



 Estimation of the Sound Field at Arbitrary
Positions in Distributed Microphone Networks
Based on Distributed Ray Space Transform

M. Pezzoli, F. Borra, F. Antonacci, A. Sarti, S. Tubaro



- ❑ Goals
- ❑ State of the art
- ❑ Sound Field Estimation
- ❑ Parameter Estimation
 - Source Localization
 - Radiance Pattern Estimation
 - Signal Reconstruction
- ❑ Sound Synthesis
- ❑ Simulations and Experiments Results
- ❑ Conclusions
- ❑ Future Works

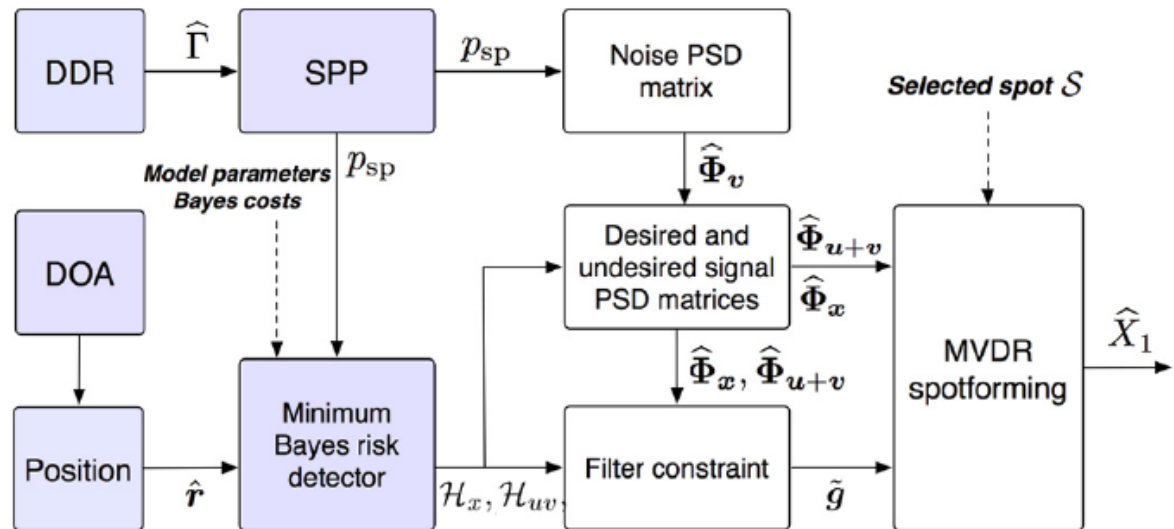


- ❑ Goal: reconstruct the sound field at an arbitrary position in the presence of multiple sources.
- ❑ Input data: signals coming from spatially-distributed compact arrays
- ❑ Assumptions:
 - ❑ Low reverberation
 - ❑ Virtual microphones has an omnidirectional sensitivity pattern
 - ❑ Coplanar arrays and sources
- ❑ Approach: sources parameterized by their signal, radiance pattern and location in space



- ❑ Sources are extracted from a spot of interest (SOI), while reducing interferers and background noise in a fully data-dependent framework, which requires the estimation of the PSD of the sources on the scene (*)
- ❑ Sources are assumed to be omnidirectional

Spotforming method [1]



(*) M. Taseska, E.A.P. Habets, "Spotforming: Spatial Filtering With Distributed Arrays for Position-Selective Sound Acquisition," in IEEE/ACM TrASLP, Vol. 24, no. 7, pp. 1291-1304, July 2016.

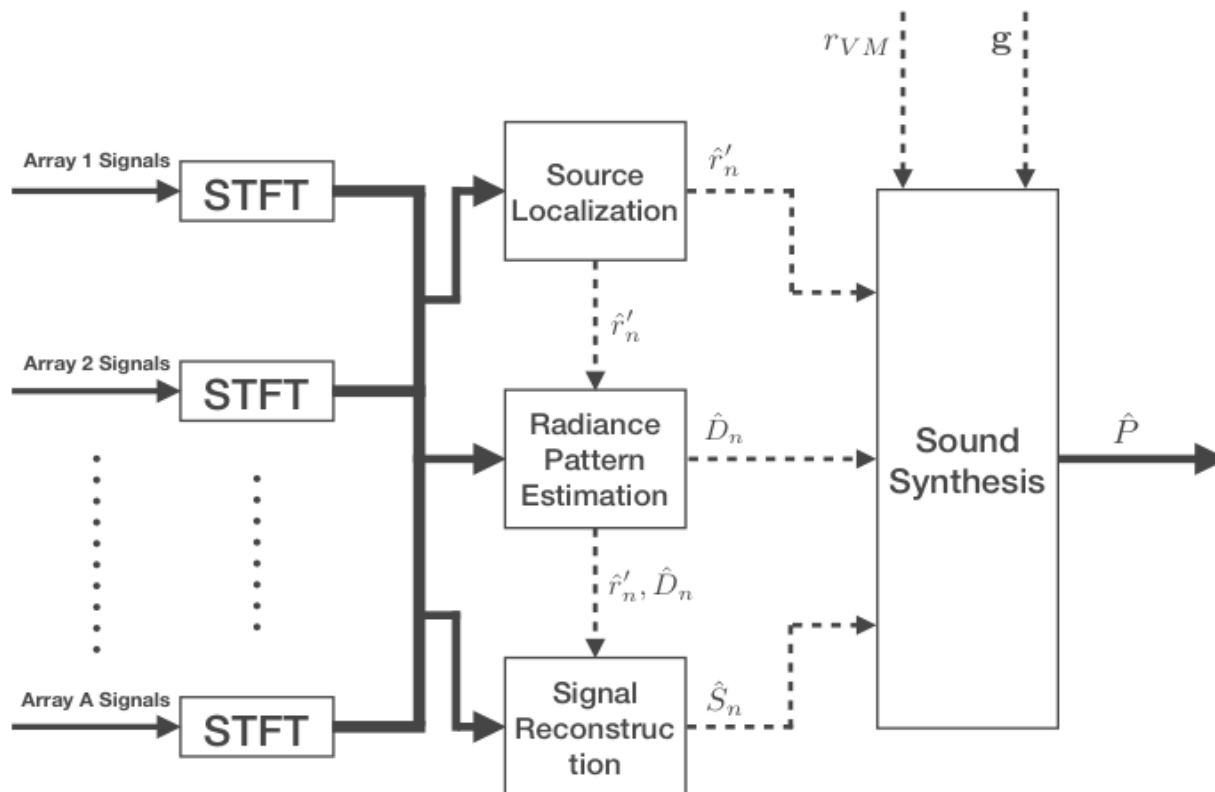


- Higher-Order Ambisonics (HOA) and Wave Field Synthesis (WFS) aim at reconstructing the sound field at a given location (HOA) or region (WFS) in space
- HOA usually has a small rendering region and WFS needs a large number of spatially distributed microphones





We propose a **parametric sound field reconstruction** approach based on the estimation of a **limited number of acoustic source descriptors** (*position, radiation pattern and source signal*), given the signals acquired by a **few arbitrarily placed microphone arrays**.





Signal Model

Signal acquired by the m -th
microphone in a -th array

$$P(\mathbf{r}_m^{(a)}, \omega) = \sum_{n=0}^{N-1} D_n(\theta_{m,n}^{(a)}, \omega) g(\mathbf{r}_m^{(a)}, \mathbf{r}'_n, \omega) S_n(\omega) + e_m^{(a)}(\omega)$$



Signal Model

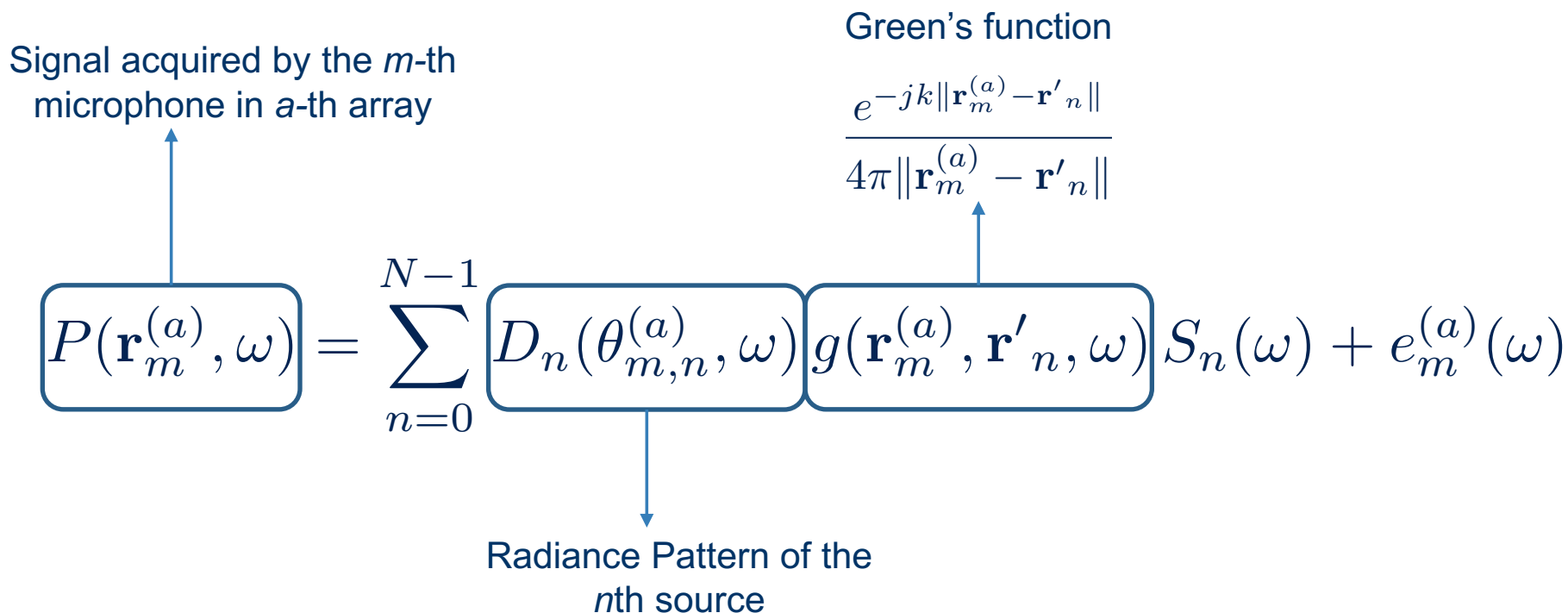
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Radiance Pattern of the
 n th source

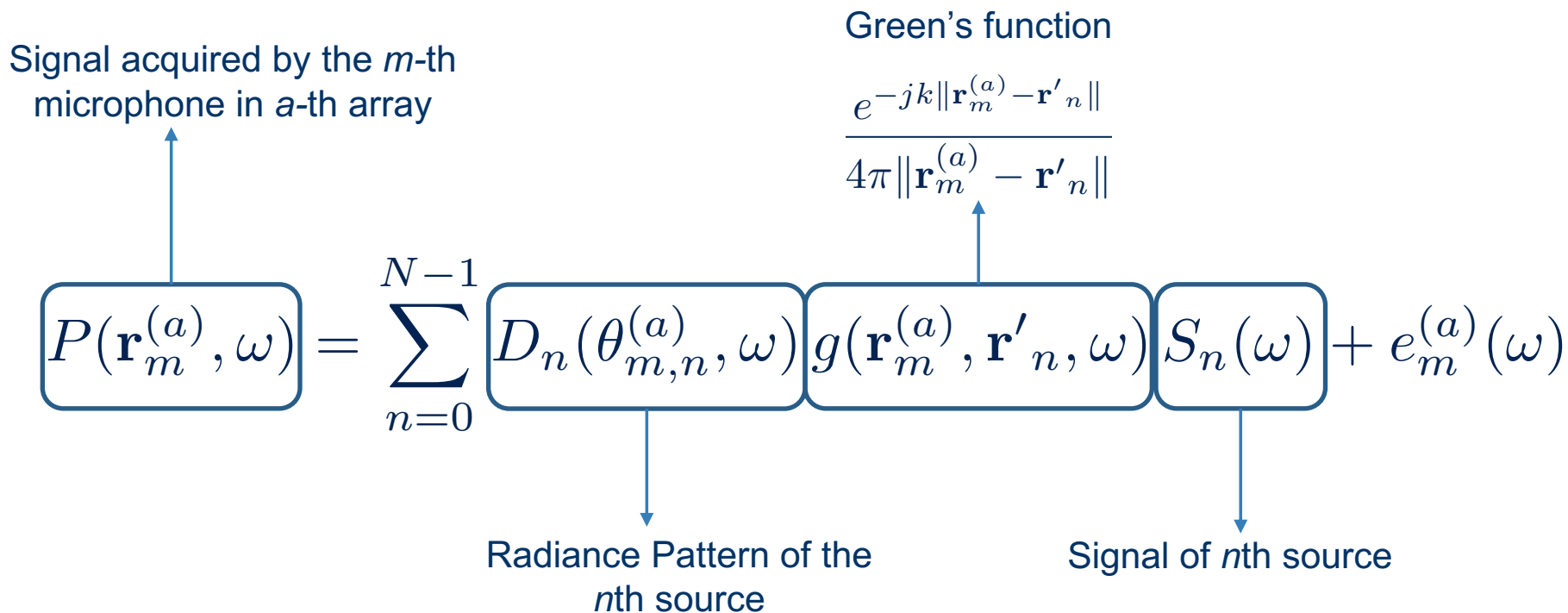


Signal Model



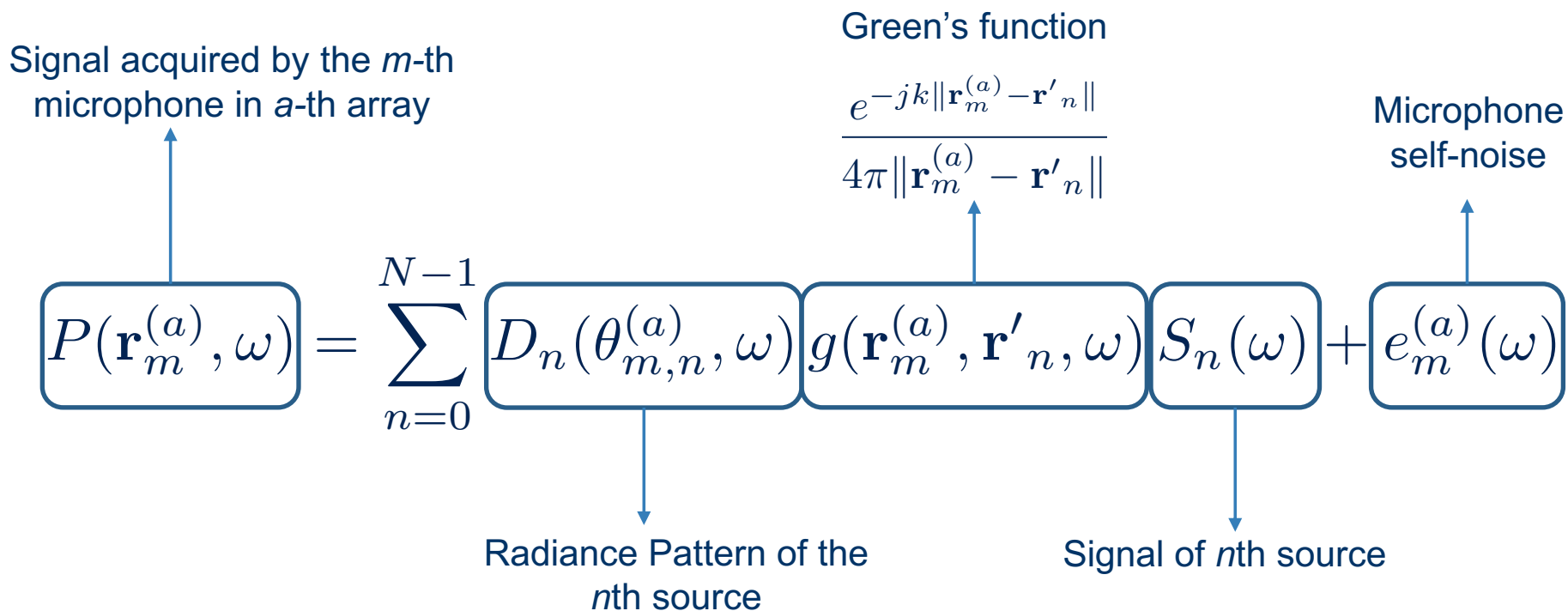


Signal Model





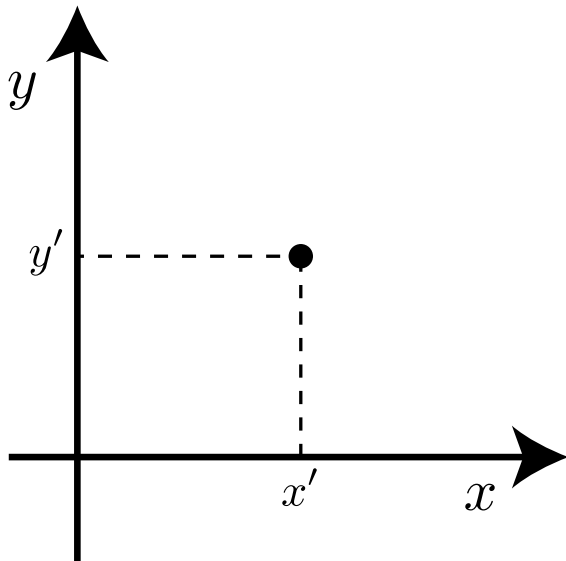
Signal Model





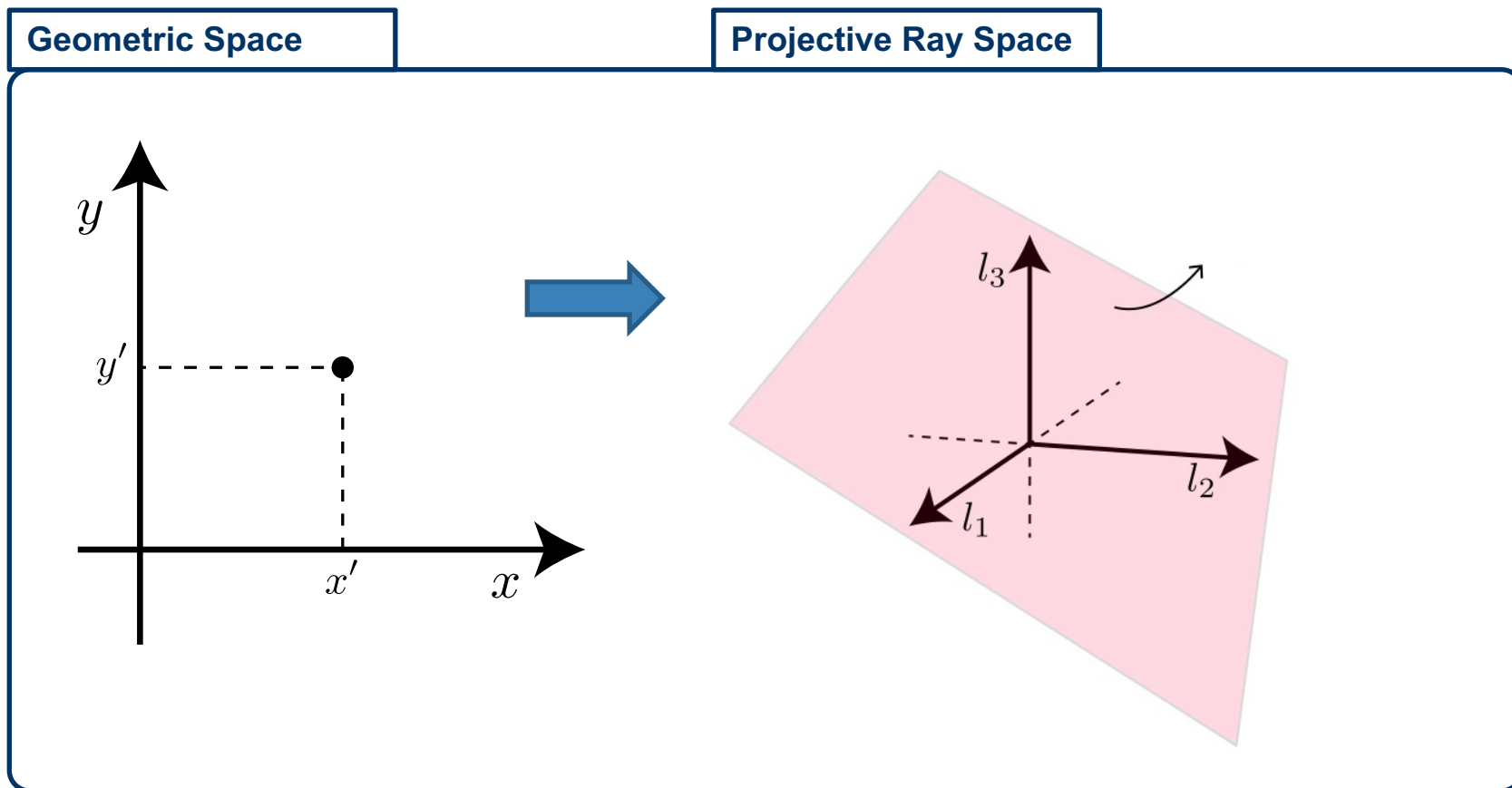
Review of the **Projective Ray Space** representation of a point source

Geometric Space





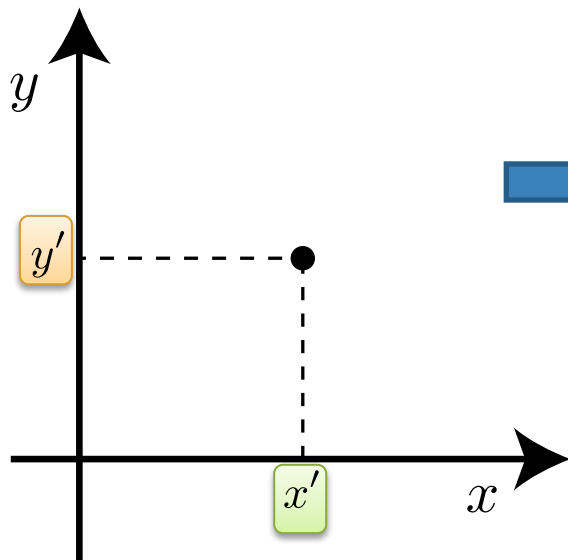
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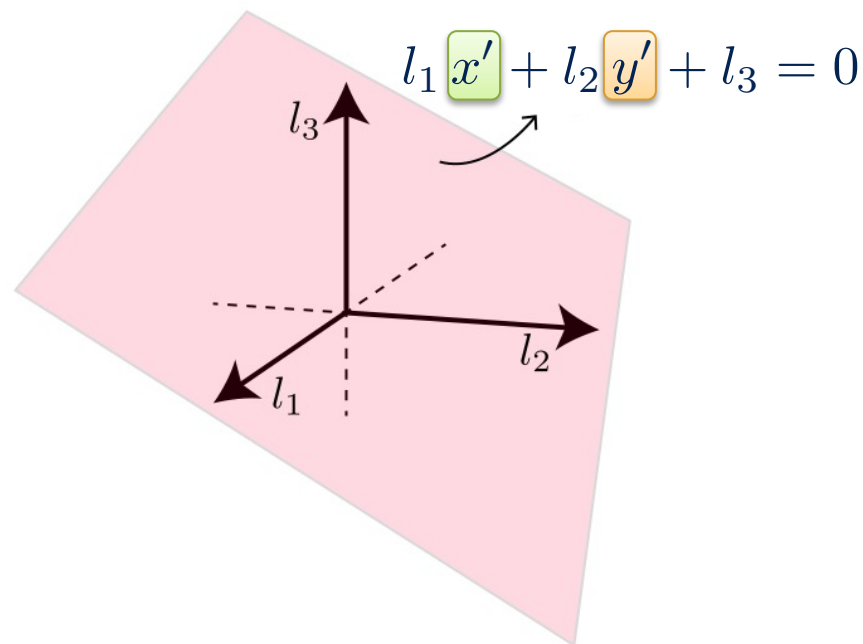


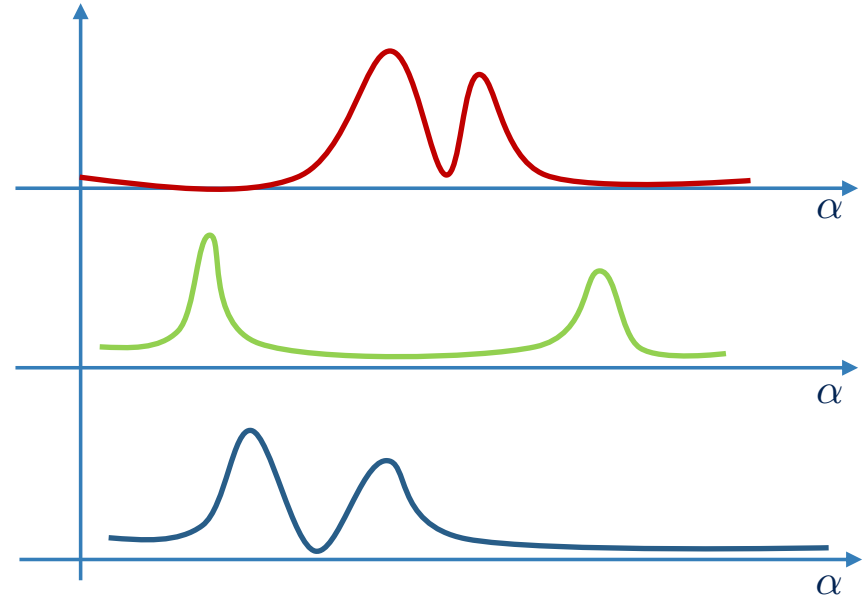
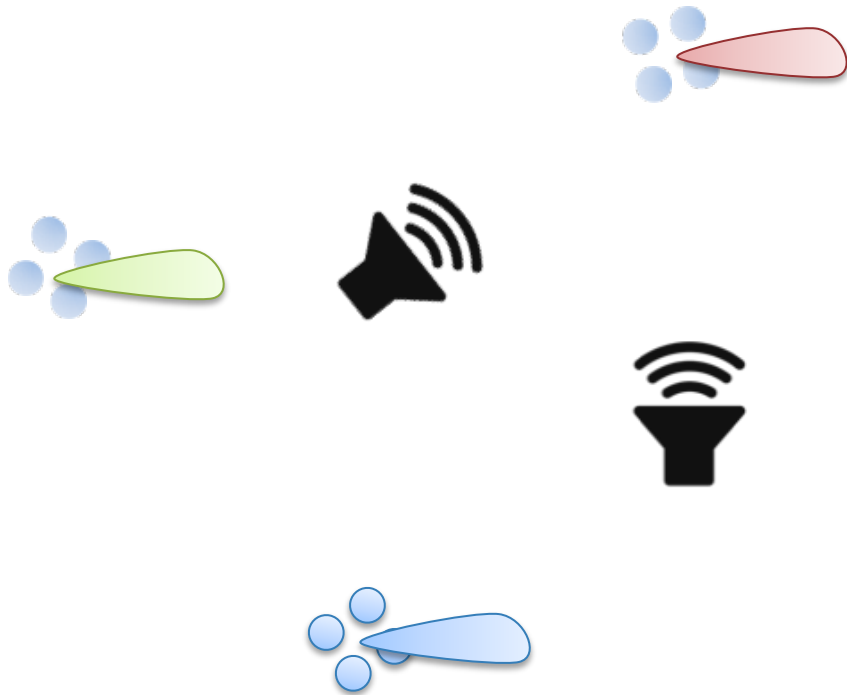
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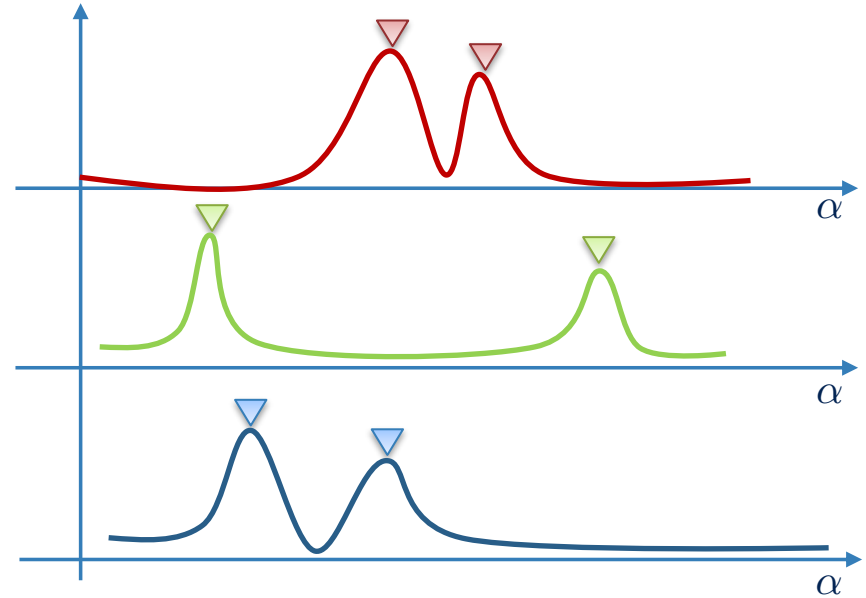
Projective Ray Space







Peak the maxima





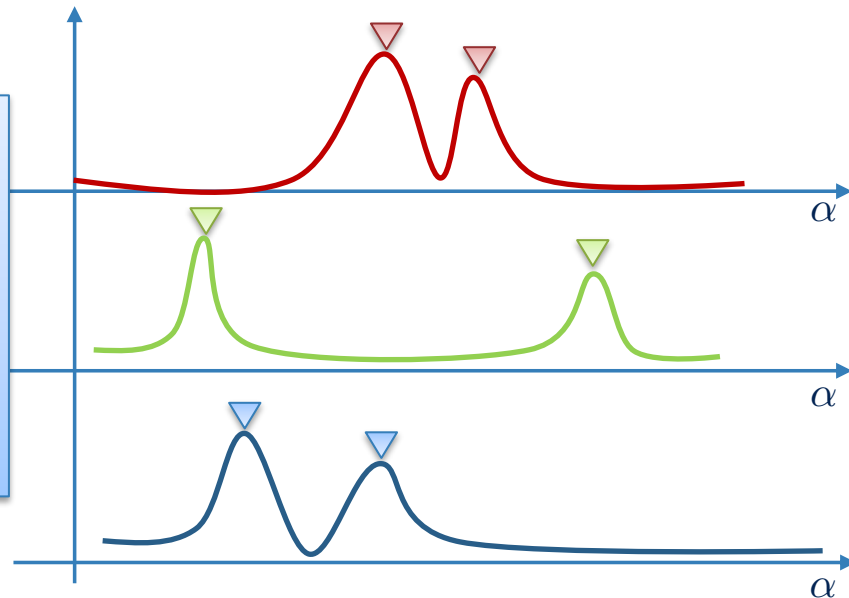
Peak the maxima

Map to the Projective Ray Space

$$l_{1,n}^{(a)} = \gamma \sin(\hat{\alpha}_n^{(a)});$$

$$l_{2,n}^{(a)} = -\gamma \cos(\hat{\alpha}_n^{(a)});$$

$$l_{3,n}^{(a)} = \gamma[y^a \cos(\hat{\alpha}_n^{(a)}) - x^a \sin(\hat{\alpha}_n^{(a)})]$$





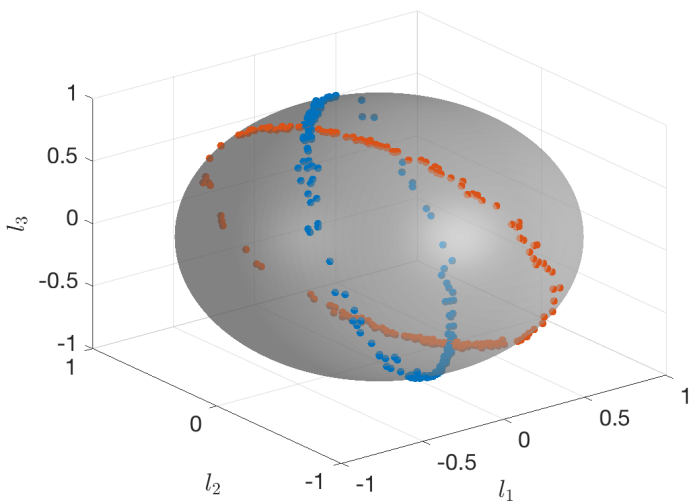
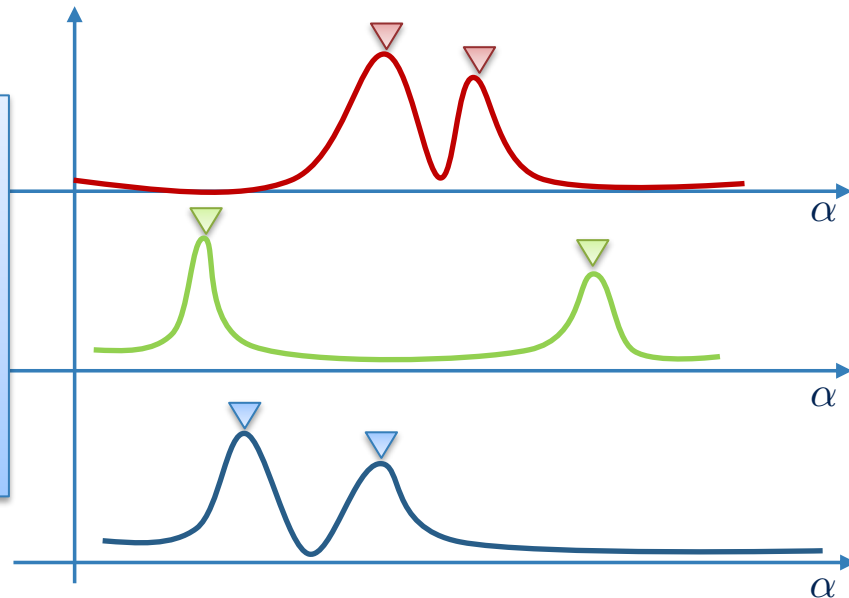
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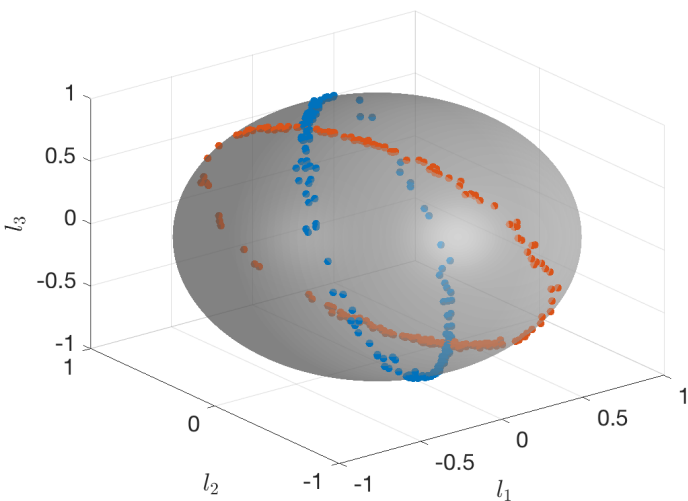
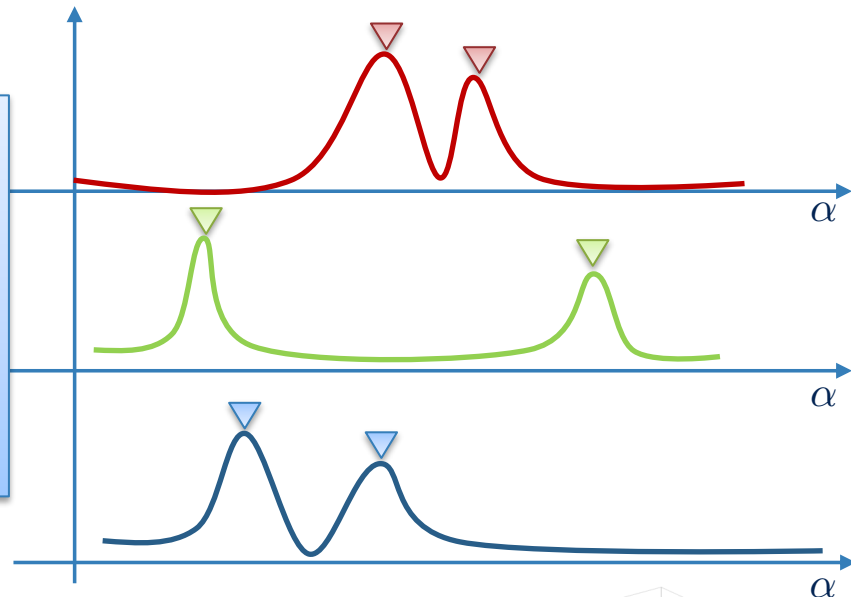
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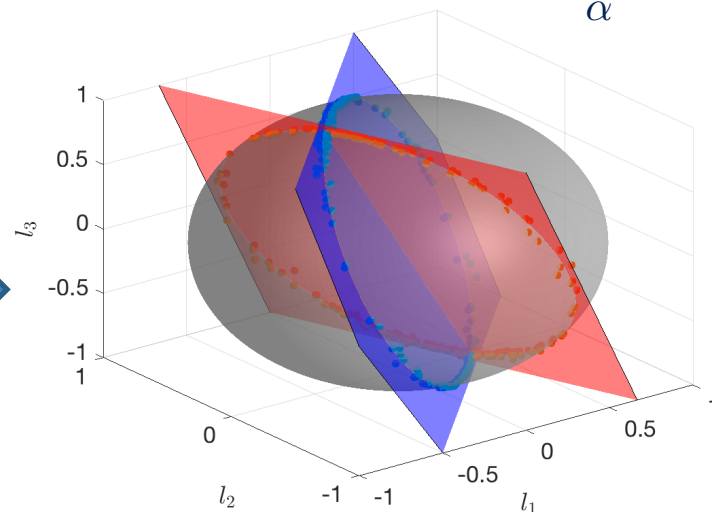
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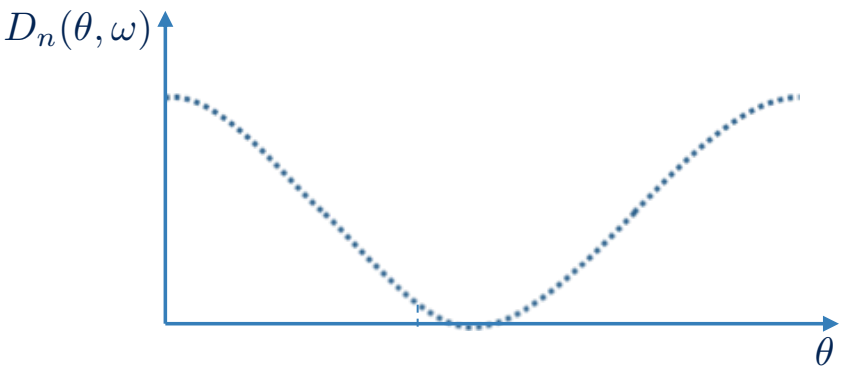
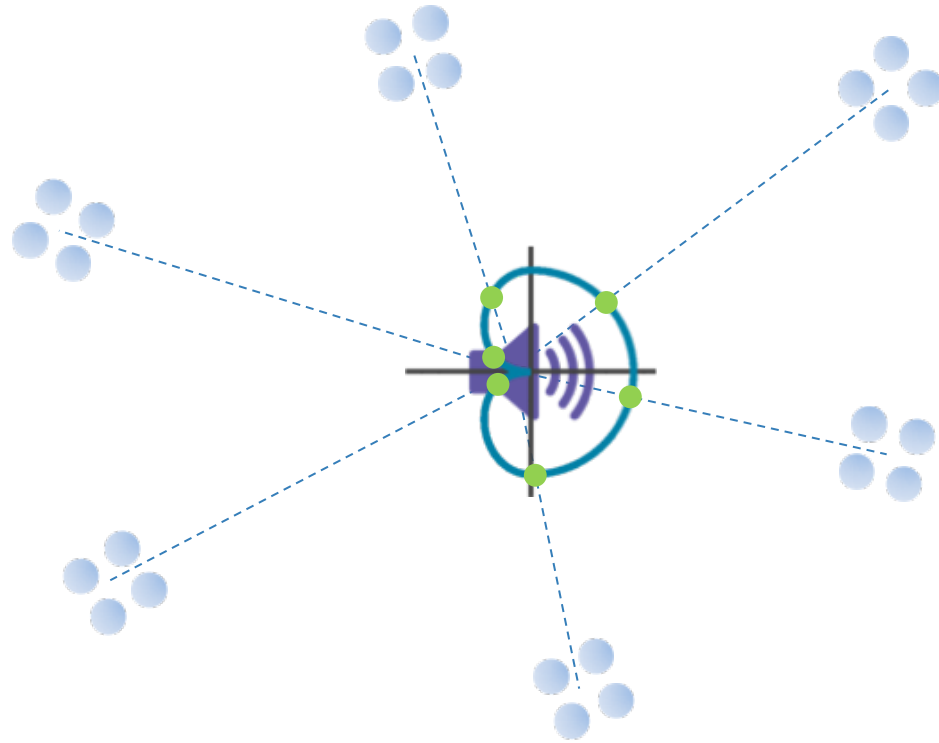


Clustering and linear regression (RANSAC)



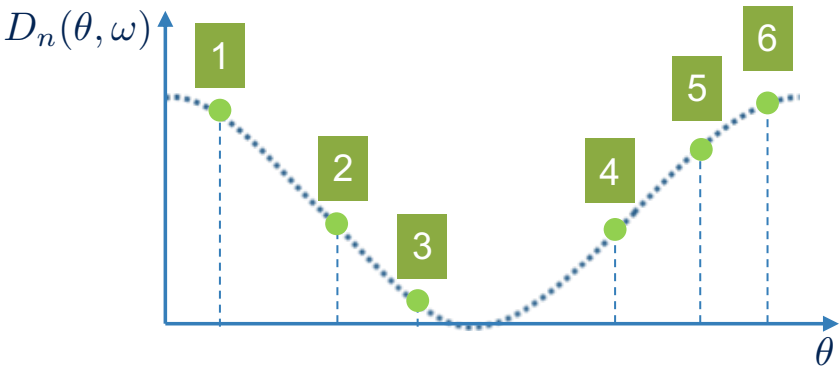
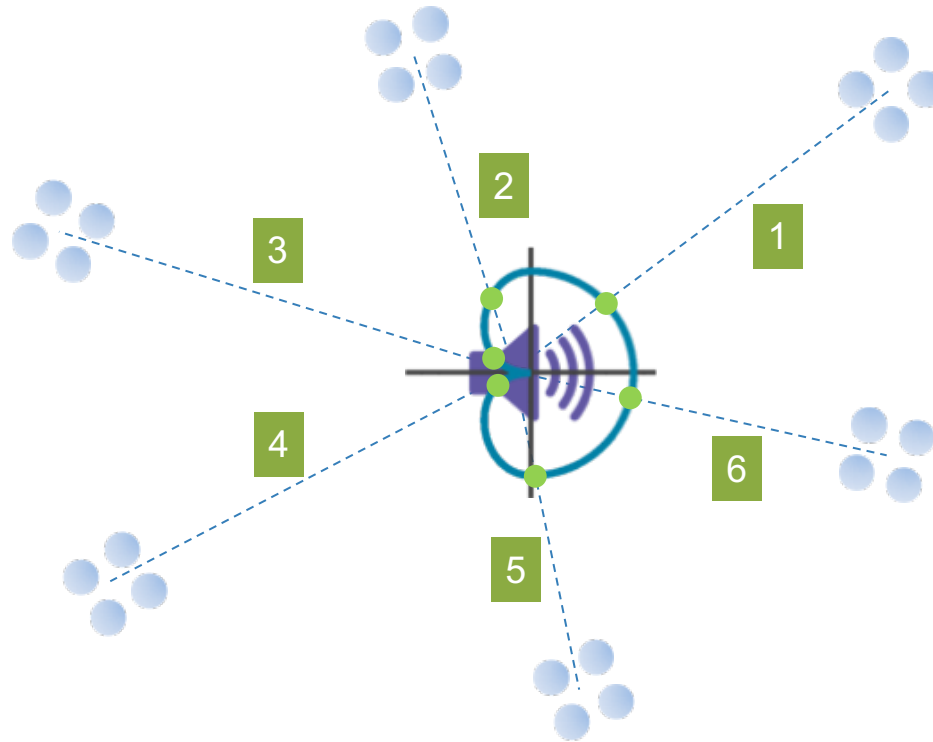


Radiance Pattern Estimation (1/3)



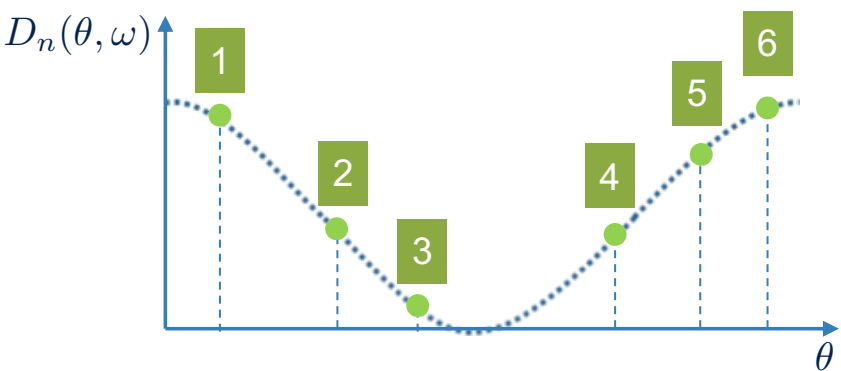
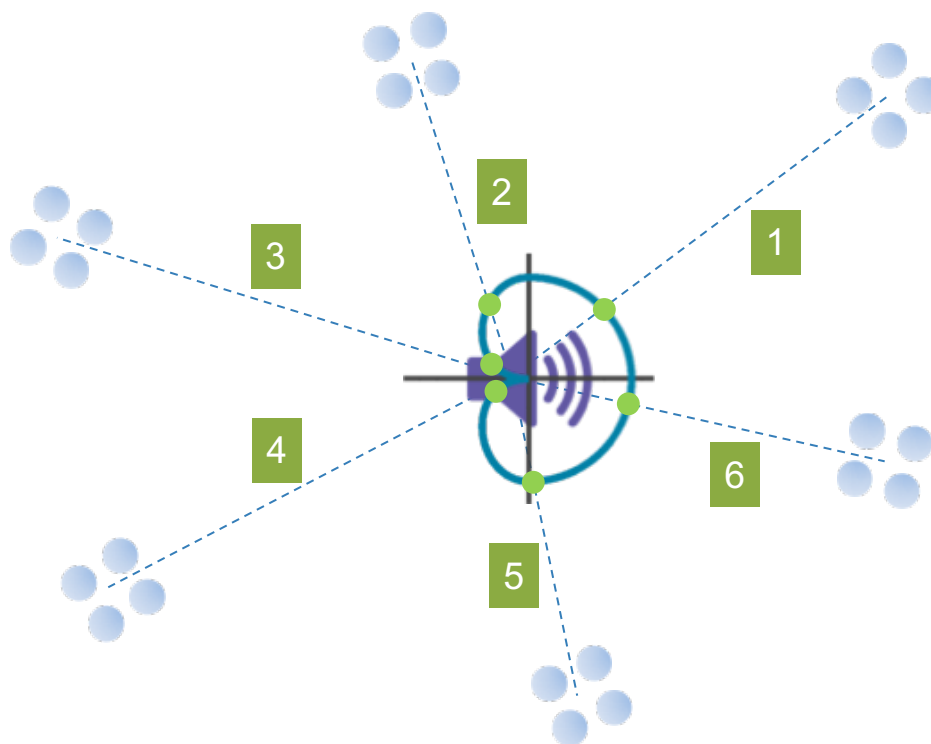


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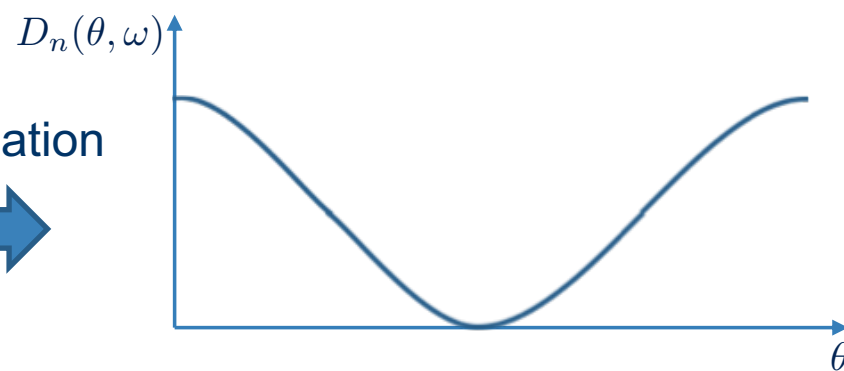




Radiance Pattern Estimation (1/3)



Interpolation





Estimate of the Radiance Pattern at the microphone arrays

Assumption: sources in the far field w.r.t to each array



Microphones within each array “see” the n th source under the same angle

LCMV beamformer in order to estimate the radiance pattern value

$$\mathbf{h}_n^{(a)} = \underset{\mathbf{h}}{\operatorname{argmin}} \mathbf{h}^H \mathbf{h} \quad \text{s.t.} \quad \mathbf{h}^H \hat{\mathbf{G}}^{(a)} = \mathbf{c}$$



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↓
Green's functions of the sources obtained after the localization

$$\hat{D}_n \left(\theta_n^{(a)} \right) = \mathbf{h}_n^{(a)H} \mathbf{p}^{(a)}$$



Interpolation of the Radiance Pattern

Idea: adopt a **model** to reconstruct the whole radiance pattern from the estimates at a **limited set of angles**

$$D_n(\theta_n^{(a)}) = \sum_{l=0}^{L-1} w_{n,l} \cos(l\theta_n^{(a)}) + r_{n,l} \sin(l\theta_n^{(a)})$$



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Estimate of the nth
source Radiation
Pattern Coefficients

Constrain the
Radiance Pattern to
be positive



We exploit the estimate of the **radiance patterns** and the **source positions** to **design an informed spatial filter**



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$$\mathbf{u}_n^* = \underset{\mathbf{u}}{\operatorname{argmin}} \mathbf{u}^H \mathbf{u} \quad \text{s.t.} \quad \mathbf{u}^H \mathbf{Y} = \mathbf{d}$$



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$$\mathbf{Y} = \left[\underbrace{\left(\hat{\mathbf{G}}^{(0)} \otimes \hat{\mathbf{D}}^{(0)} \right)^T, \dots, \left(\hat{\mathbf{G}}^{(A-1)} \otimes \hat{\mathbf{D}}^{(A-1)} \right)^T}_{\substack{\text{Estimates of source} \\ \text{Radiation Patterns and} \\ \text{Positions}}} \right]^T$$



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Estimates of source
Radiation Patterns and
Positions

$$\hat{S}_n(\omega) = (\mathbf{u}_n^*)^H \mathbf{p}$$



Once the **parameters are estimated** as described in the previous sections, the **sound field at the virtual microphone location can be computed through the parametric model**

$$\hat{P}(\mathbf{r}_{VM}, \omega) = \sum_{n=0}^{N-1} \hat{D}_n(\theta_{VM,n}, \omega) g(\mathbf{r}_{VM}, \hat{\mathbf{r}}'_n, \omega) \hat{S}_n(\omega)$$



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Virtual Miking signal estimate

Radiation Pattern estimate

Source Position estimate

Source Signal estimate

Metrics

Localization Error

$$LE(\mathbf{r}') = \frac{1}{N} \sum_{n=0}^{N-1} \|\hat{\mathbf{r}}'_n - \mathbf{r}'_n\|$$

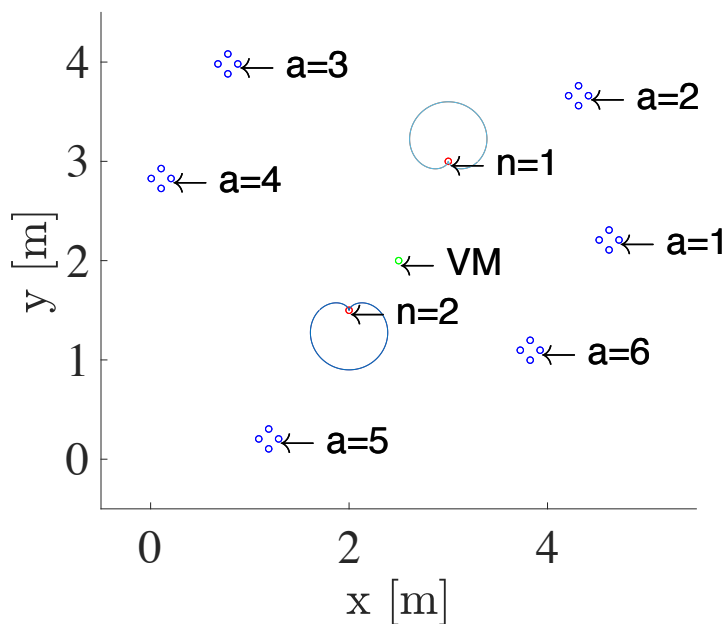
Directivity Error

$$DE_n = \frac{1}{F} \sum_{f=0}^{F-1} \frac{1}{I} \sum_{i=0}^{I-1} (\hat{D}_n(\theta_i, \omega_f) - D_n(\theta_i, \omega_f))^2$$

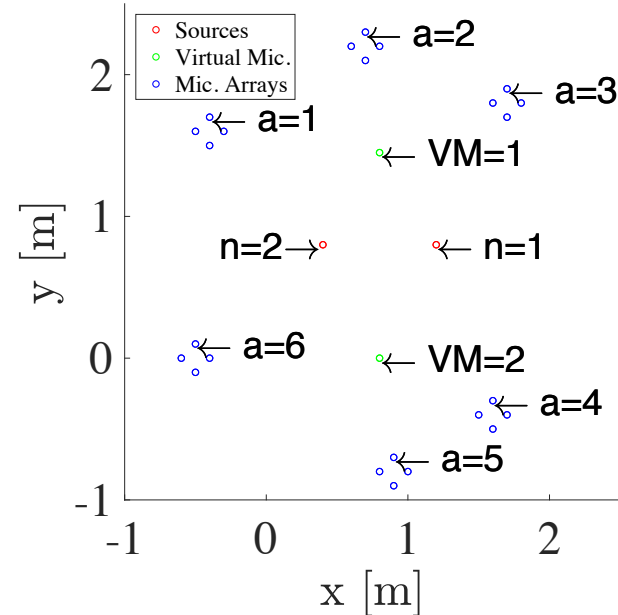
Synthesized Signal Error

$$SSE_{VM} = \sqrt{\frac{1}{T} \sum_{t=0}^T (x_{VM}(t) - x_{ref}(t))^2}$$

Simulations Setup



Experiments Setup



Simulations Results

Simulations					
Sources	Orientation	Source Signal	LE [m]	DE	SSE
1	0°	White Noise	1.12×10^{-2}	6.61×10^{-3}	6.74×10^{-4}
2	0°		7.40×10^{-3}	2.66×10^{-2}	5.41×10^{-4}
1, 2	0°, 0°	Speech	3.50×10^{-2}	$8.56 \times 10^{-2}, 5.24 \times 10^{-2}$	2.42×10^{-2}
1	-90°		1.10×10^{-2}	3.57×10^{-3}	8.18×10^{-4}
2	90°		8.00×10^{-3}	1.12×10^{-2}	7.72×10^{-4}
1, 2	-90°, 0°		4.39×10^{-2}	$8.97 \times 10^{-2}, 9.94 \times 10^{-2}$	5.75×10^{-3}

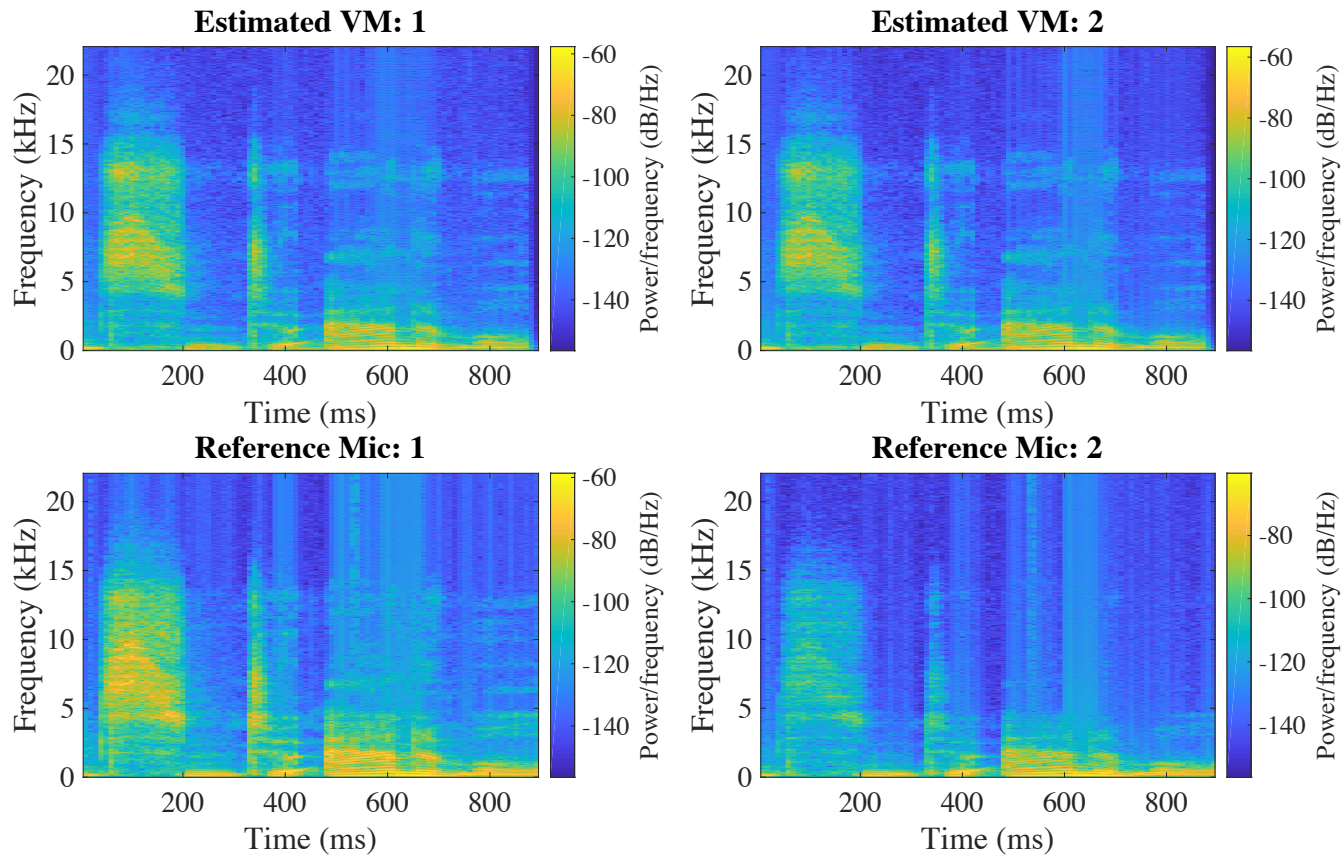
Experiments Results

Experiments					
Sources	Orientation	Source Signal	LE [m]	SSE	
1	90°	White Noise	3.54×10^{-2}	$1.00 \times 10^{-2}, 2.84 \times 10^{-3}$	
2	-90°		5.33×10^{-2}	$3.44 \times 10^{-3}, 9.30 \times 10^{-3}$	
1, 2	90°, -90°	Speech	6.40×10^{-2}	$1.32 \times 10^{-2}, 1.29 \times 10^{-2}$	
1	90°		6.03×10^{-2}	$5.83 \times 10^{-3}, 4.26 \times 10^{-3}$	
2	90°		3.18×10^{-2}	$9.72 \times 10^{-3}, 6.98 \times 10^{-3}$	
1, 2	90°, 90°		5.53×10^{-2}	$1.34 \times 10^{-2}, 1.00 \times 10^{-2}$	



Experiments Results

Example of a spectrogram of the VM signal w.r.t to the two reference microphones





□ We proposed a parametric technique for sound field reconstruction in the presence of multiple sources. The parameters are

- Sources locations;
- Radiance pattern of the sources;
- Sources dry signals

and they are estimated directly from the audio data.

Results demonstrate that the approach is promising even with some errors in the estimation of the parameters.



- ❑ Include a directional sensitivity pattern of the virtual microphone.
- ❑ Extend the presented technique also to reverberant environments:
 - ❑ Estimate the parameters robustly with respect to the reverberation
 - ❑ Synthesize the sound field including the direct and diffuse components of the sound field.