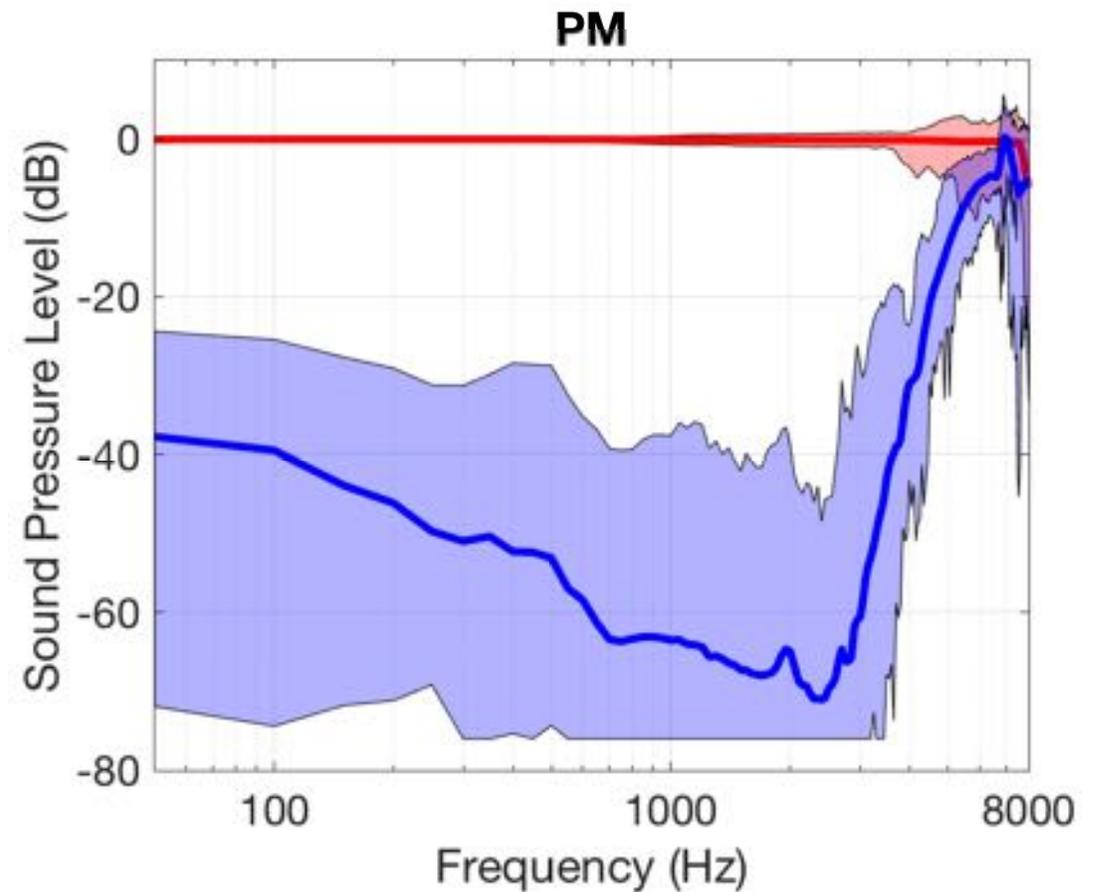
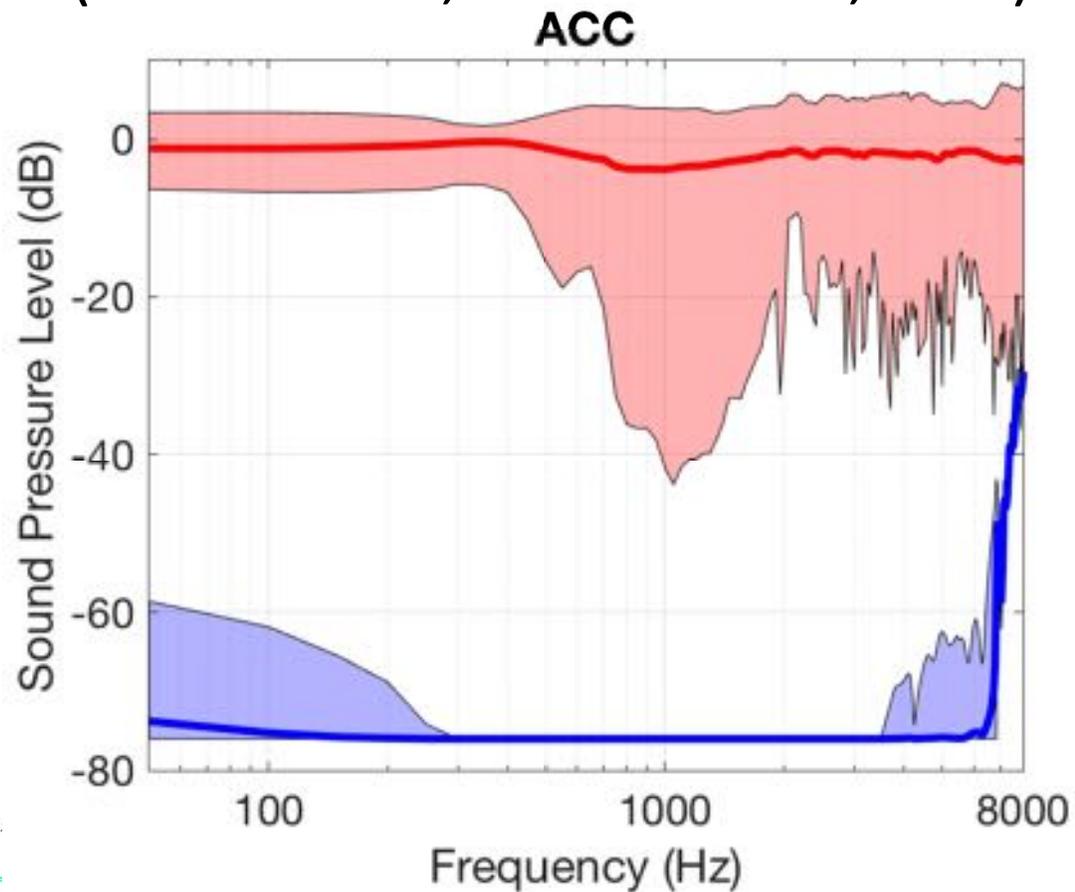
Francombe, J. et al., 2015. "A model of distraction in an audio-on-audio interference situation with music program material". *JAES*.

Rämö, J., et al., 2017. "Real-time perceptual model for distraction in interfering audio-on-audio scenarios". *IEEE Sig. Proc. Letters*.

Rämö, J., et al., 2018. "Validating a real-time perceptual model predicting distraction caused by audio-on-audio interference". *JASA*.

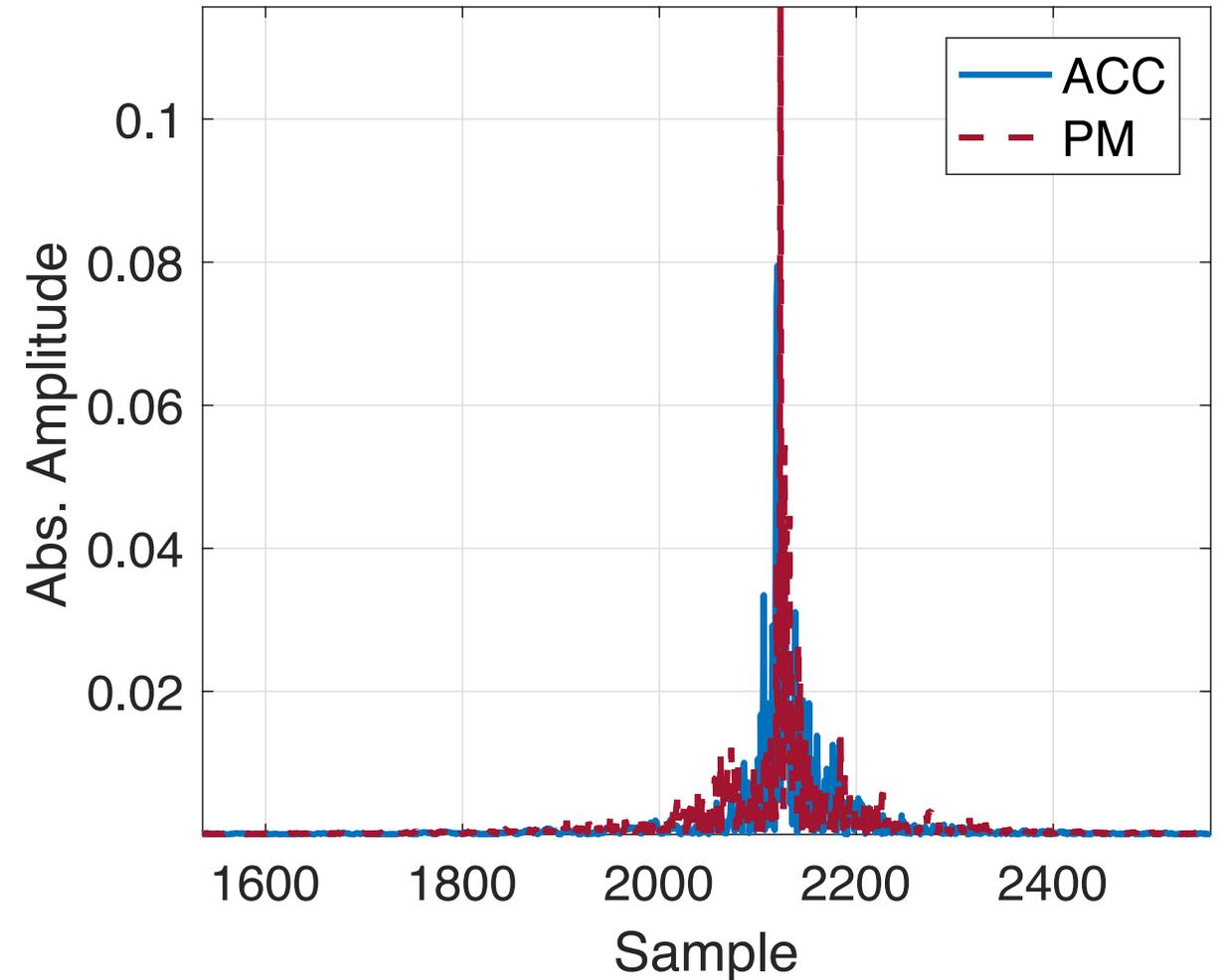
Sound Quality

- Frequency response across zone microphones (solid: mean; shaded: min, max)



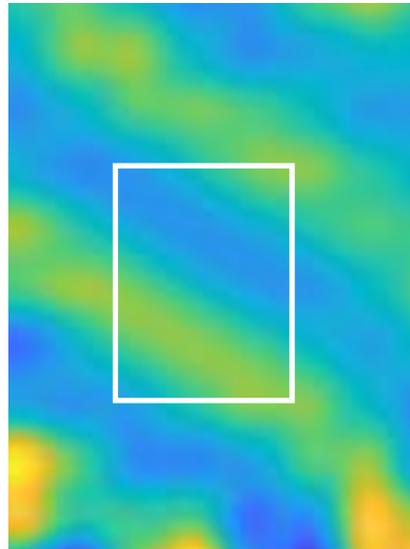
Sound Quality

- Ringing in filters
- Illustrated with simulated time response at central microphone in zone A

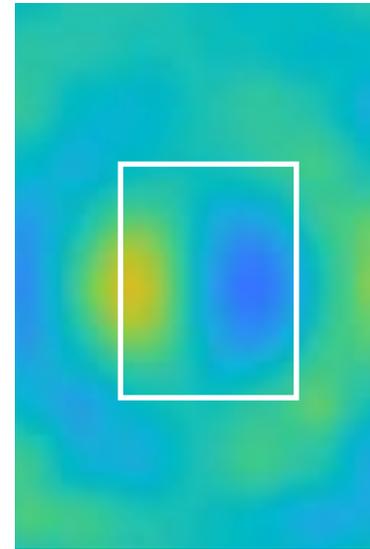


Sound Quality

- Spatial properties of the bright zone



PM: plane wave



ACC: ?

Planarity Control

- PC Cost function

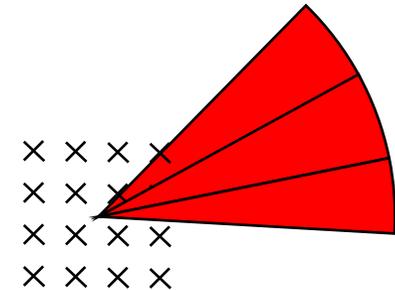
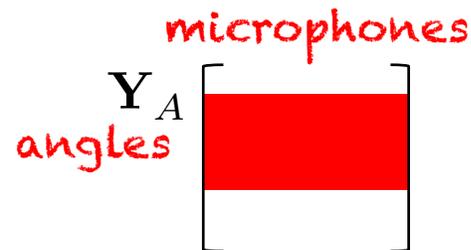
constraint on bright zone pressure is now spatially weighted

$$\text{Min. } J_{PC} = \mathbf{p}_B^H \mathbf{p}_B + \mu(\mathbf{p}_A^H \mathbf{Y}_A^H \mathbf{\Gamma} \mathbf{Y}_A \mathbf{p}_A - A) + \lambda(\mathbf{q}^H \mathbf{q} - Q)$$

pressure in the dark zone (like ACC)

constraint on effort

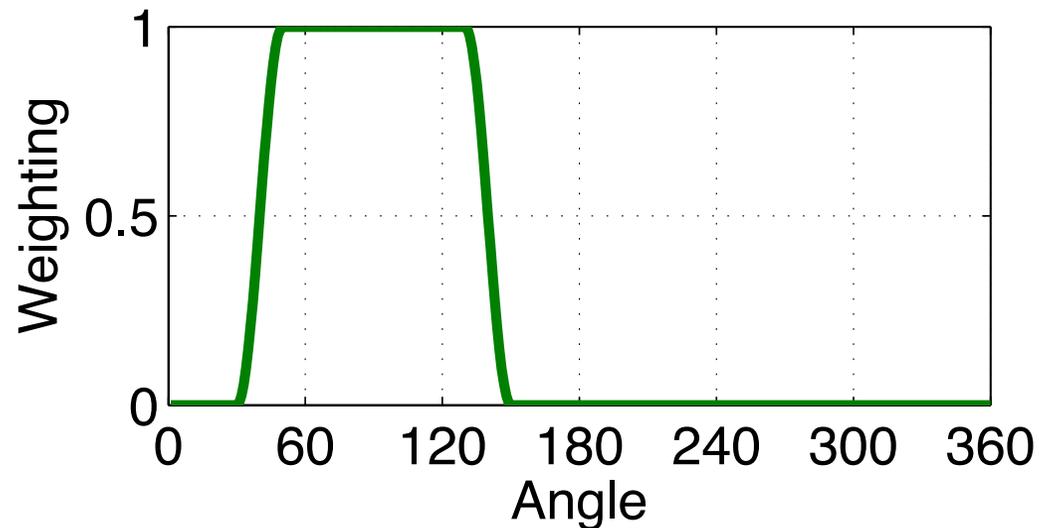
- Spatial steering matrix
 - Populated by superdirective beamforming



- Specify pass range of directions by means of Gamma term $\mathbf{\Gamma} = \text{diag}[\gamma_1, \gamma_2, \dots, \gamma_I]$

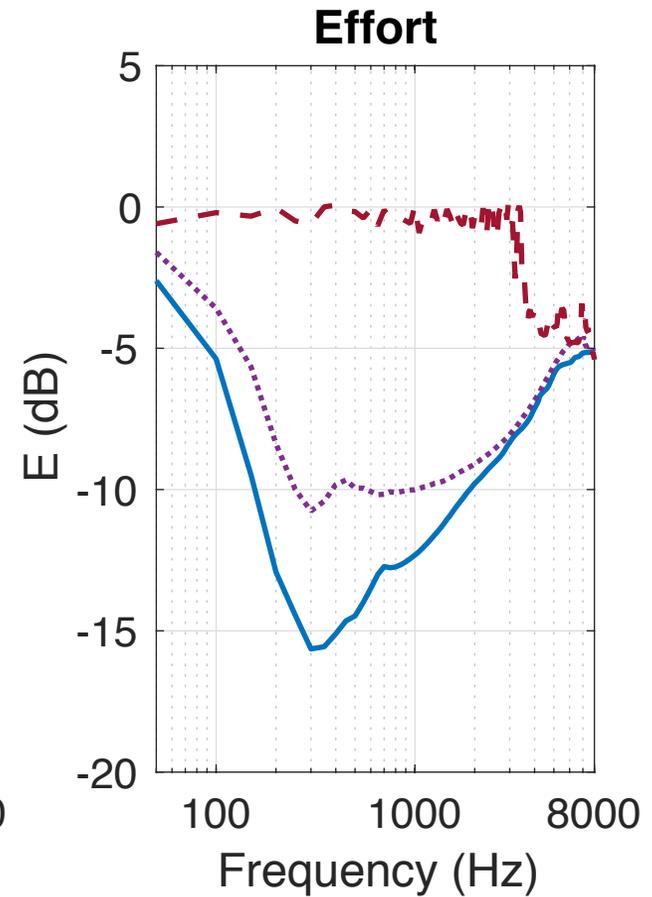
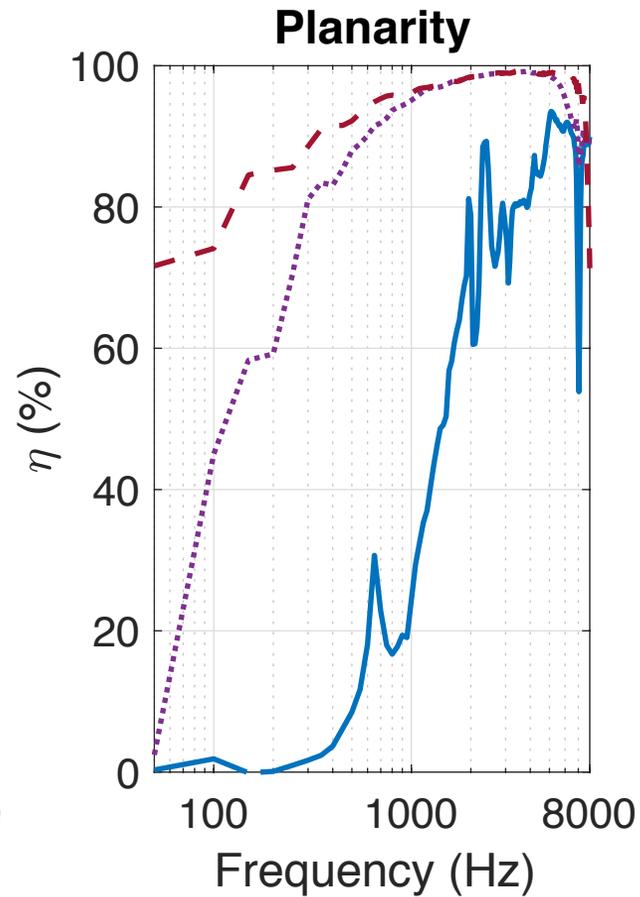
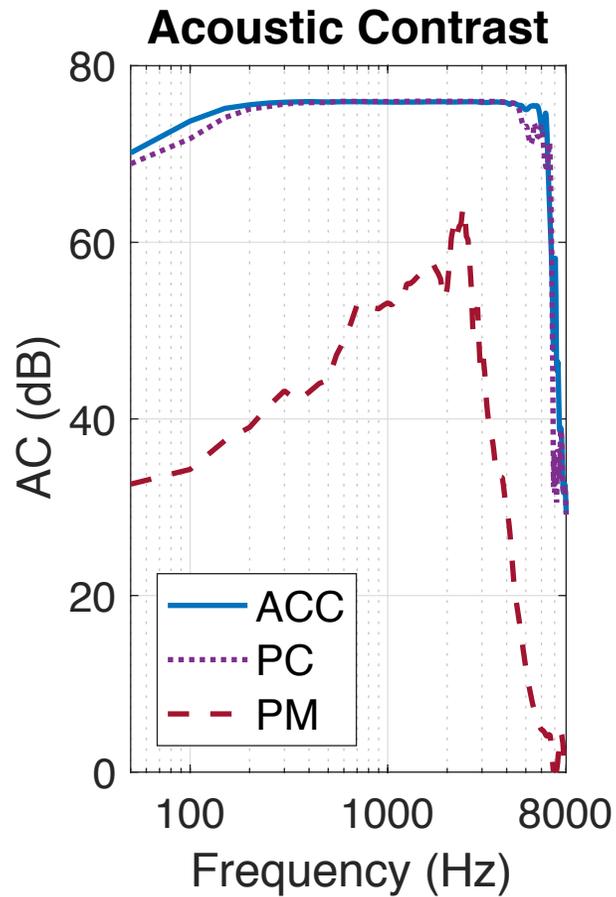
Angular pass range design

- Direction of impinging energy at bright zone
 - ACC not constrained: can leads to spatial problems
 - PM narrowly constrained
 - PC constrained to a range of angles



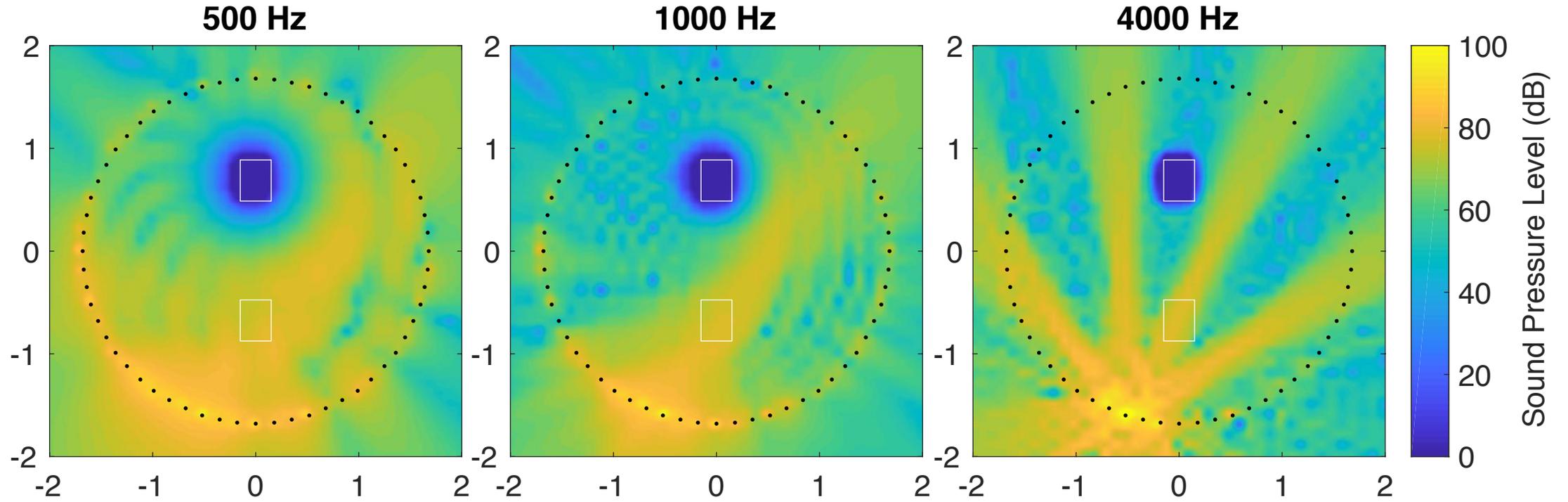
Planarity Control

- Performance over frequency



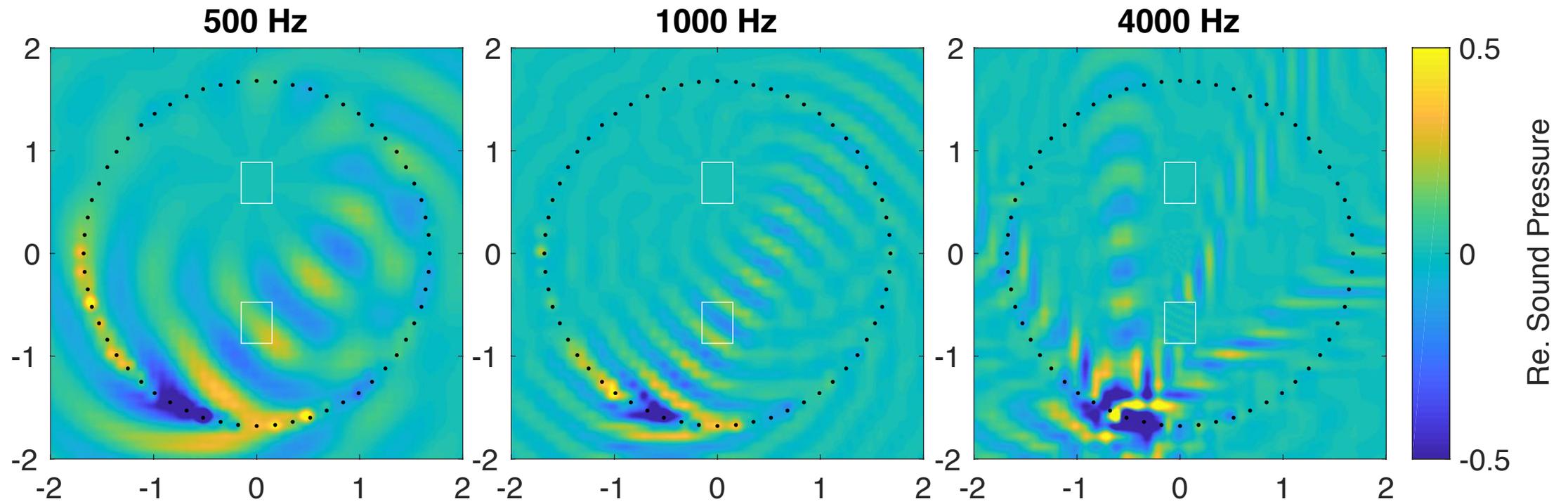
Planarity Control

- Sound field: sound pressure level



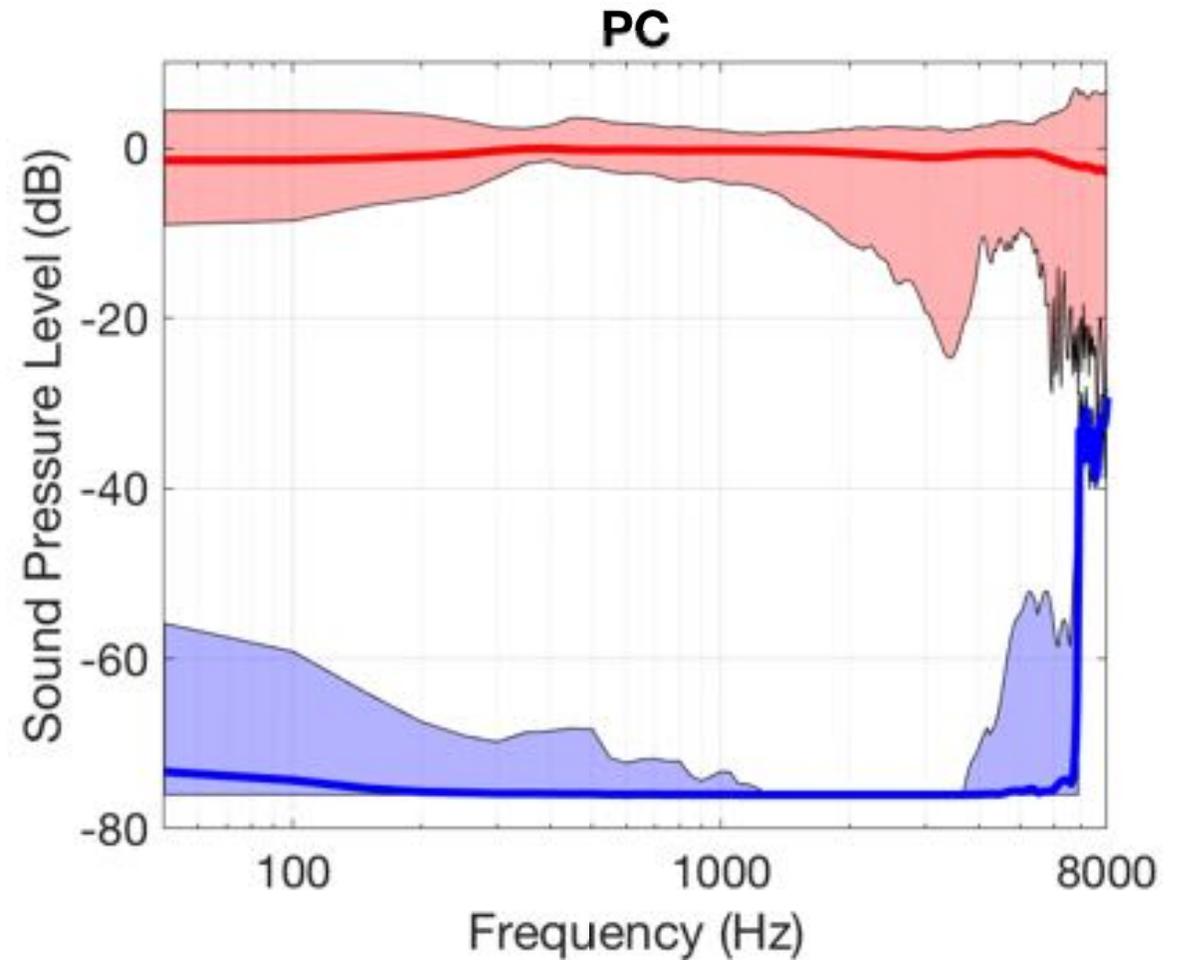
Planarity Control

- Sound field: real part of the sound pressure



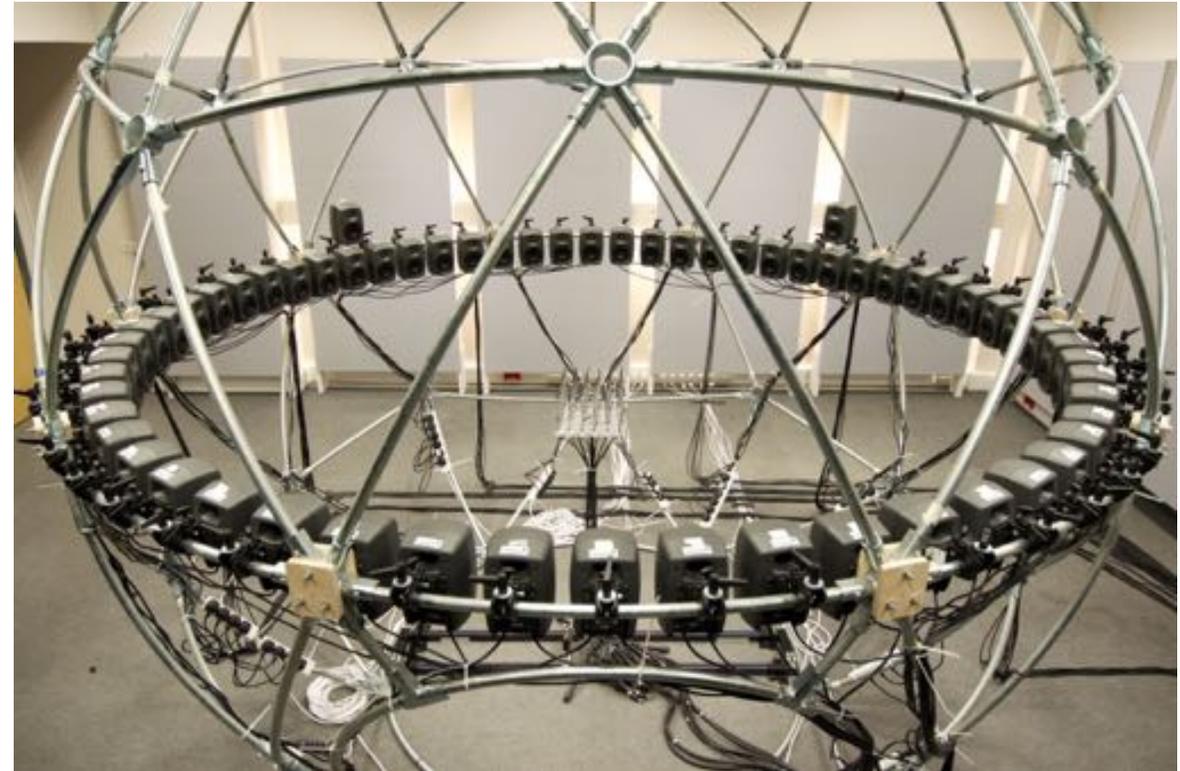
Planarity Control

- Frequency response across zone microphones
(solid: mean; shaded: min, max)



Perceptual Evaluation

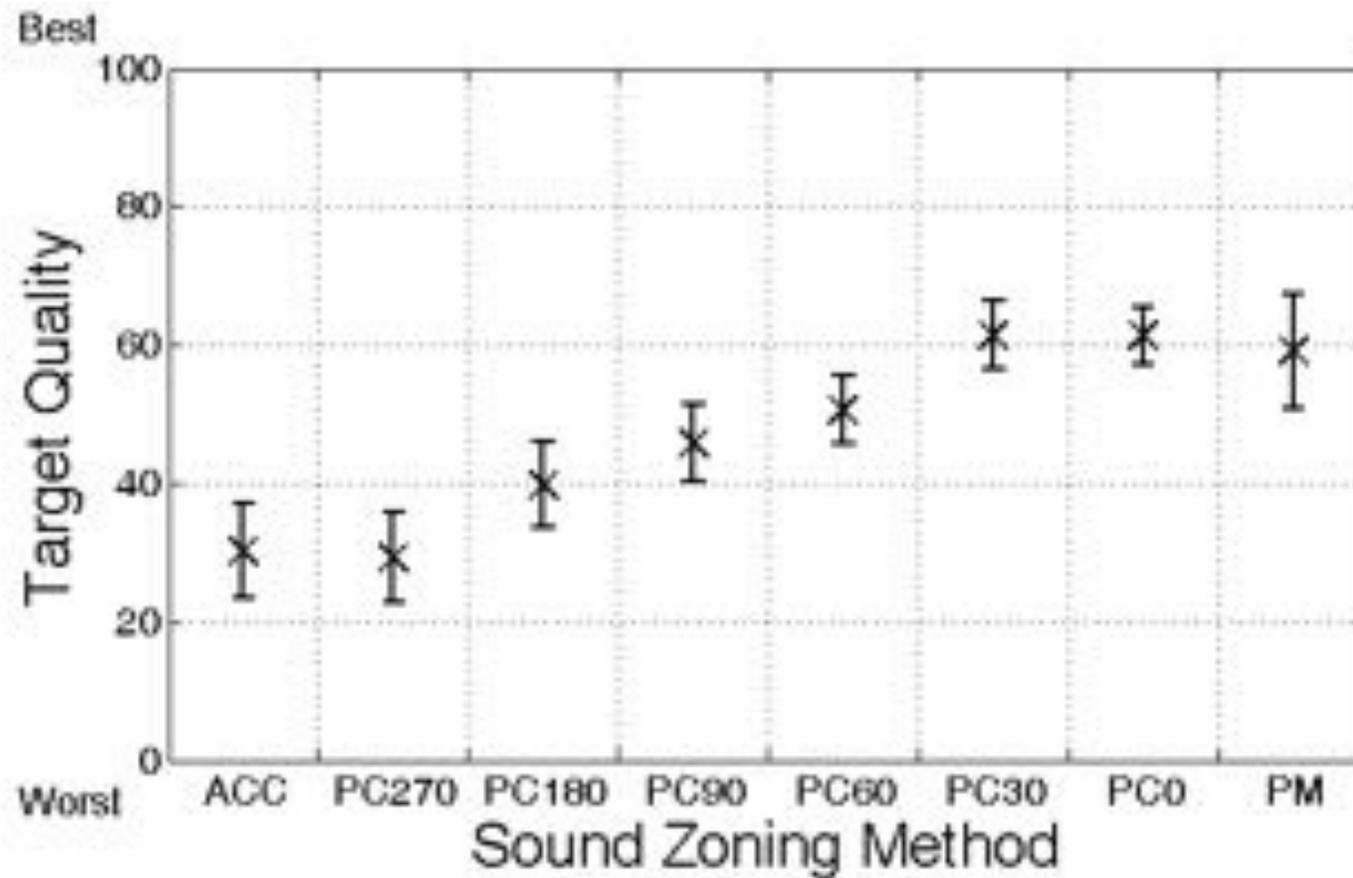
- We ran listening tests to compare ACC, PC and PM
 - Including some variants of PC
- Implemented in the Surrey Sound Sphere (nothing simulated!)
- Listeners separately rated:
 - Target quality
 - Distraction
 - Overall quality



Baykaner, K. et al., 2015. "The relationship between target quality and interference in sound zones". *JAES*.

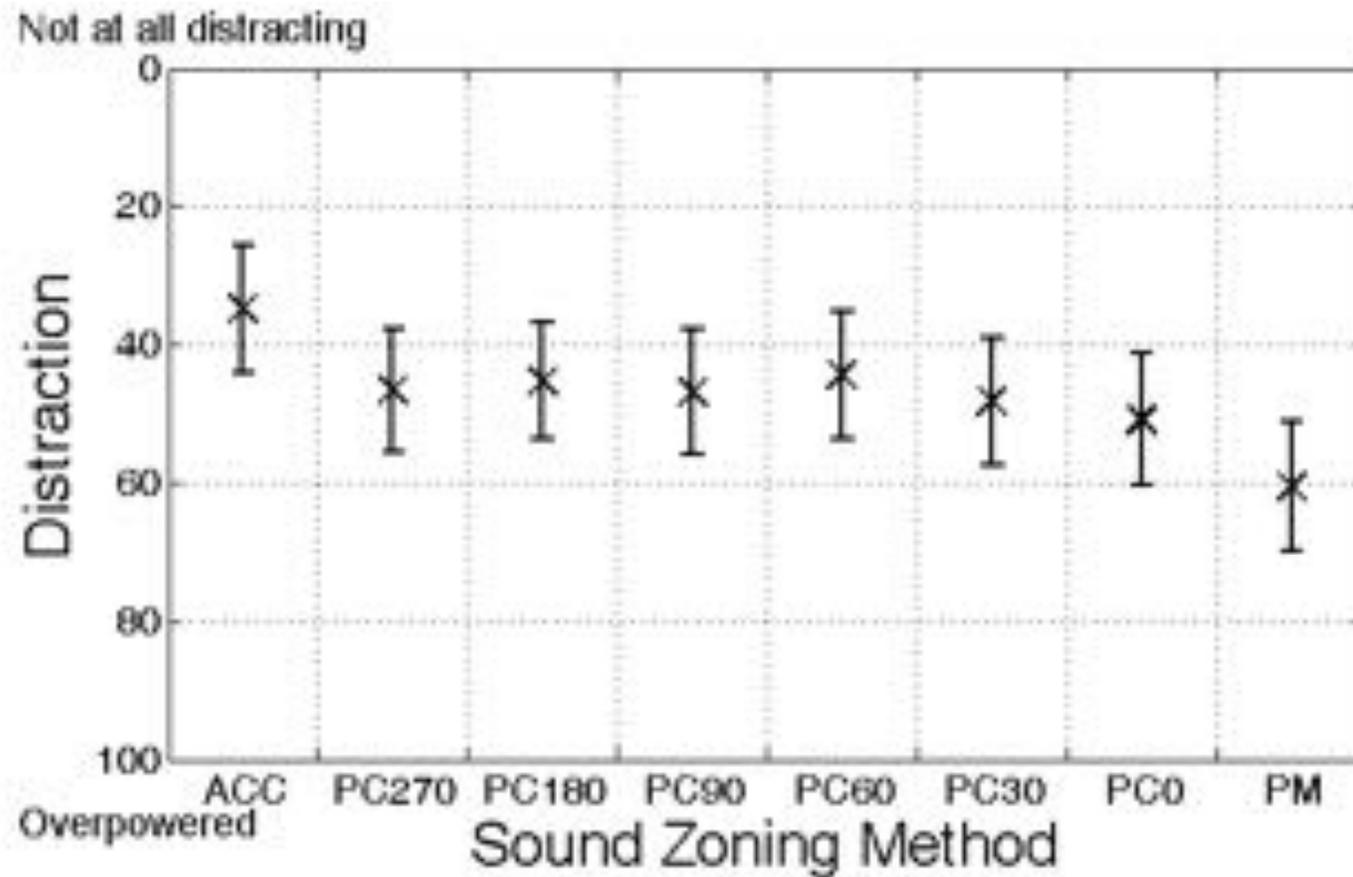
Perceptual Evaluation

- Target Quality



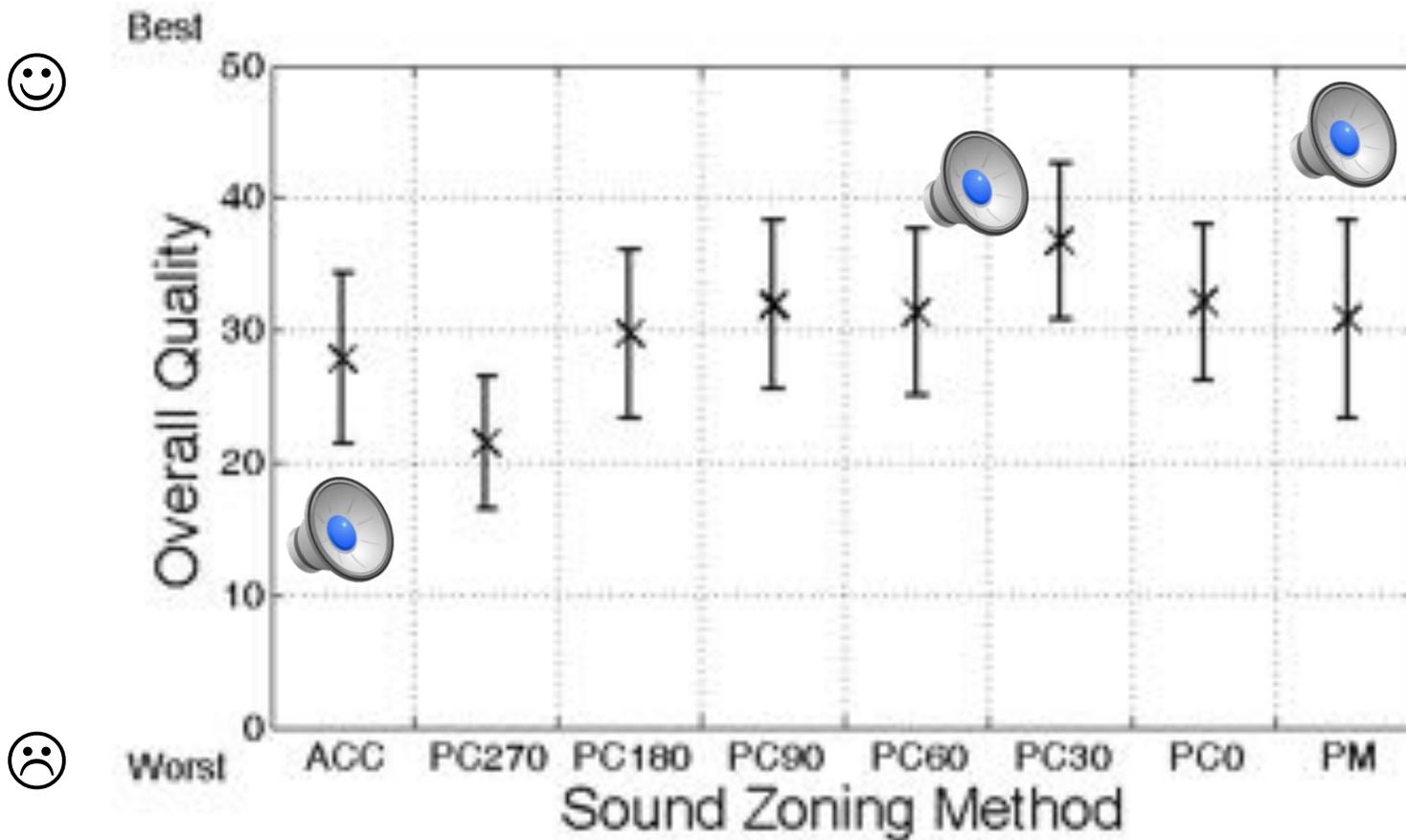
Perceptual Evaluation

- Distraction



Perceptual Evaluation

- Overall Quality





Perceptual Evaluation

- Conclusions:
 - In choosing a cost function, there is a trade-off between target quality and interference
 - The spatial energy distribution and phase in the bright zone was the main difference between methods as tested
 - Planarity control, with a soft DOA constraint, gives a good trade-off between the characteristics of the baseline methods (ACC and PM)
 - Other aspects (frequency response, ringing in filters) have not been explored to date in the context of sound zones





Alternative approaches to personal sound zones



Spherical Harmonic Representation of Sound Fields

- A sound field that is solution the homogeneous Helmholtz equation can be represented as

Basis functions

$$p(r, \theta, \varphi, \omega) = \sum_{n=0}^{\infty} \sum_{m=-n}^n B_{nm}(\omega) i^n j_n(kr) Y_n^m(\theta, \varphi)$$

Sound field

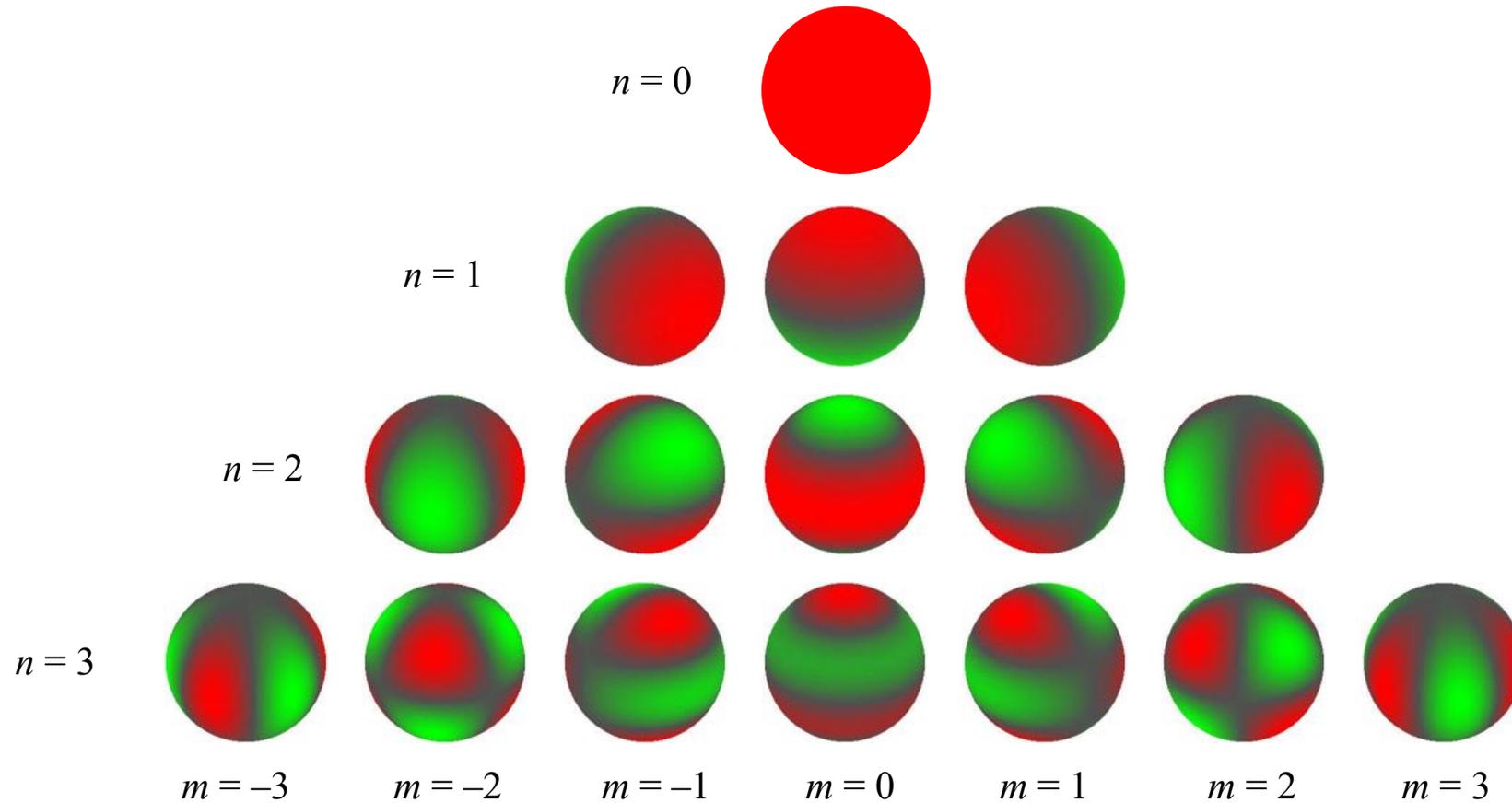
Series coefficients

Spherical Bessel functions

Spherical harmonics

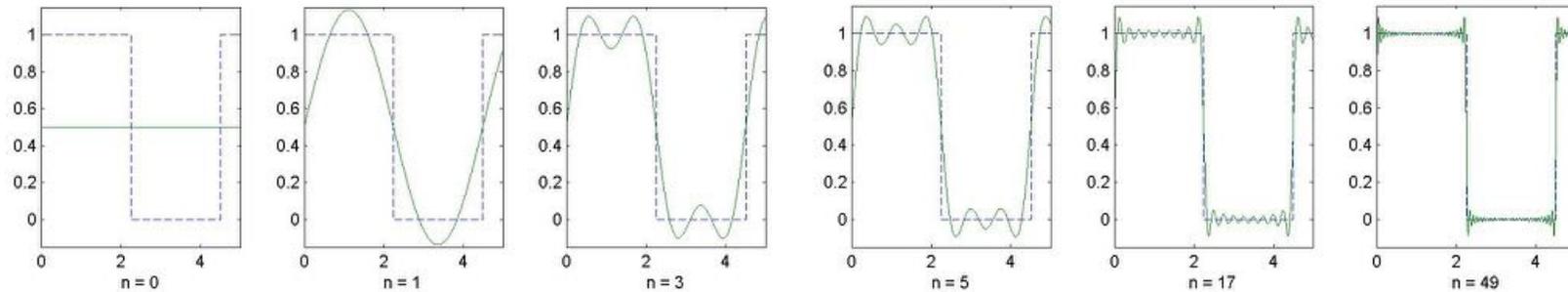


The Spherical Harmonics

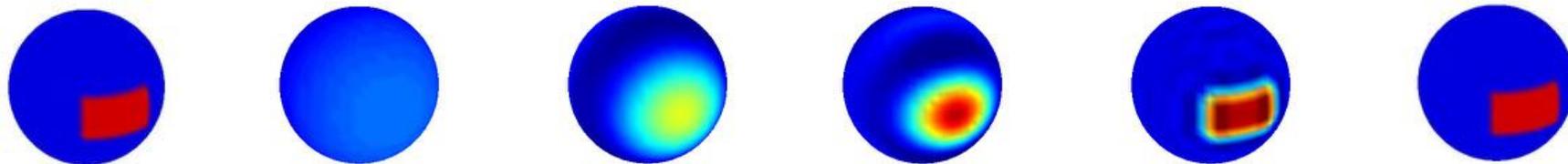


Spherical Harmonics and Fourier Series

- Fourier series



- Spherical harmonic expansion (generalised Fourier series)



Target

n=1

n=3

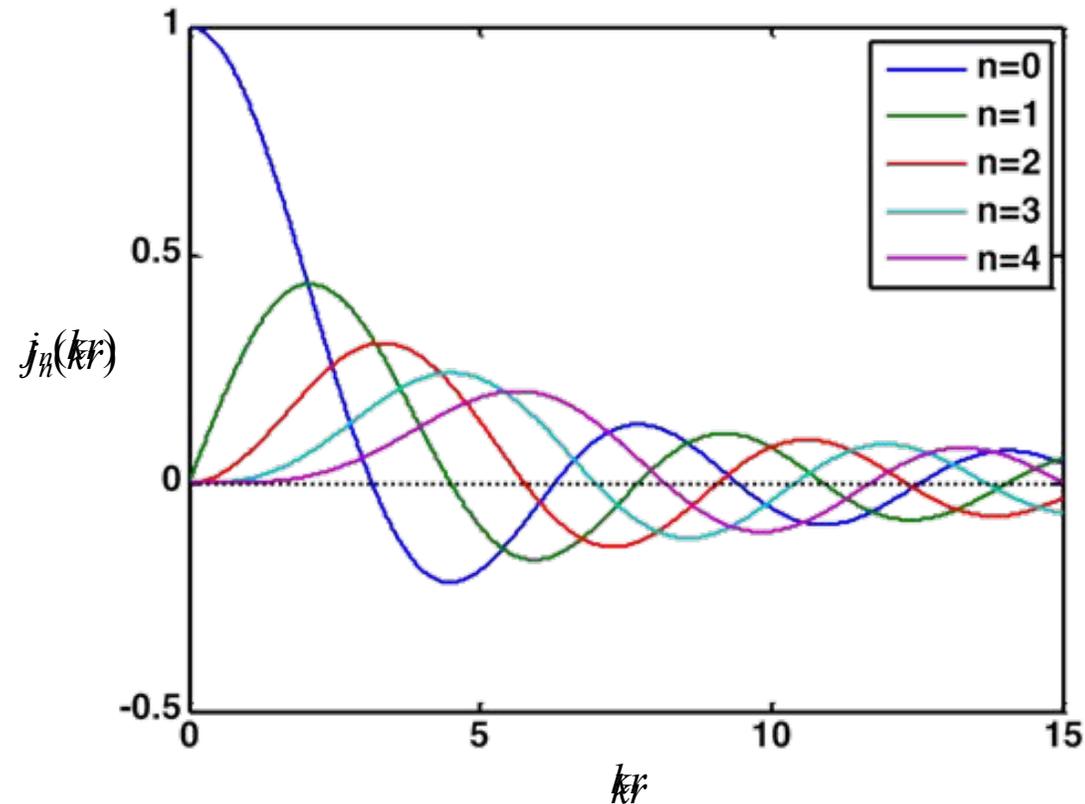
n=5

n=17

n=49

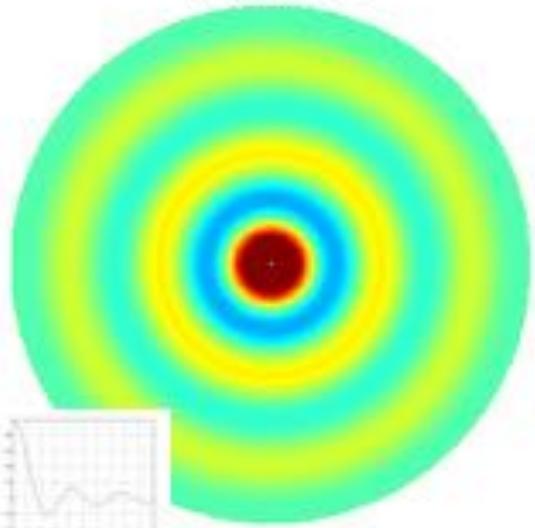
The Spherical Bessel Functions

- The radial dependence of a sound field is expressed by the spherical Bessel functions

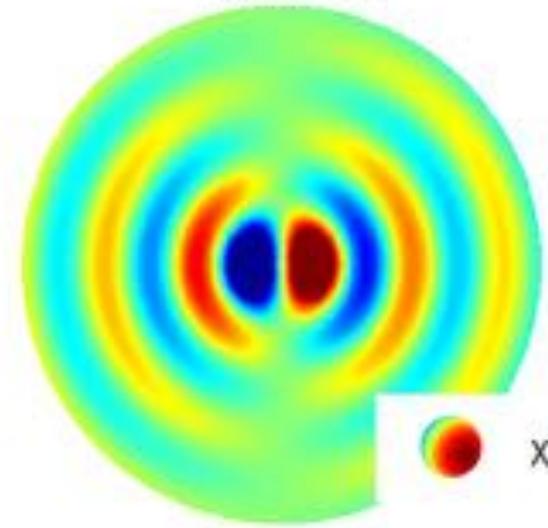




$$i^0 j_0(kr) Y_0^0(\theta, \varphi)$$



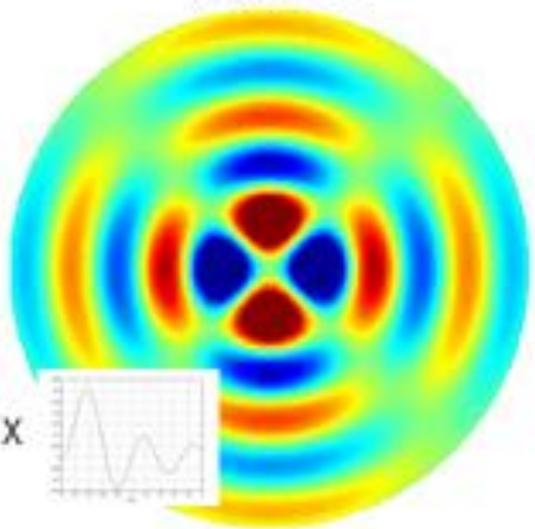
$$i^1 j_1(kr) Y_1^0(\theta, \varphi)$$



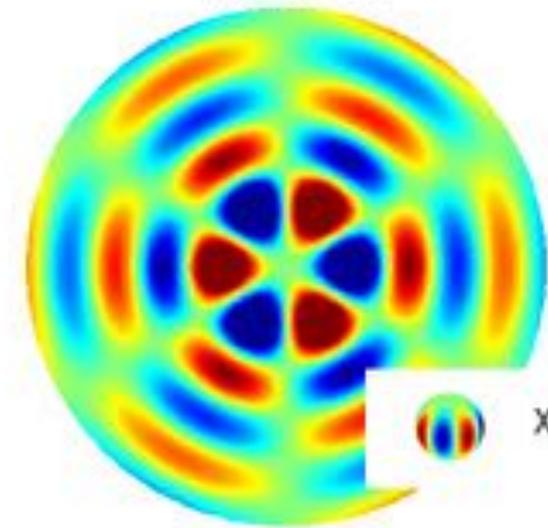
Basis functions

$$i^n j_n(kr) Y_n^m(\theta, \varphi)$$

$$i^2 j_2(kr) Y_2^0(\theta, \varphi)$$



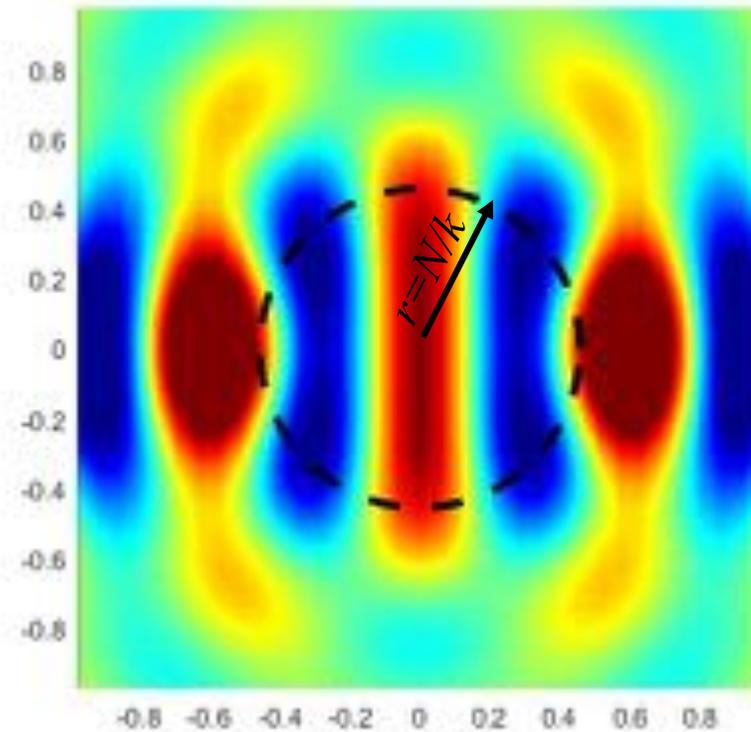
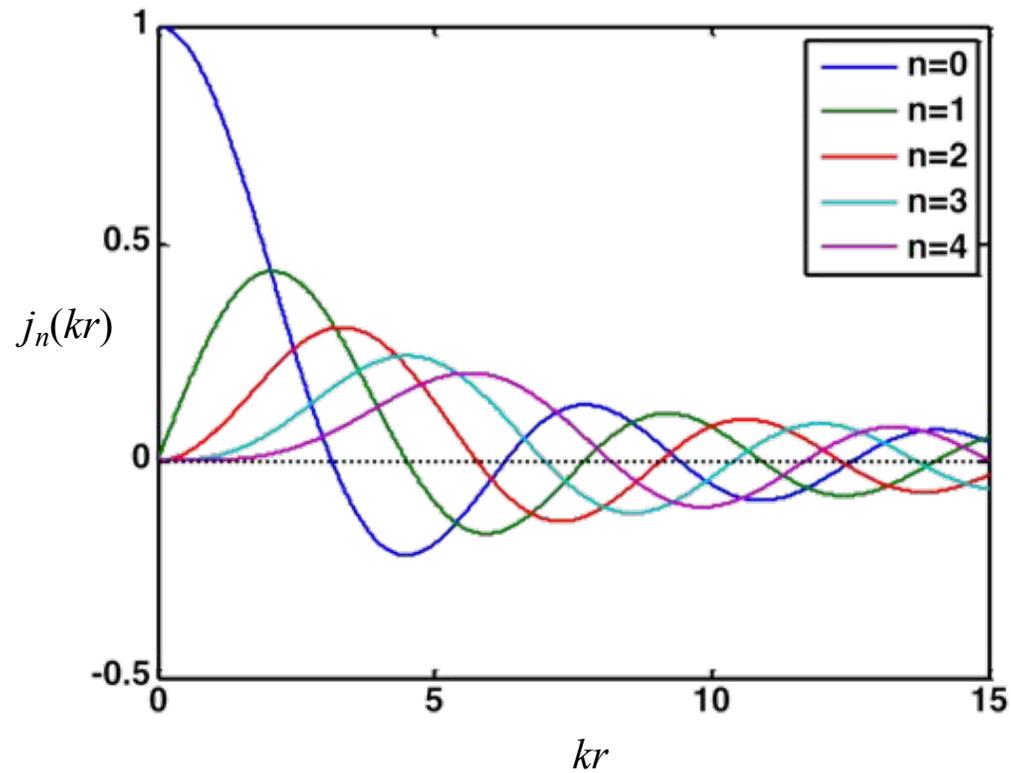
$$i^2 j_2(kr) Y_2^2(\theta, \varphi)$$



Series truncation

$$p(r, \theta, \varphi) \approx \sum_{n=0}^N \sum_{m=-n}^n B_{nm} i^n j_n(kr) Y_n^m(\theta, \varphi)$$

$$N \geq kr$$



Mode matching

Target sound field

$$p_T(r, \theta, \varphi) \approx \sum_{n=0}^N \sum_{m=-n}^n B_{nm} i^n j_n(kr) Y_n^m(\theta, \varphi)$$

Basis functions

Reproduced sound field

$$p(r, \theta, \varphi) \approx \sum_{n=0}^N \sum_{m=-n}^n \sum_{\ell=1}^L H_{nm\ell} q_\ell i^n j_n(kr) Y_n^m(\theta, \varphi)$$

Least squares minimisation

$$J = \|p - p_T\|^2 \quad \longrightarrow \quad J = \sum_{n=0}^N \sum_{m=-n}^n \left| \sum_{\ell=1}^L H_{nm\ell} q_\ell - B_{nm} \right|^2 \quad \longrightarrow \quad J = \| \mathbf{H}\mathbf{q} - \mathbf{b} \|^2$$

Loudspeaker field coeff. matrix
Loudspeaker signals

Target field coefficients

Multi-zone mode matching

$$\mathbf{b}_{M \times 1} = \begin{bmatrix} \mathbf{b}_A \\ \dots \\ \mathbf{b}_B \end{bmatrix}, \quad \mathbf{H}_{M \times L} = \begin{bmatrix} \mathbf{H}_A \\ \dots \\ \mathbf{H}_B \end{bmatrix}$$

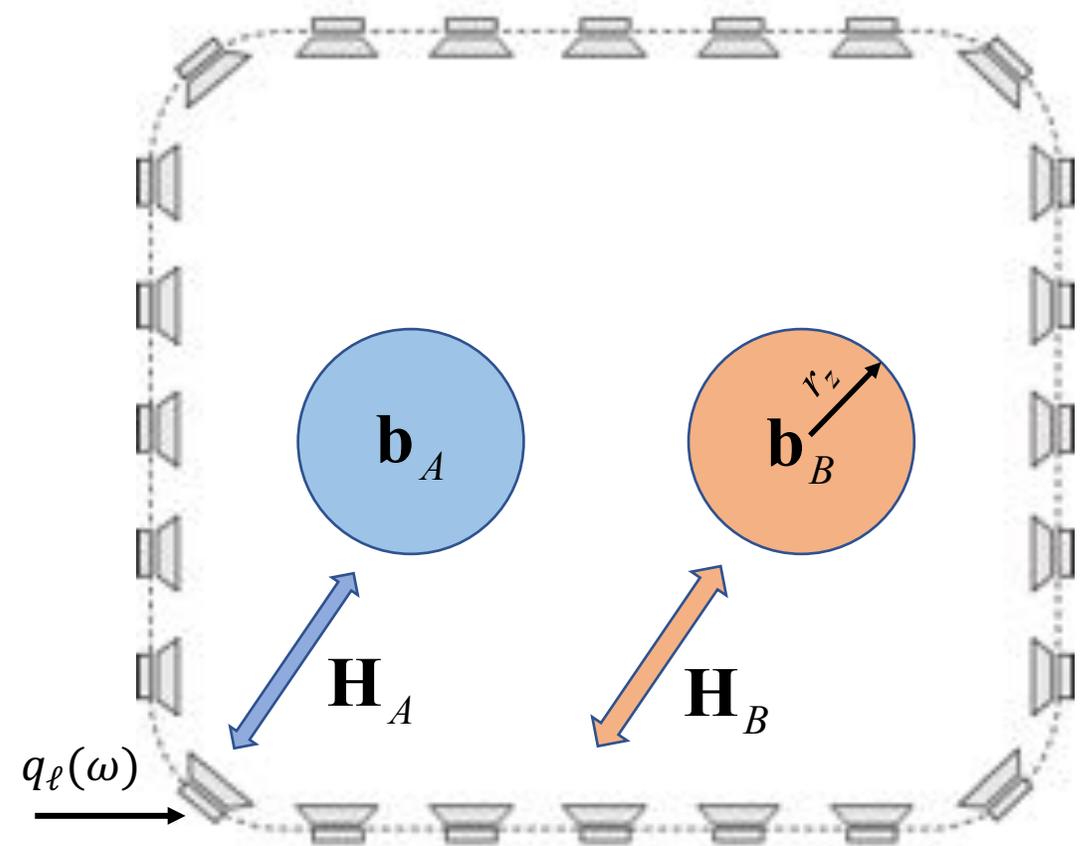
$$M \geq N_z (kr_z + 1)^2$$

- Least squares minimisation

$$J = \|\mathbf{H} \mathbf{q} - \mathbf{b}\|^2$$

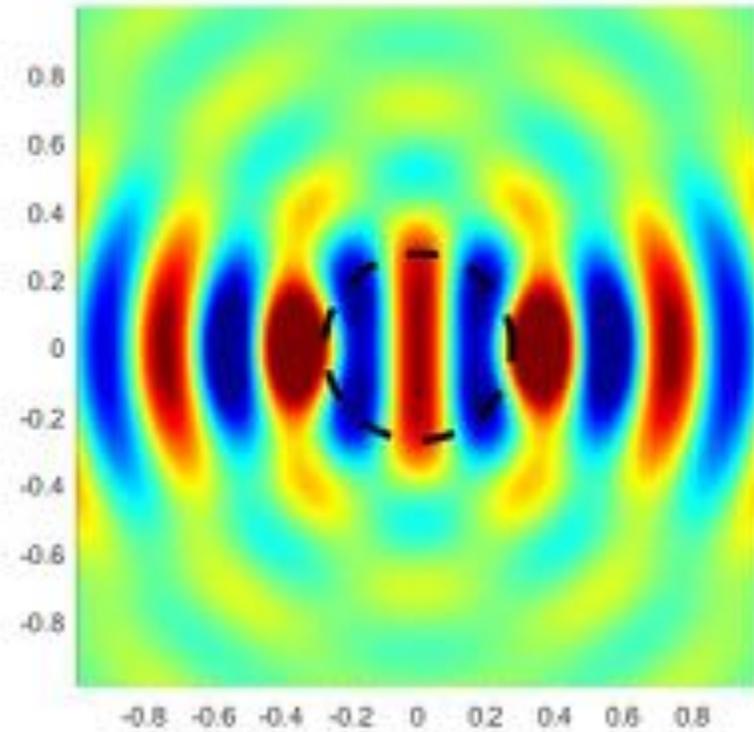
$$\mathbf{q}_{opt} = \mathbf{H}^\dagger \mathbf{b} = \left(\mathbf{H}^H \mathbf{H} + \beta \mathbf{I} \right)^{-1} \mathbf{H}^H \mathbf{b}$$

- Conceptually similar to pressure matching

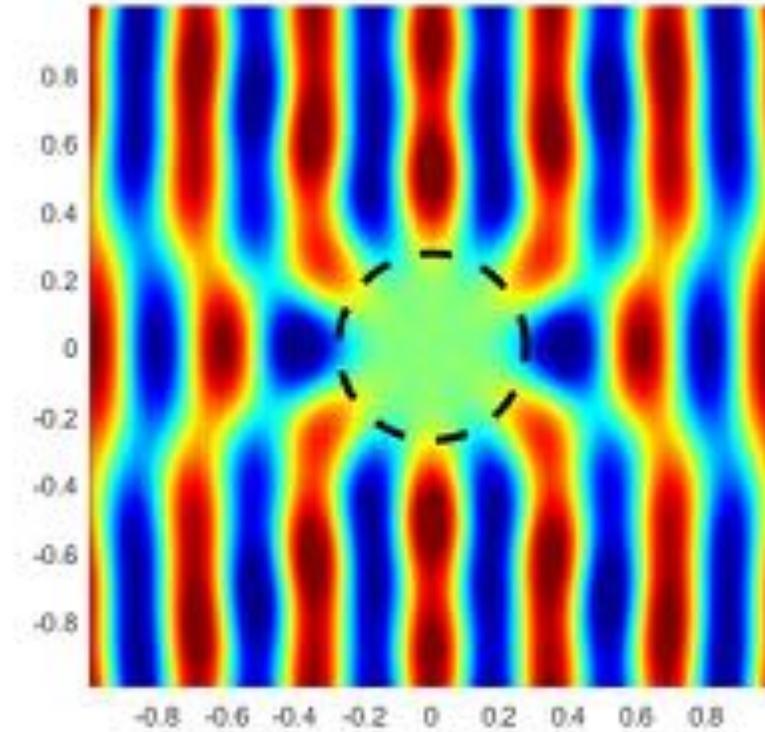


Zone of silence

Truncated field



Complementary field



$$p_N = \sum_{n=0}^N \sum_{m=-n}^n B_{nm} i^n j_n(kr) Y_n^m(\theta, \varphi)$$

$$p_N^{(C)} = \sum_{n=N+1}^{\infty} \sum_{m=-n}^n B_{nm} i^n j_n(kr) Y_n^m(\theta, \varphi)$$

Zone of silence

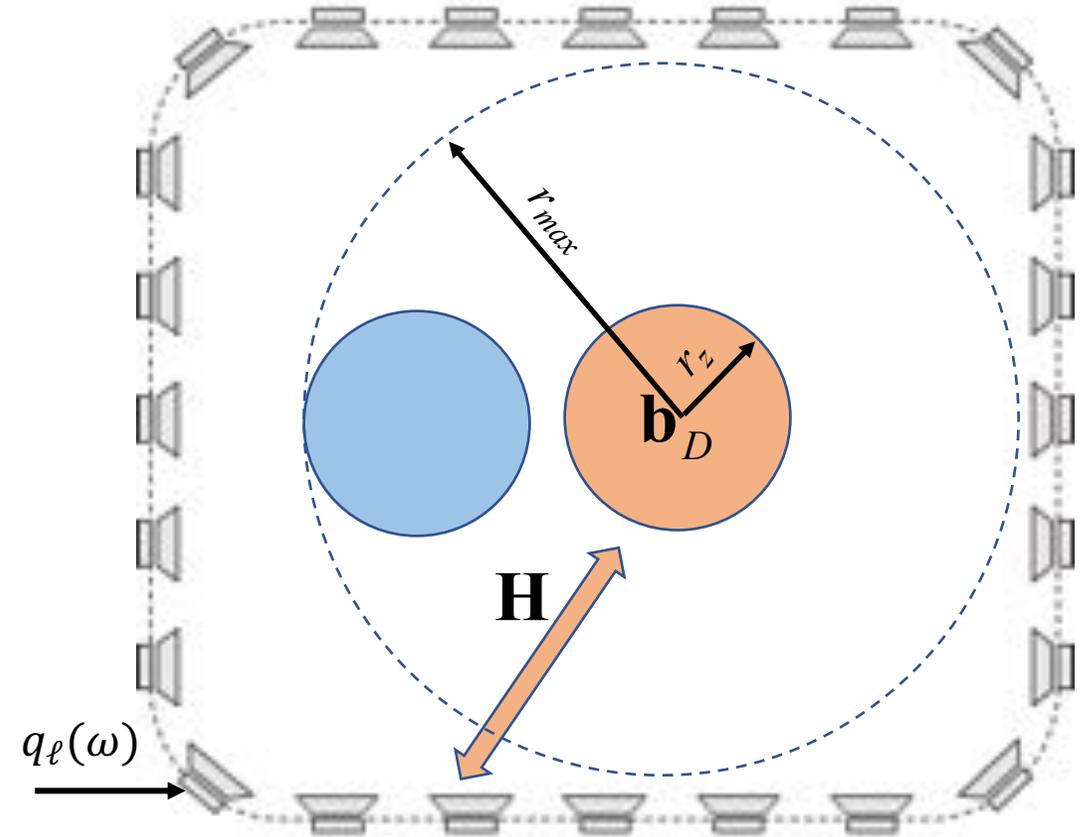
$$\mathbf{b}_D = [0, 0, 0, \dots, B_{M+1,0}, \dots, B_{NN}]^T$$

$$M = kr_z, \quad N \geq kr_{\max}$$

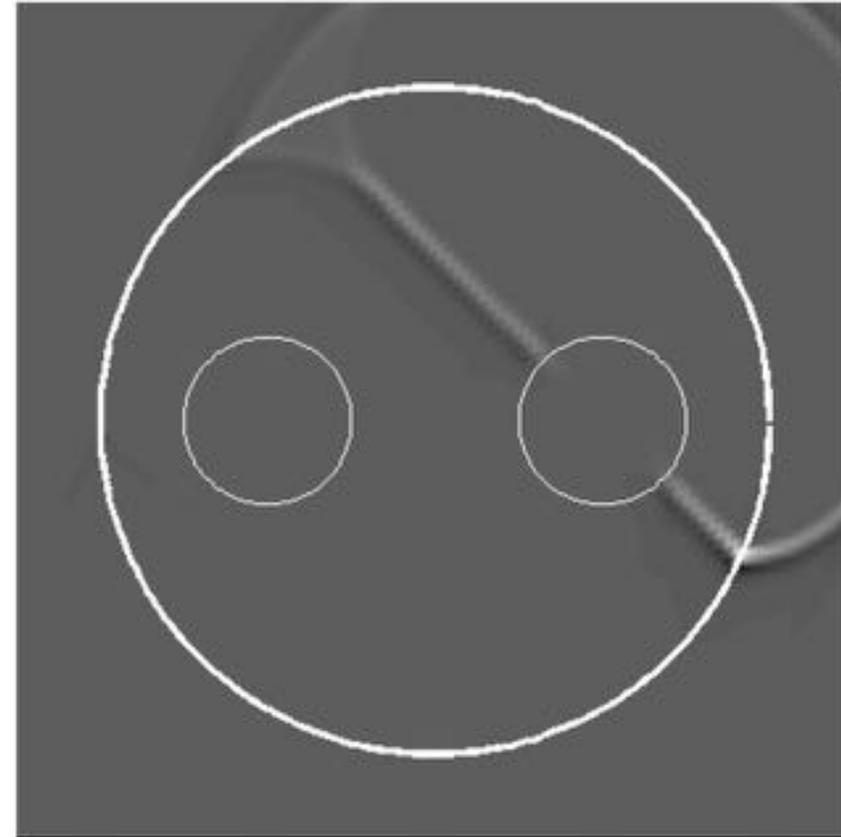
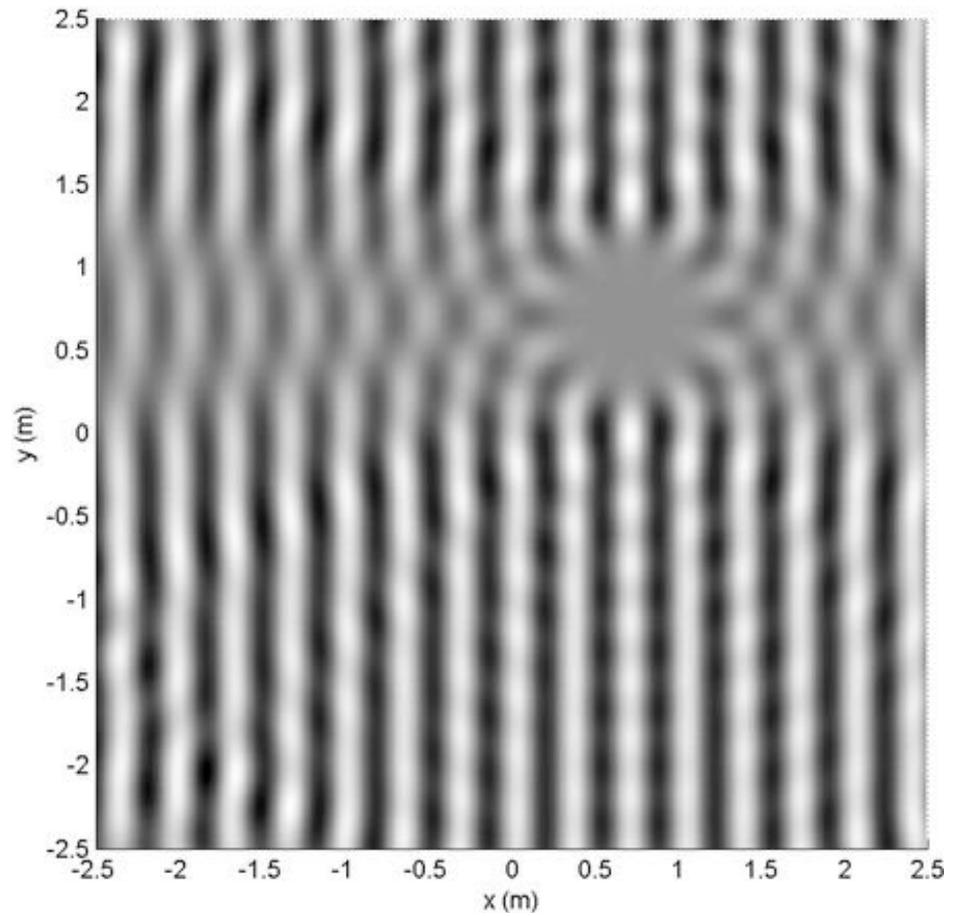
- Least squares minimisation

$$J = \|\mathbf{H} \mathbf{q} - \mathbf{b}\|^2$$

$$\mathbf{q}_{opt} = \mathbf{H}^\dagger \mathbf{b} = (\mathbf{H}^H \mathbf{H} + \beta \mathbf{I})^{-1} \mathbf{H}^H \mathbf{b}$$

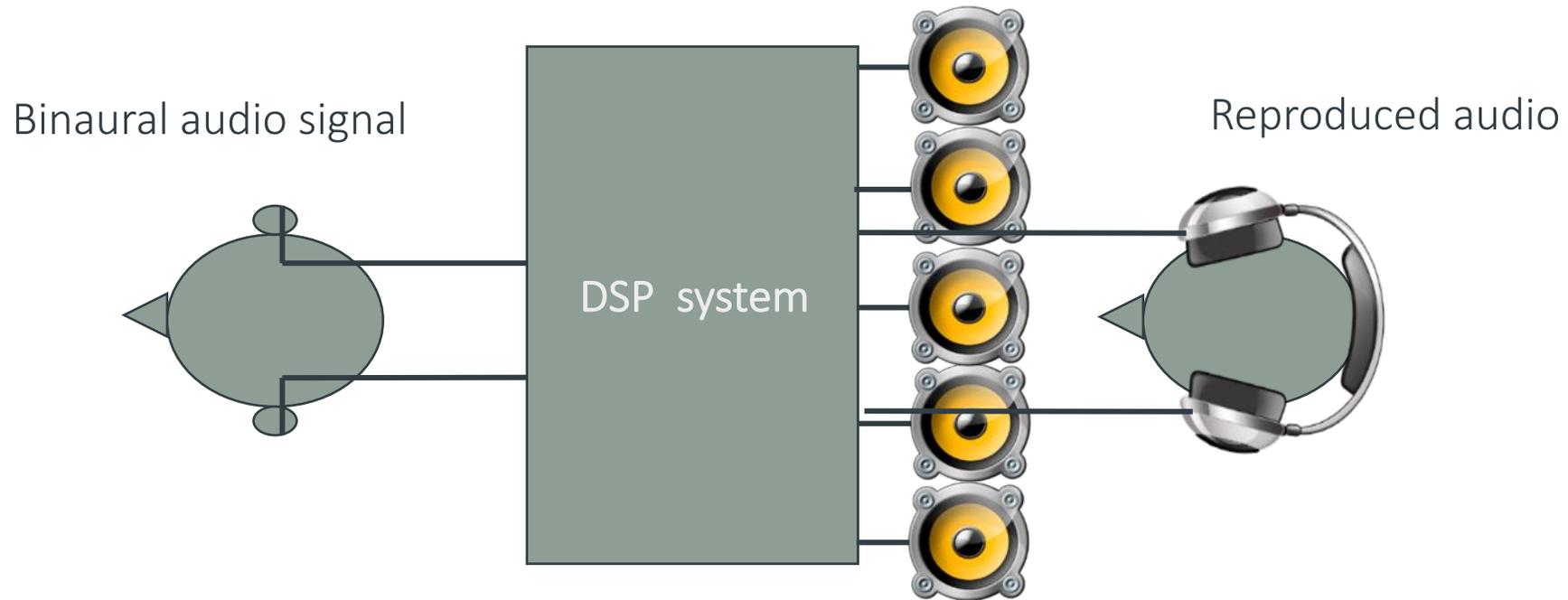


Zone of silence



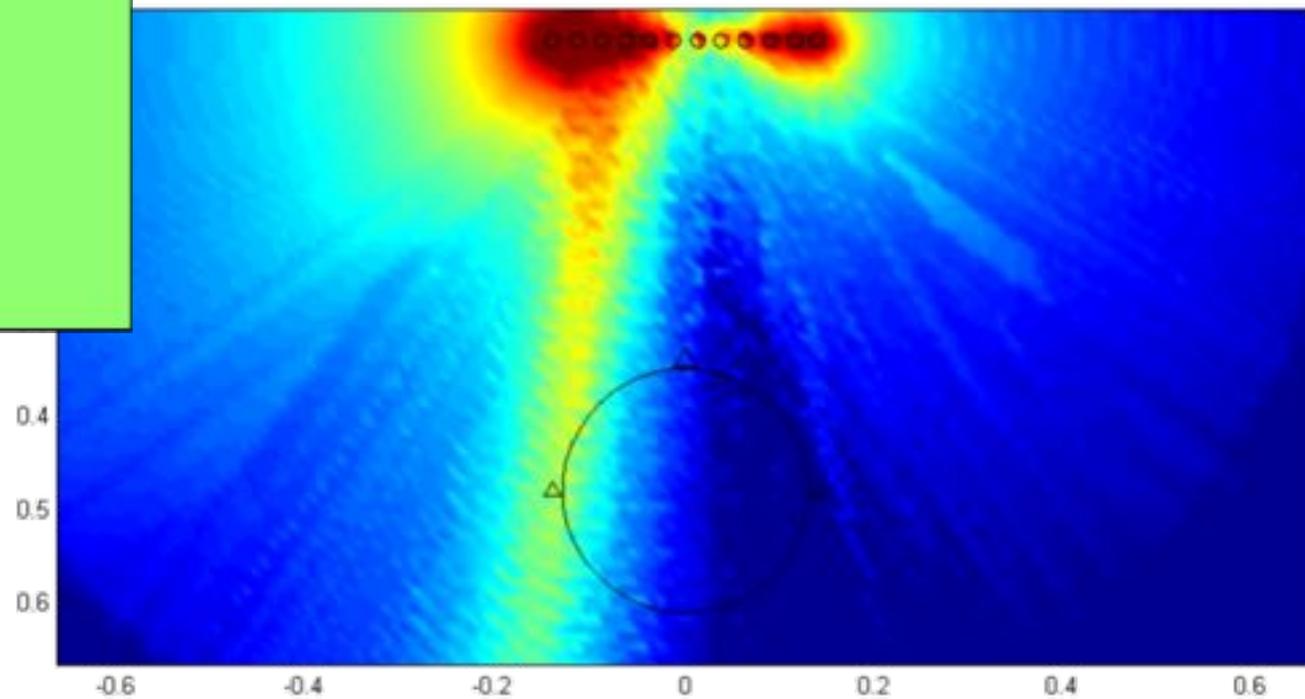
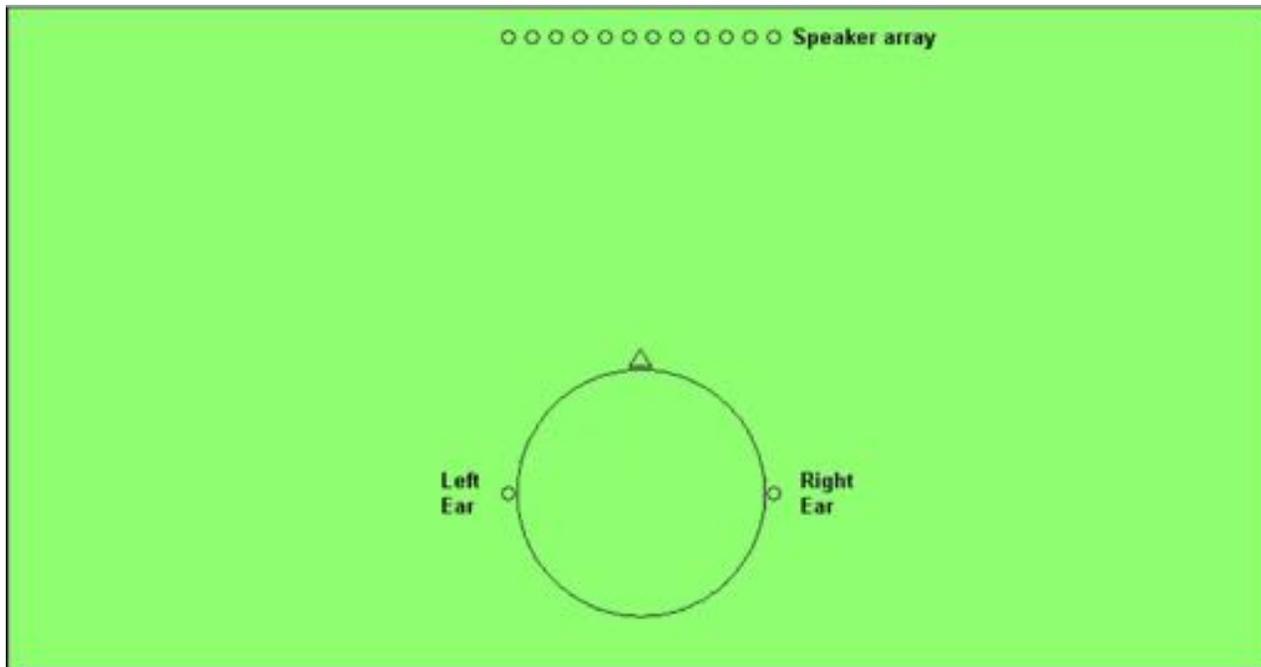
Poletti and Fazi, 2015. "An approach to generating two zones of silence with application to personal sound systems". JASA

Cross-Talk Cancellation

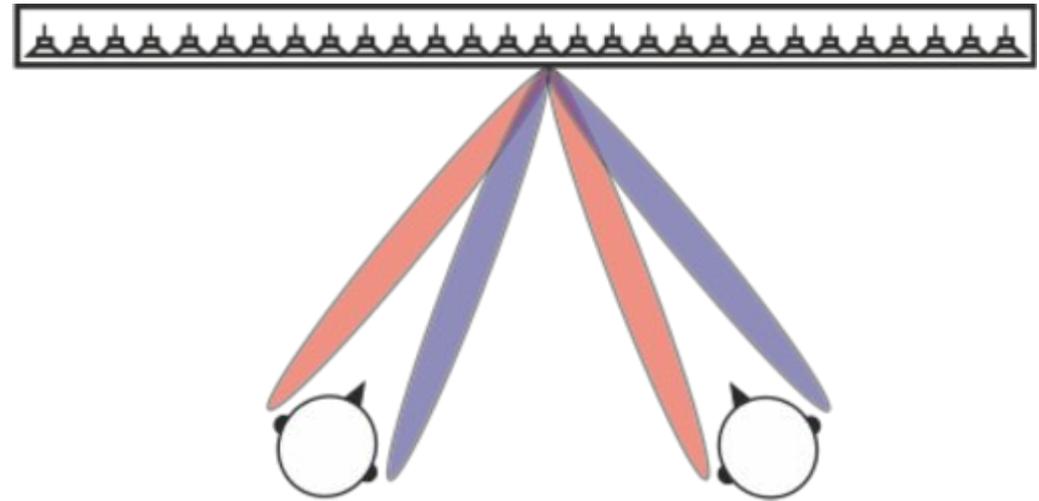
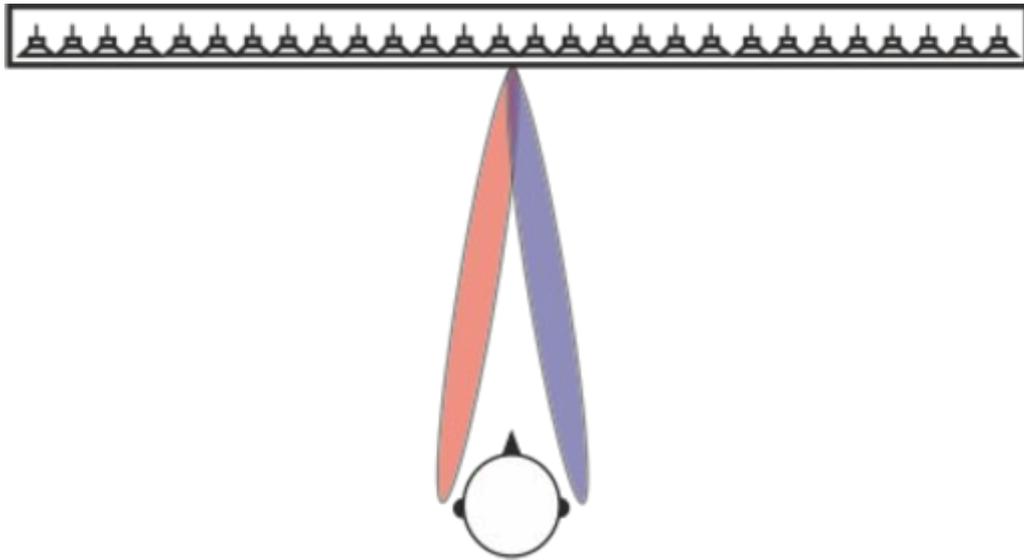


Cross-Talk Cancellation allows for the delivery of independent signals to the two ears of the listener

Cross-Talk Cancellation with a loudspeaker array



Multi-listener Cross-Talk Cancellation



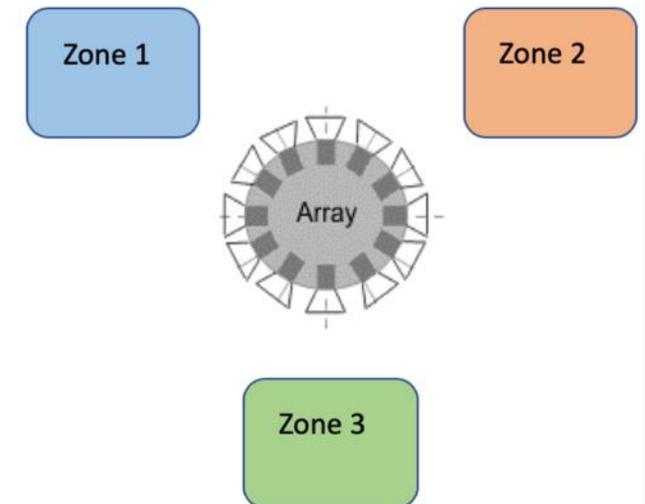
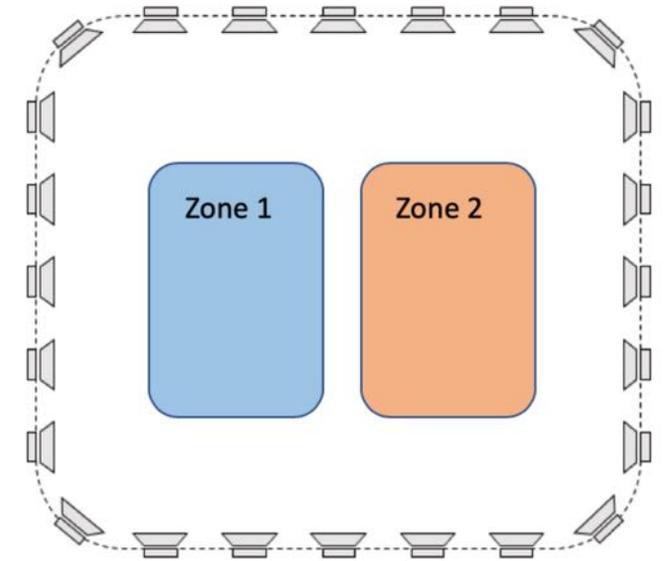


Putting it all together: summary & conclusions



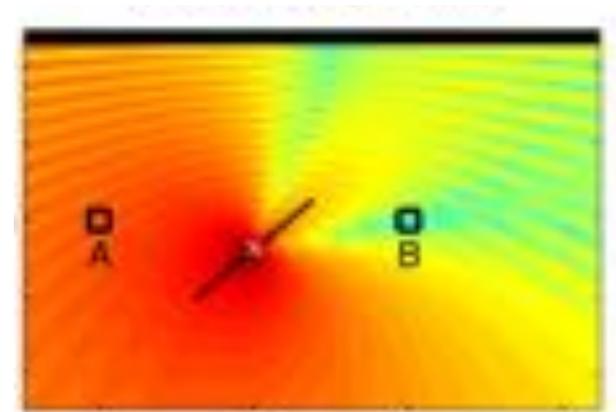
Formulation

- Why? For convenient shared experiences
- Design choices
 - Zones, target experience, system setup, calibration & evaluation
- Key approaches:
 - pressure matching (PM), acoustic contrast control (ACC)
- Performance
 - Goodness of fit, contract, effort, bandwidth, planarity



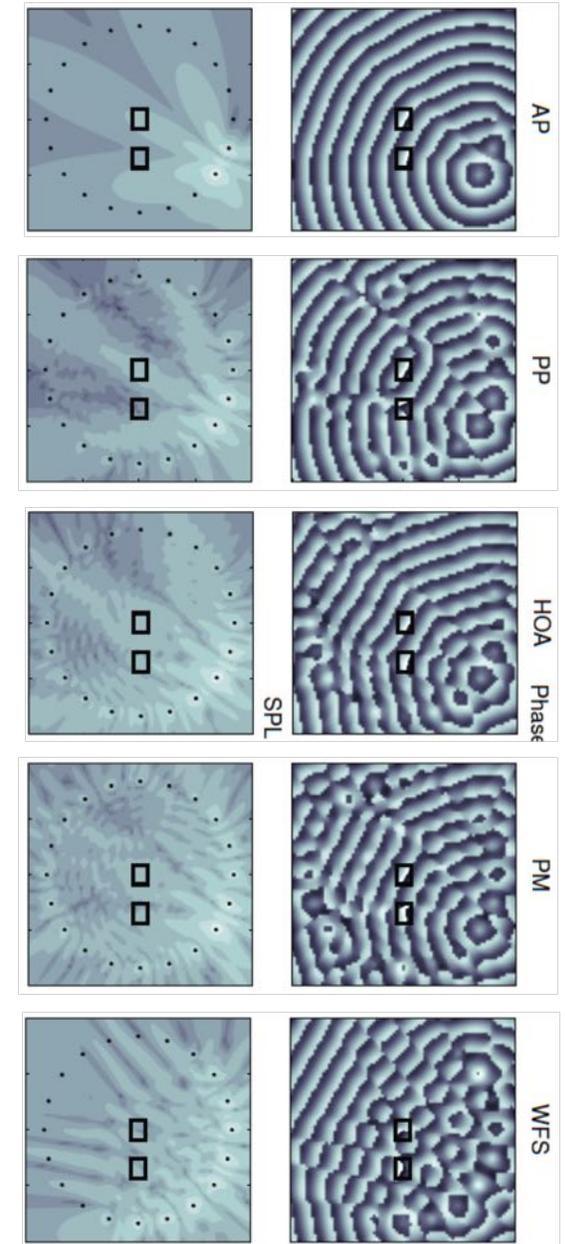
Engineering

- Design-performance axioms
 - Contrast vs. target quality (ACC vs. PM), # sources \rightarrow bandwidth
- Robust design
 - Frequency-dependent regularization, conditioning/effort/error-model methods, performance-robustness trade off
- Room effects
 - Design & calibrate for the earlies
 - Truncate & regularize for the late
- PyZones interactive sound zone filter design



User experience

- Line-array sound zone listening demonstration
- Perceptual models of performance
 - Sound quality
 - Interference
- Alternative approaches
 - Perceptually-inspired planarity control (PC)
 - Spherical harmonics domain methods



Challenges & opportunities

- Perceptual models
 - Timbral & spatial target quality
 - Distraction & interference
 - Roles of context & of listener goals (attention)
- Adaptive systems
 - Moving listeners
 - Large & dynamic setups
 - Slick & robust calibration and synchronisation
- Paradigm
 - General framework for multi-listener spatial sound
 - Deployment of personalisable content



ICASSP Tutorial: Personalising sound over loudspeakers

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Thank you for your engagement!

Any questions?



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