

Multi-channel Source Separation 1

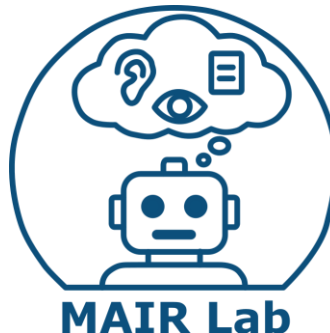
안인규 (Inkyu An)

Speech And Audio Recognition
(오디오 음성인식)

<https://mairlab-km.github.io/>

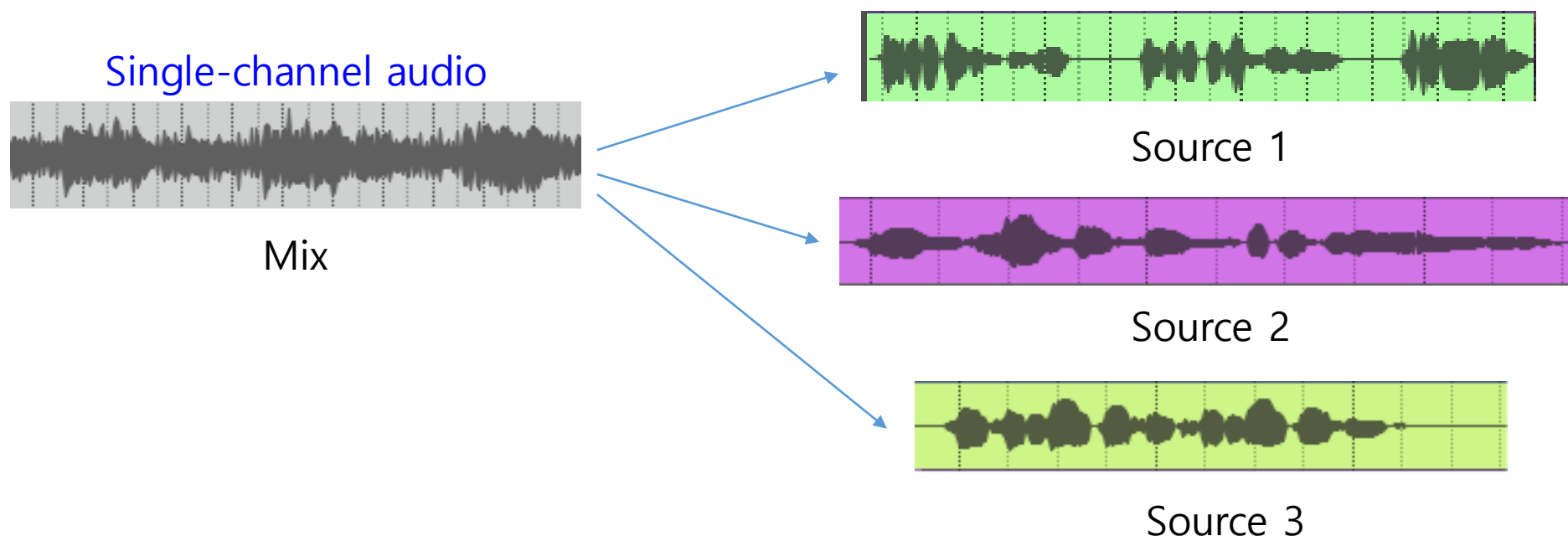


This lecture material refers to
https://github.com/yandexdataschool/speech_course?tab=readme-ov-file and
<https://github.com/markovka17/dla>



Blind Source Separation

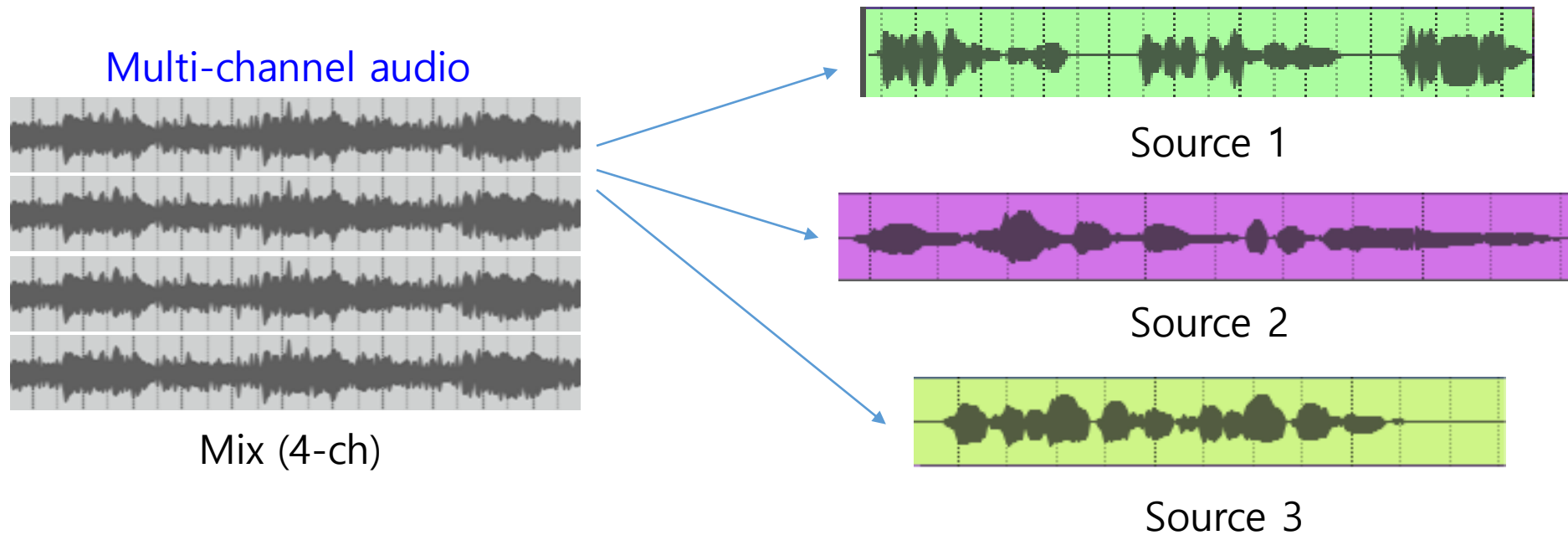
- **Goal:** extract K sources from the noisy mixture w/o (or with very little) information about the mixing process



- We know denoising and specific guided separation, how can we apply DL here?

Multi-channel Source Separation

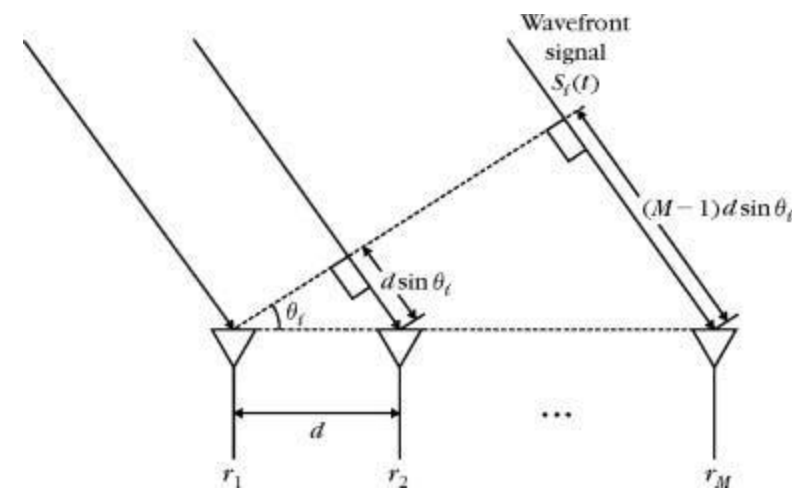
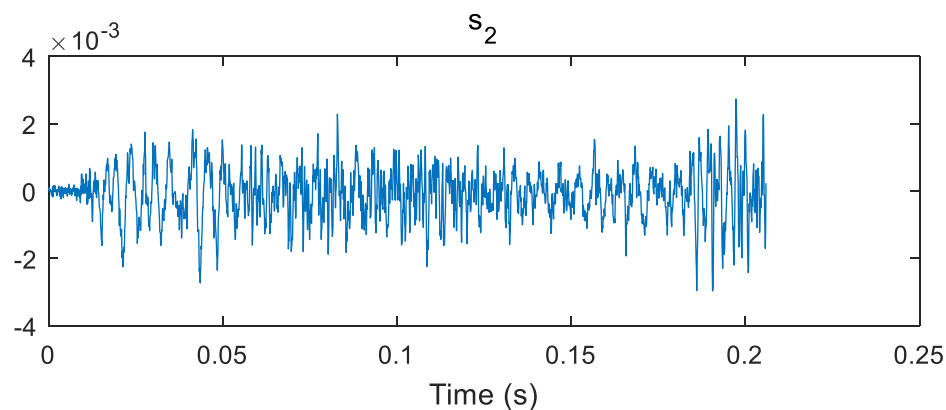
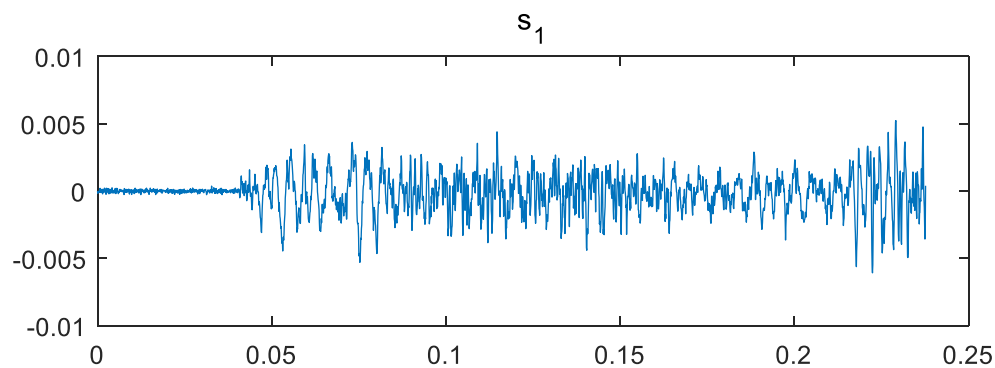
- **Goal:** extract K sources from the noisy mixture of multi-channel audio



- How can we measure the multi-channel audio signal?

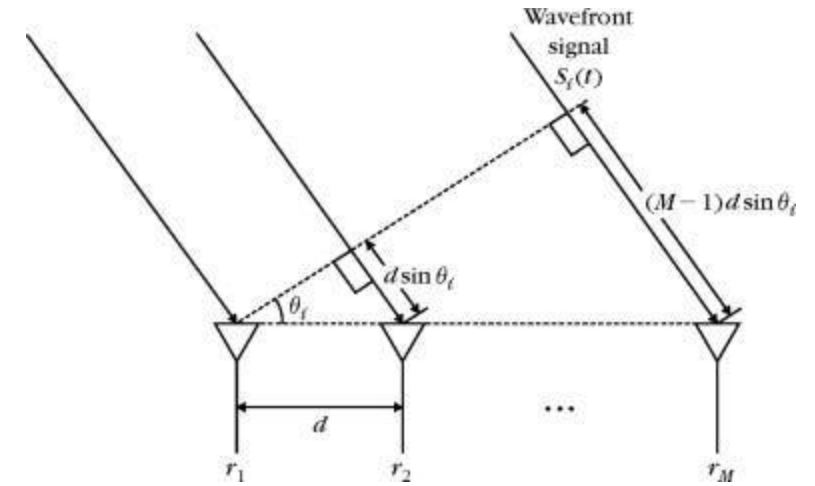
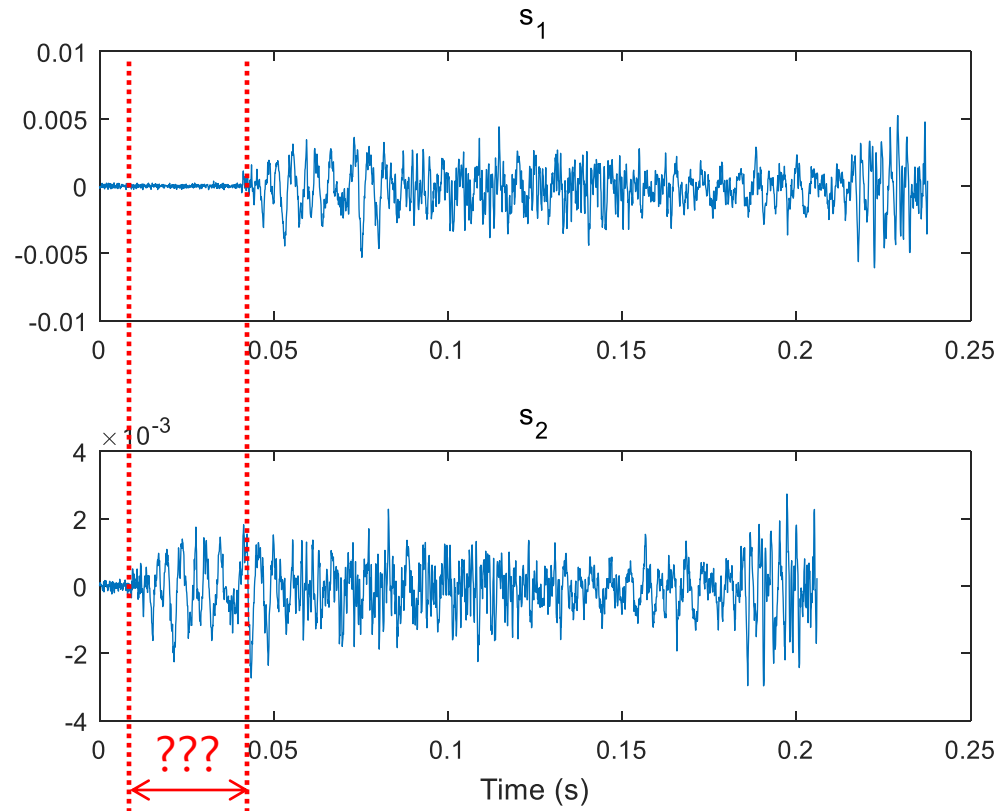
Multi-channel Source Separation

- Microphone Array
 - Consider the Uniform Linear Array
 - What's different about the audio collected at each microphone?



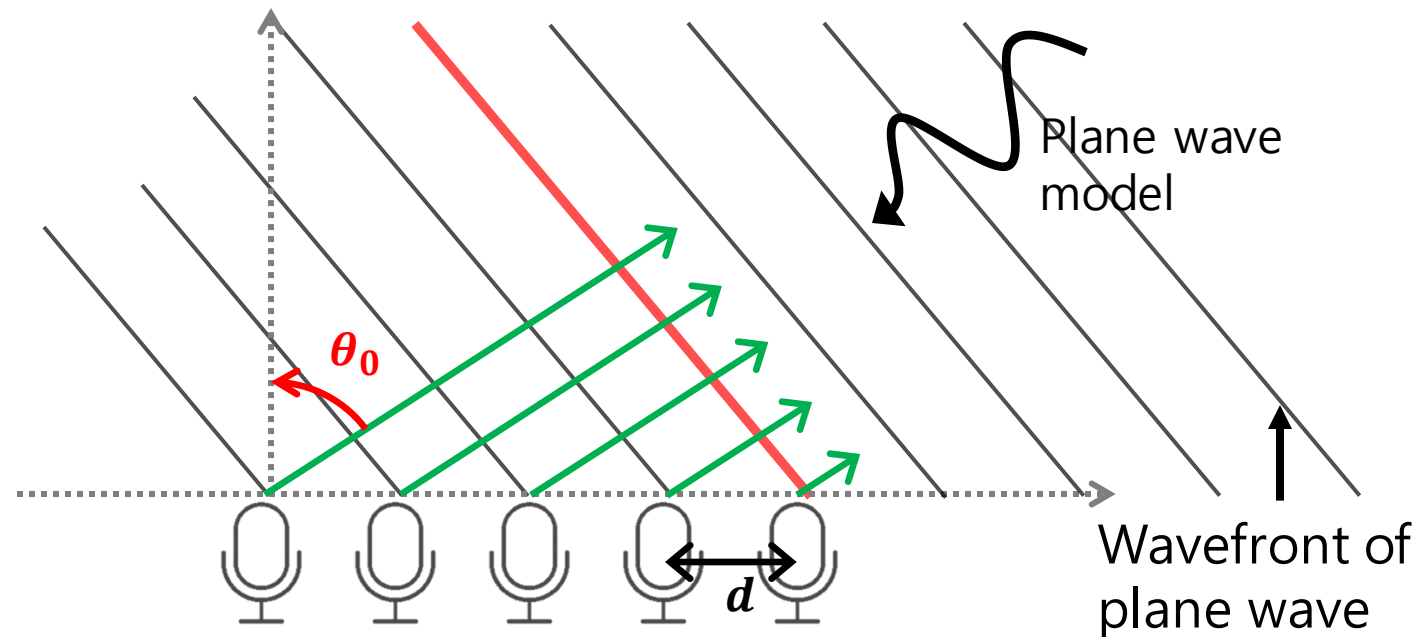
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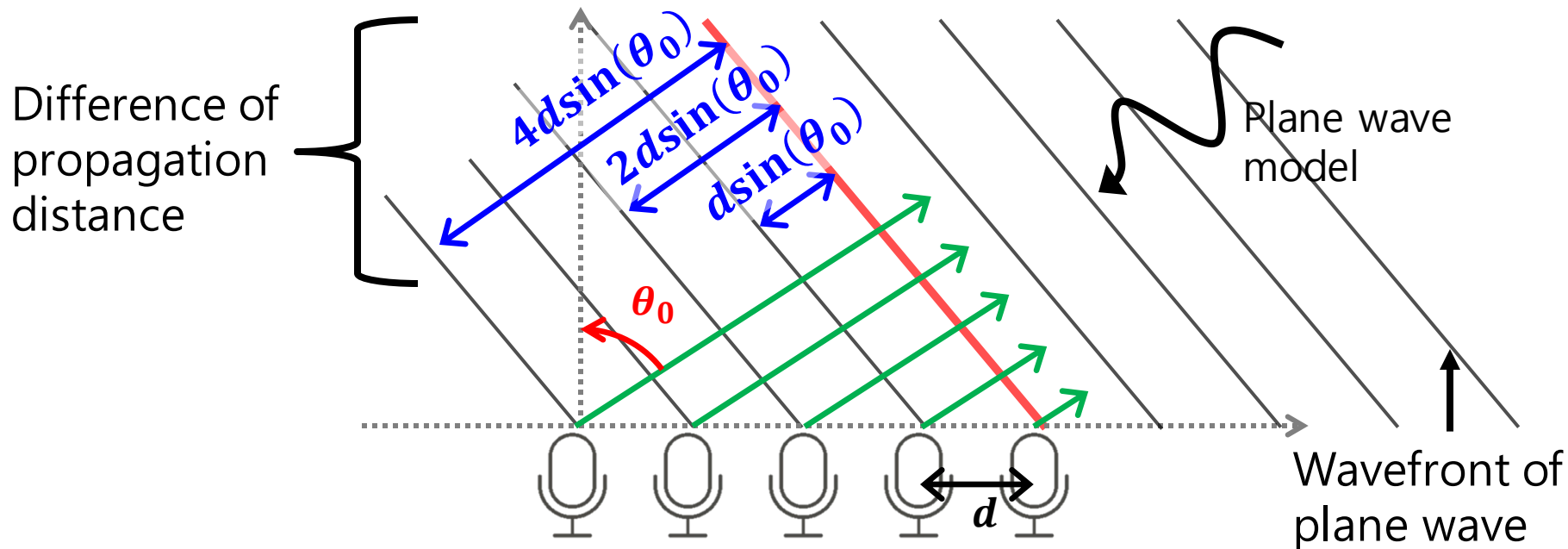
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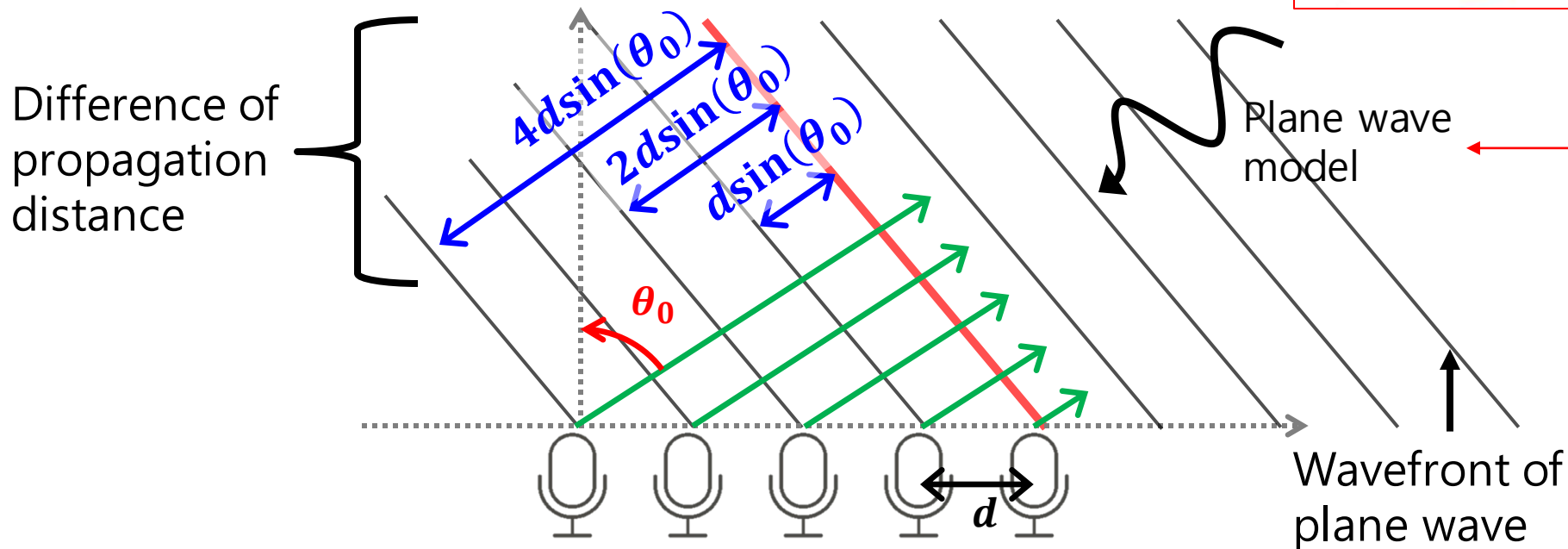
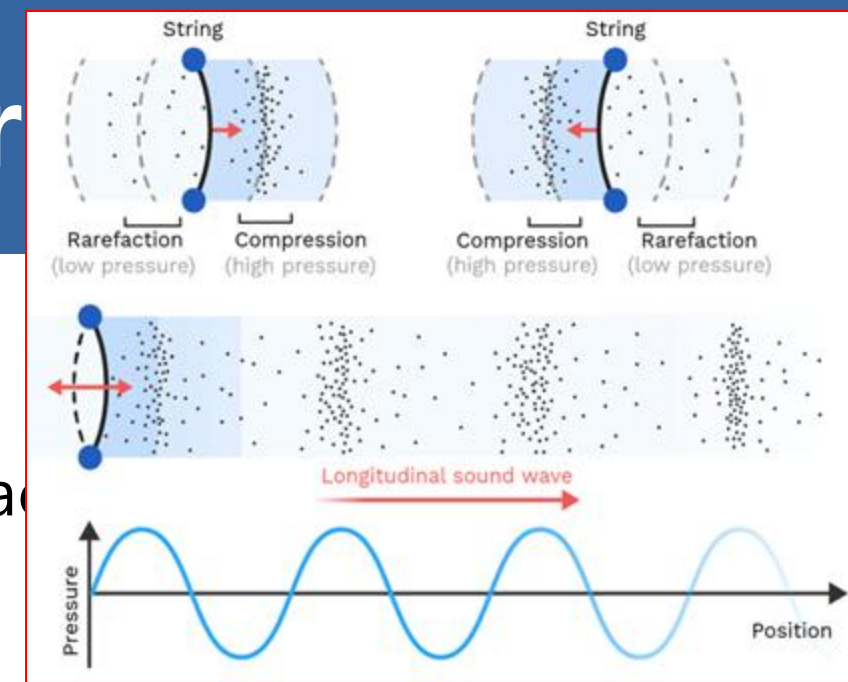
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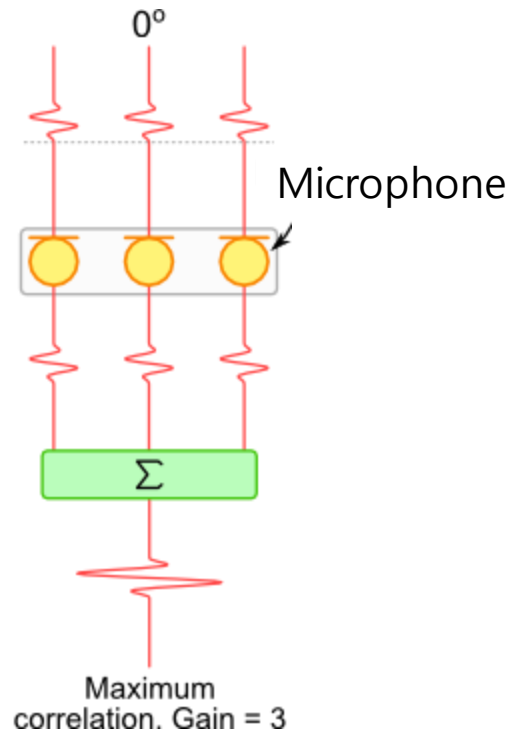
- Microphone Array
 - Is it possible to extract only the audio signal coming from a specific direction? → **Beamforming**
 - What is the beamforming algorithm?
 - Focusing a specific direction, θ , by using beamforming weights (spatial filter)

$$y(\theta, t) = w(\theta, -t)^H * x(t) \quad \longleftrightarrow \quad y(\theta, \omega) = w(\theta, \omega)^H x(\omega)$$

↑
Convolution FFT & iFFT Beamforming output Weight (Spatial filter) Measured pressure signals

Multi-channel Source Separation

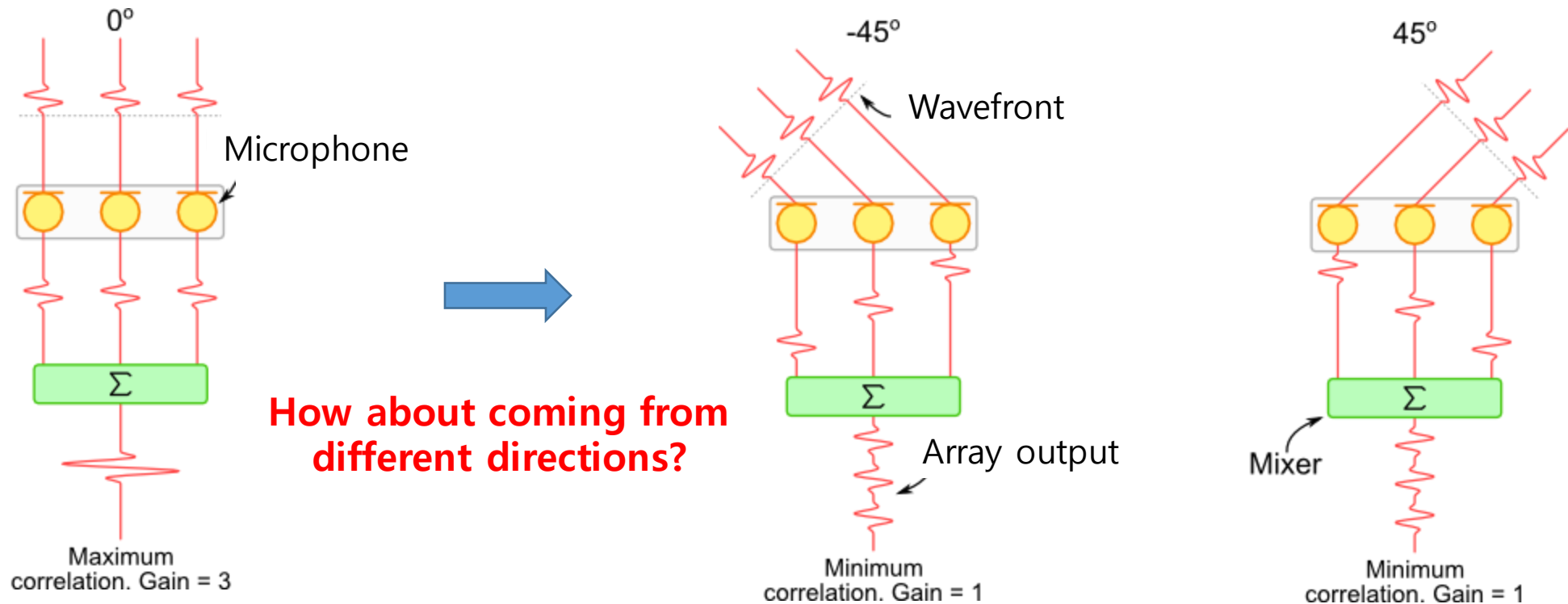
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 - Is it possible to extract only the audio signal coming from a specific direction? → **Beamforming**
 - A simple method? → **Add all the multi-channel signals together**



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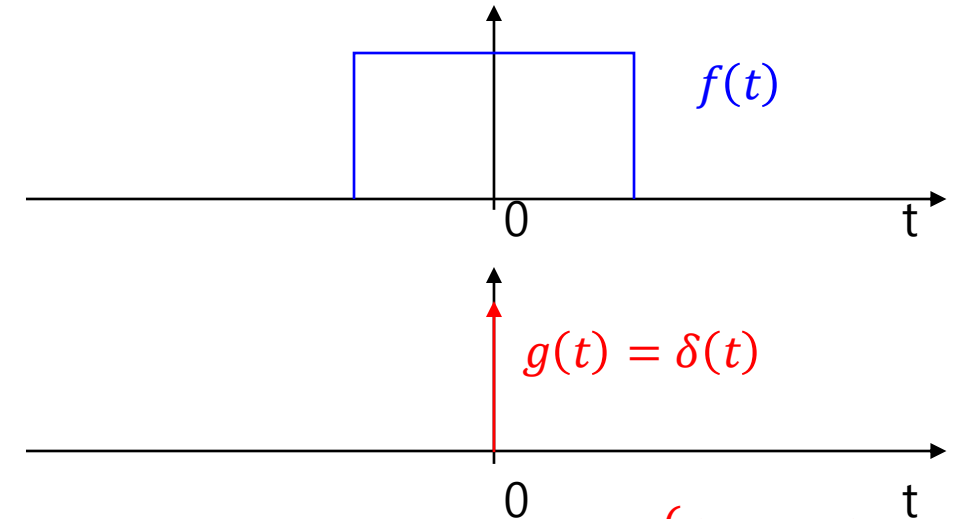
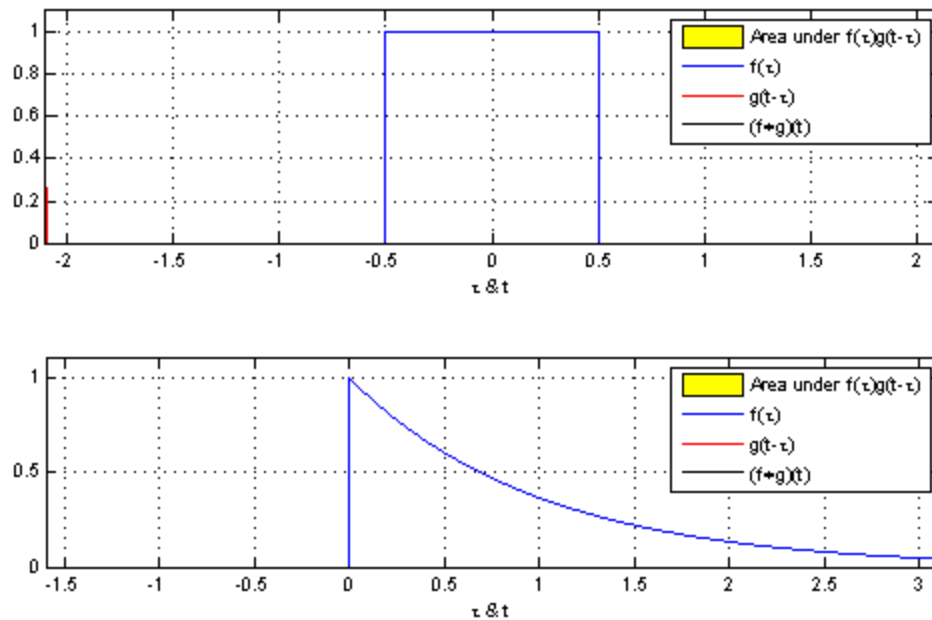
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Multi-channel Source Separation

- Microphone Array

- Convolution: $(f * g)(t) := \int_{-\infty}^{\infty} f(\tau)g(t - \tau) d\tau.$

Example of Convolution



Dirac delta function $\delta(t) = \begin{cases} 0, & x \neq 0 \\ \infty, & x = 0 \end{cases}$

$$\int_{-\infty}^{\infty} \delta(t) dt = 1$$

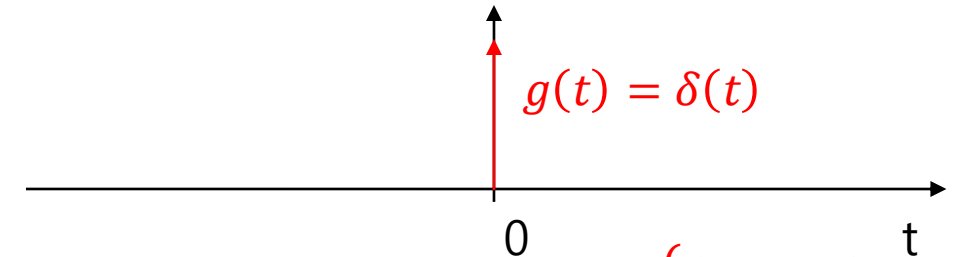
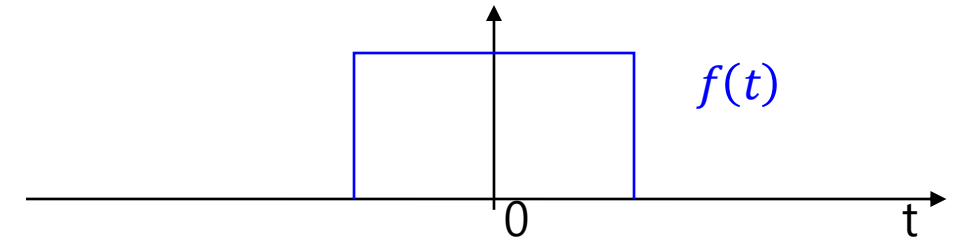
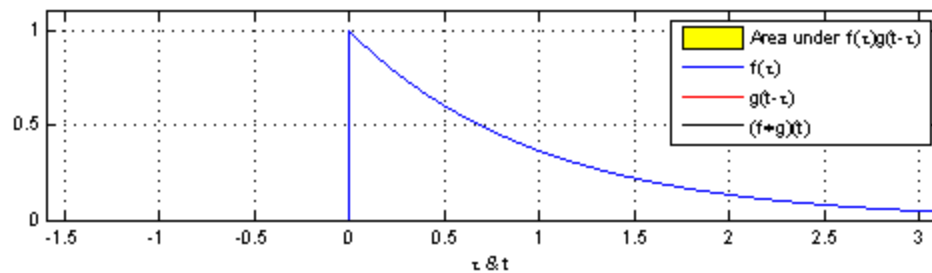
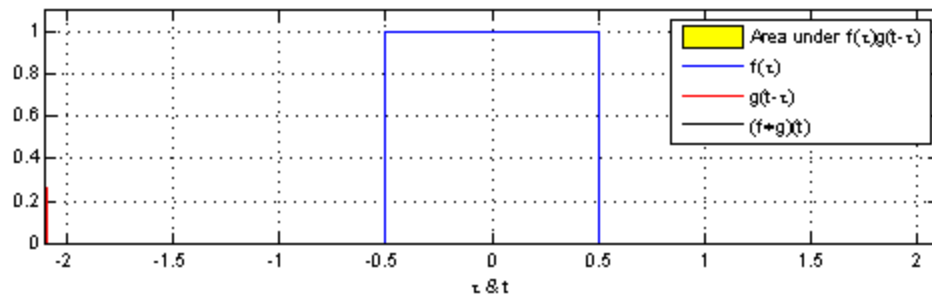
$(f * g)(t)?$

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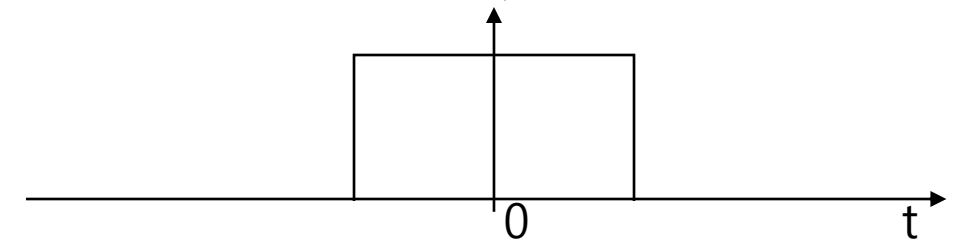


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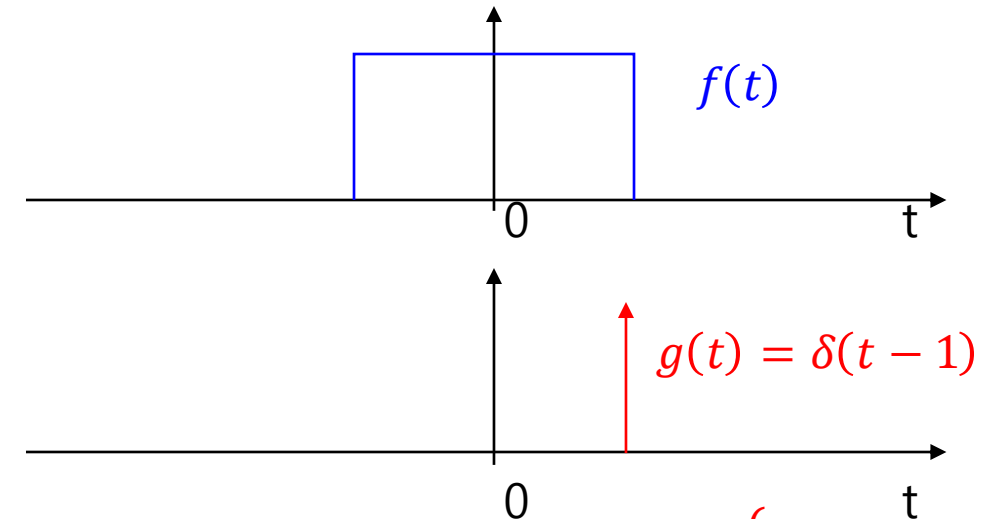
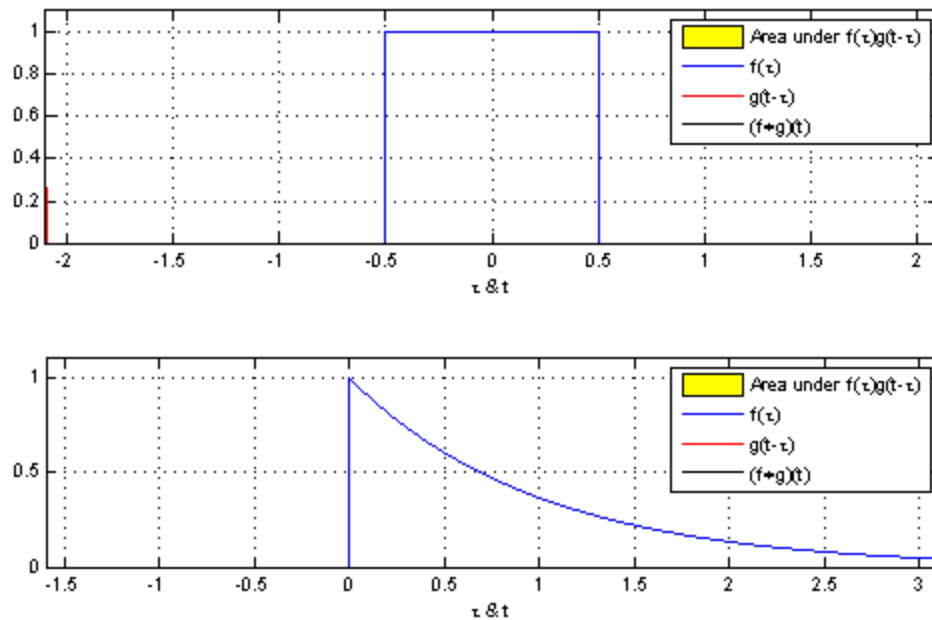


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Dirac delta function $\delta(t - 1) = \begin{cases} 0, & x \neq 1 \\ \infty, & x = 1 \end{cases}$

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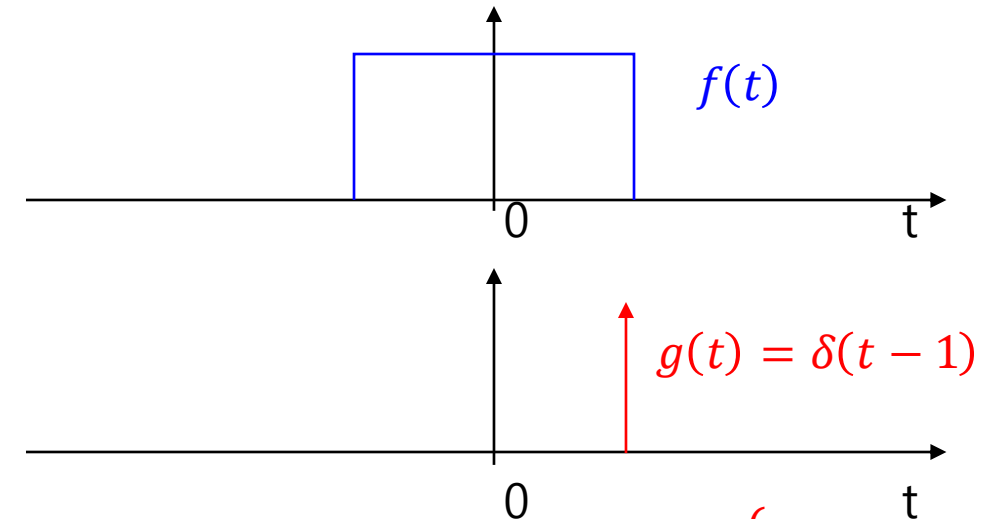
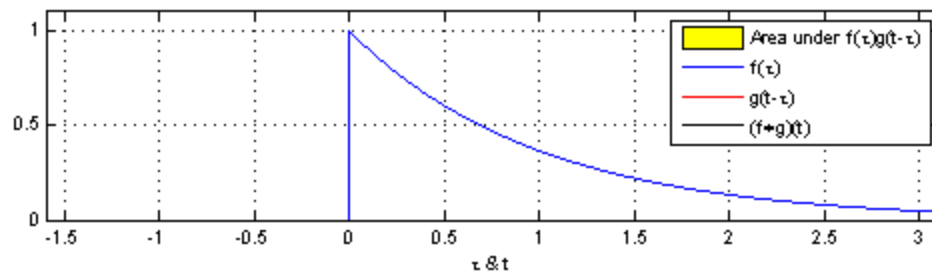
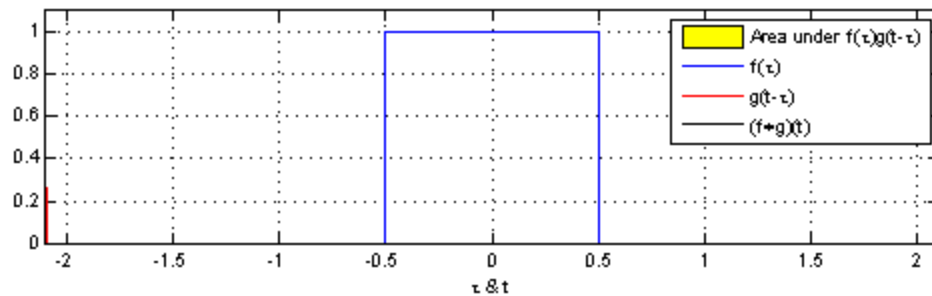
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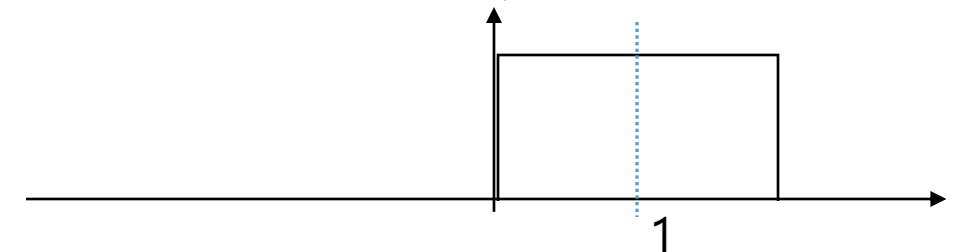
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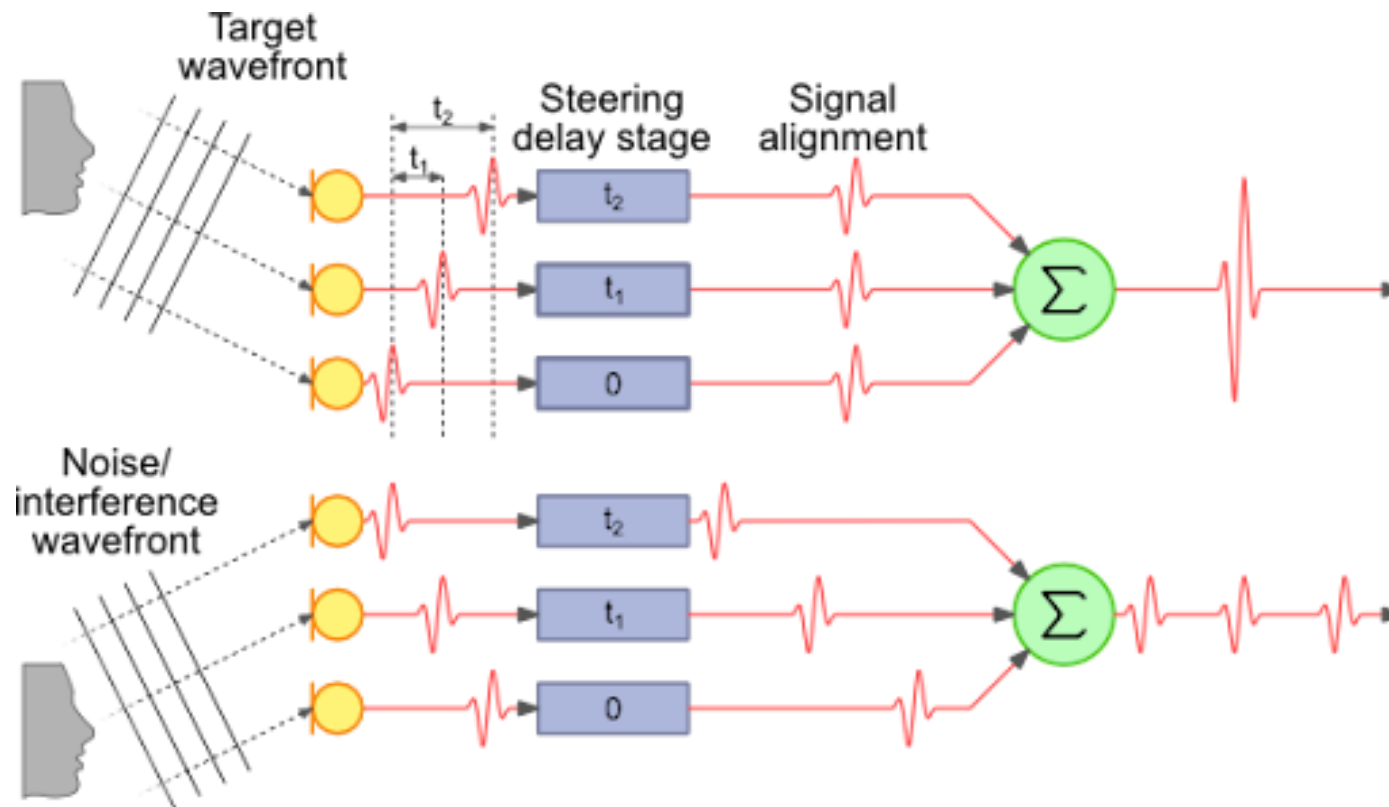


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Multi-channel Source Separation

- Delay-and-Sum Beamformer
 - Consider the propagation delays!

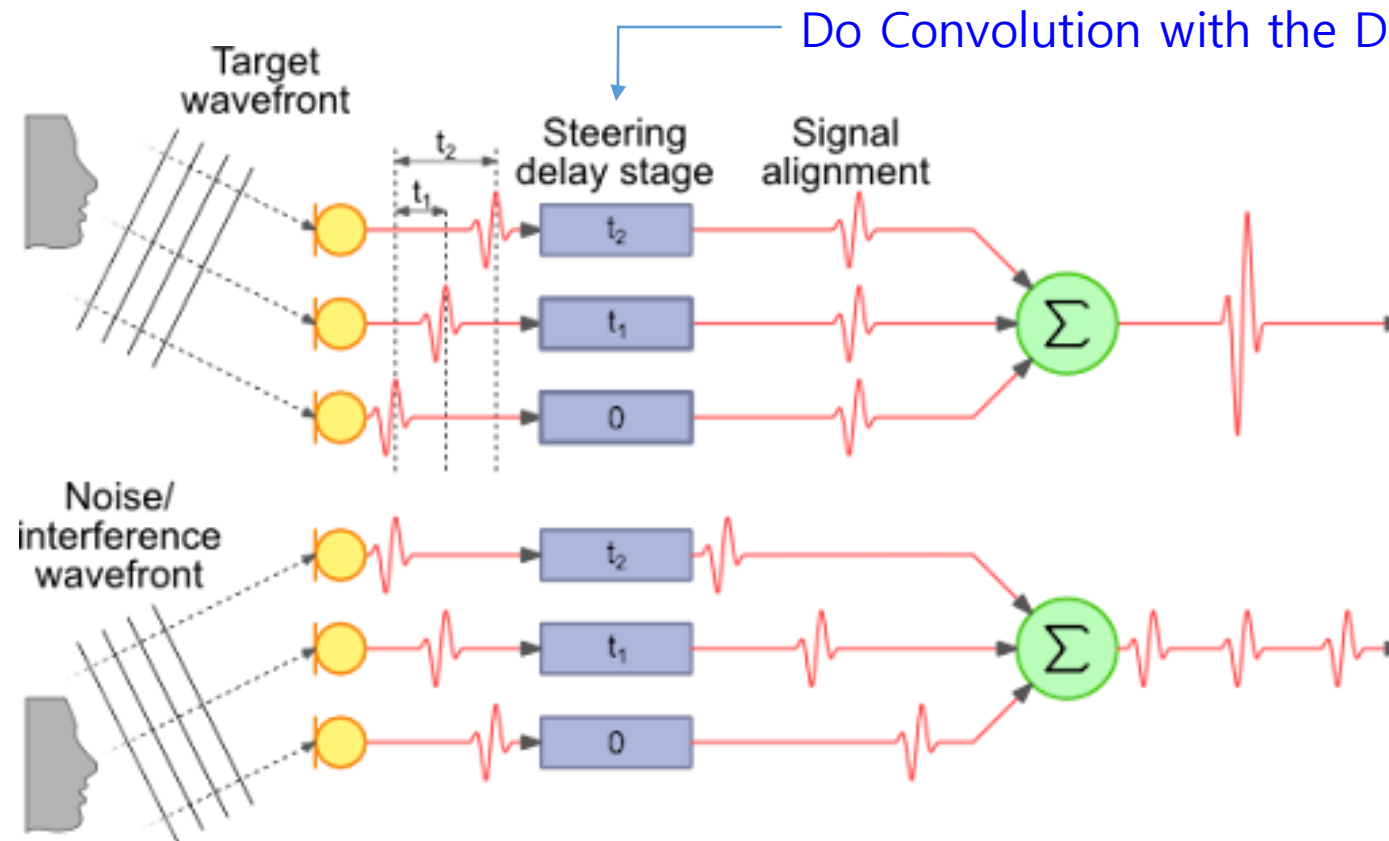


Multi-channel Source Separation

- Delay-and-Sum Beamformer
 - Consider the propagation delays!

$$g(t) = \delta(t + t_1)$$

Do Convolution with the Dirac delta function!



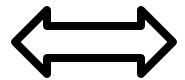
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$$y(\theta, t) = w(\theta, -t)^H * x(t)$$

Convolution

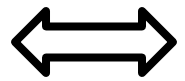


FFT & iFFT

$$y(\theta, \omega) = w(\theta, \omega)^H x(\omega)$$

Beamforming output Weight (Spatial filter) Measured pressure signals

$$\delta(t + t_1)$$



FFT & iFFT

$$e^{+j\omega t_1}$$



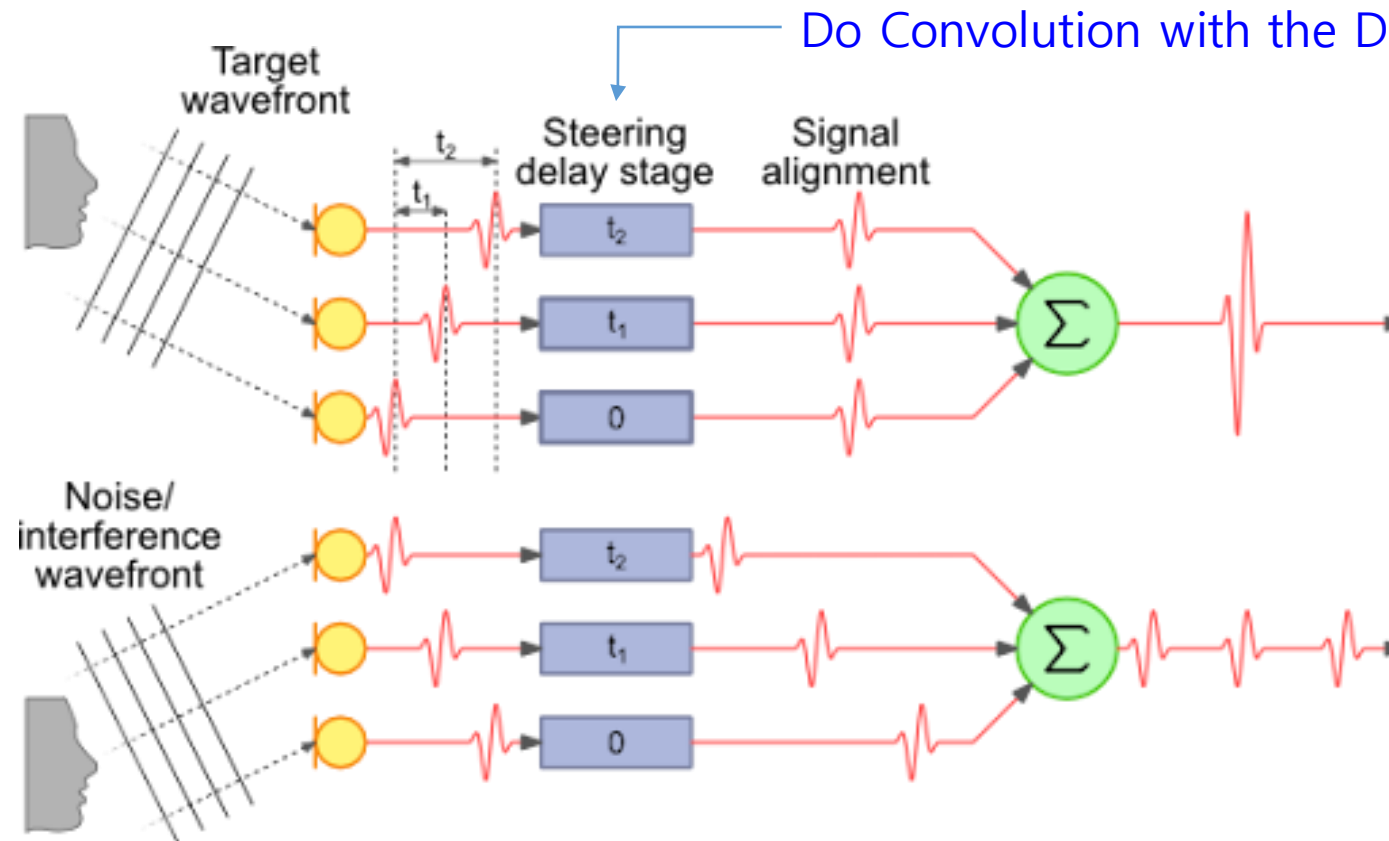
$$y(\theta, \omega) = \begin{bmatrix} e^{+j\omega t_1} \\ \dots \\ e^{+j\omega t_c} \end{bmatrix}^H \cdot x(\omega)$$

Multi-channel Source Separation

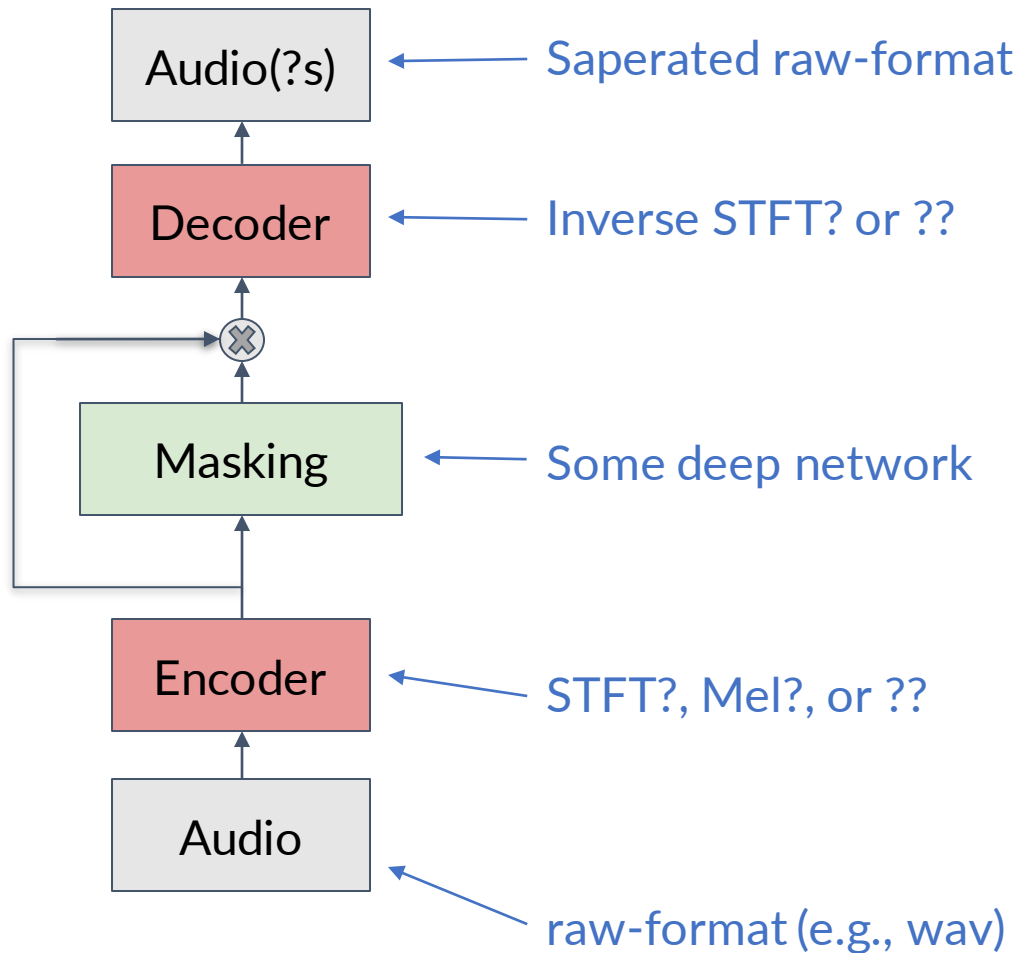
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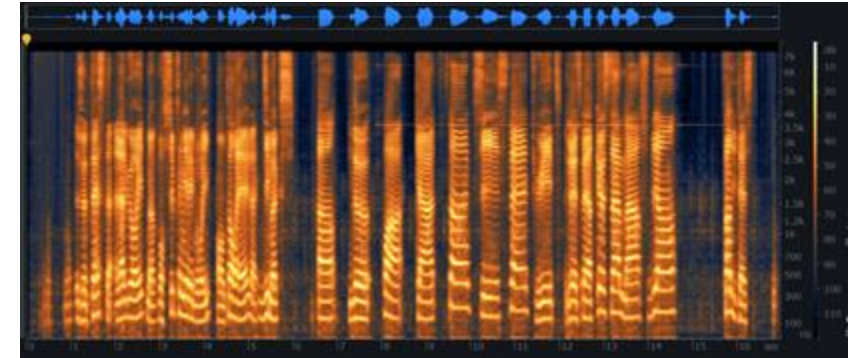
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Encoder-Separation-Decoder (ESD)

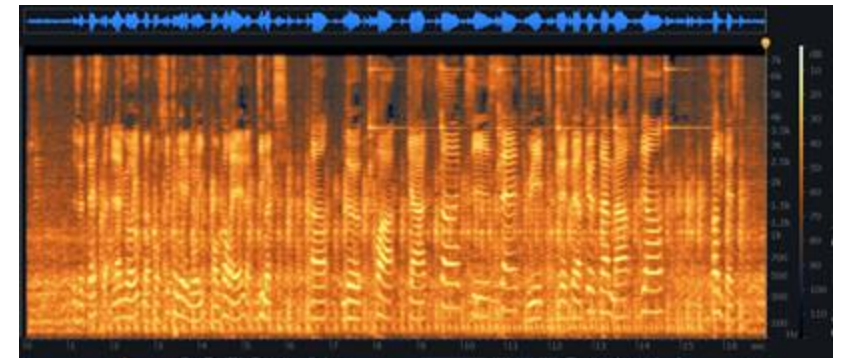


Denoised
Alex

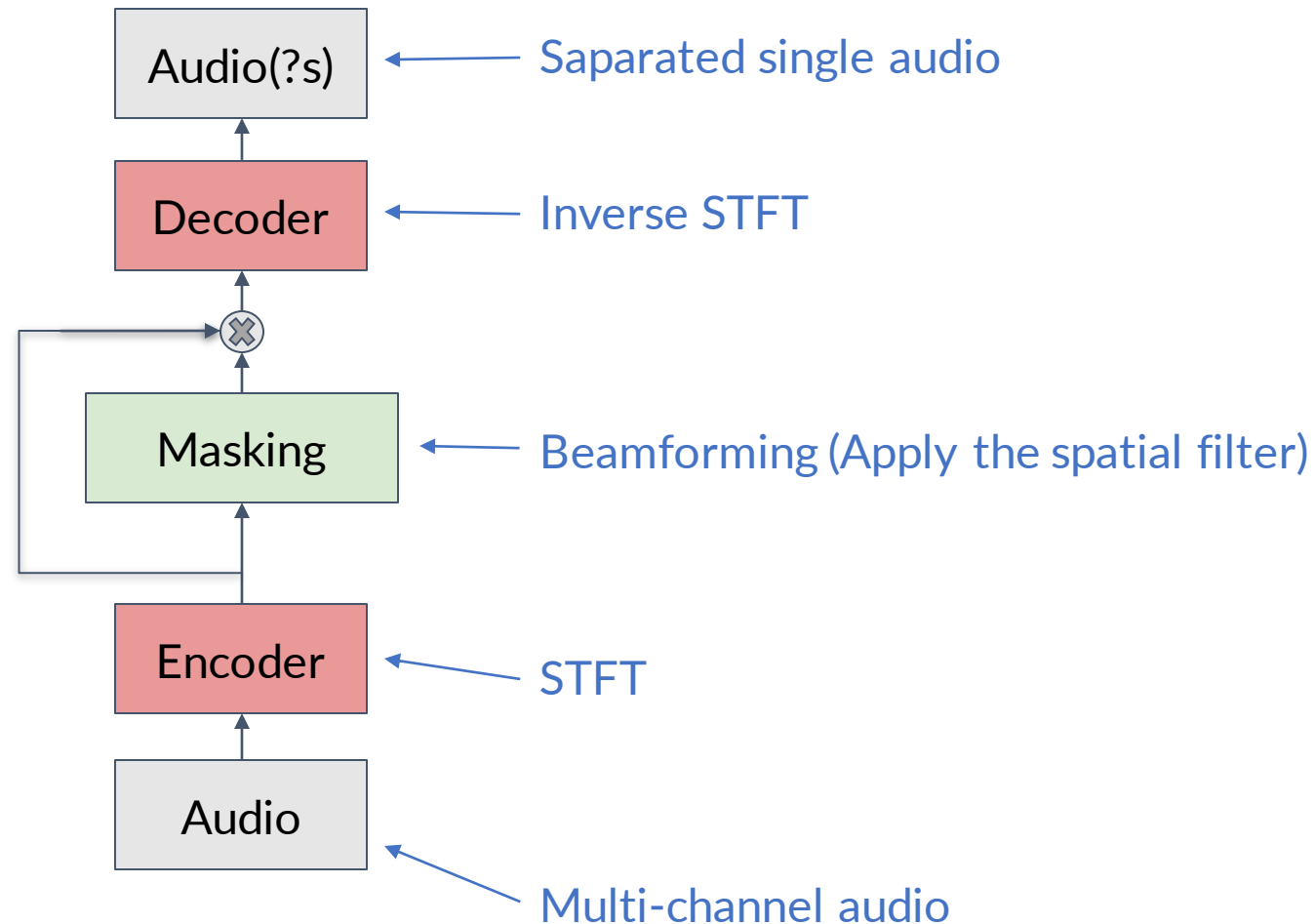


Denoising

Alex
&
noise

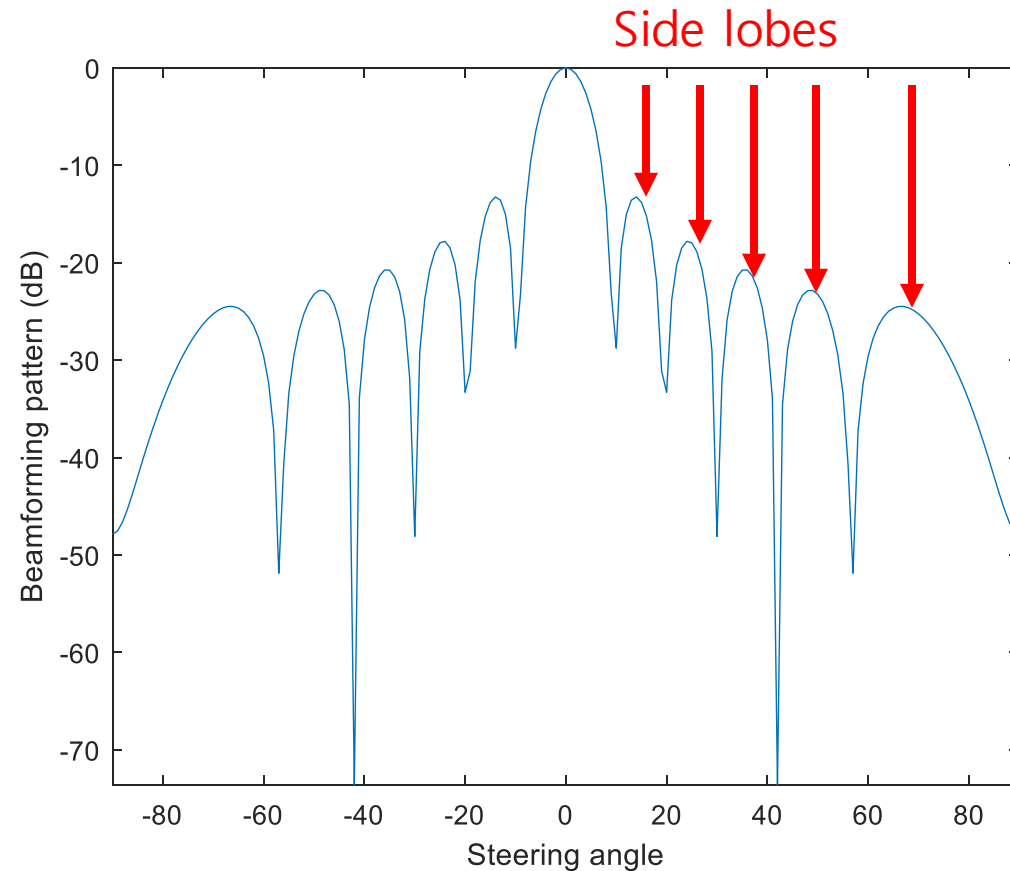


Multi-Channel Encoder-Separation-Decoder (ESD)



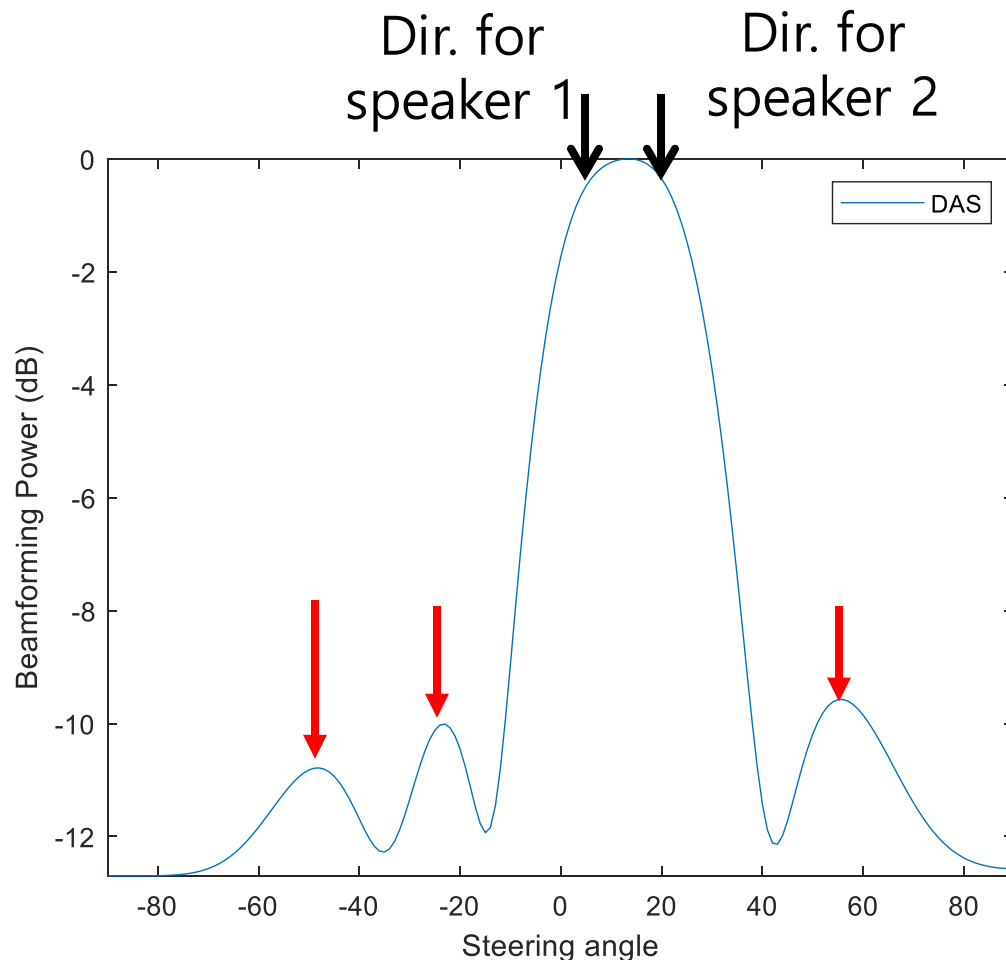
Multi-channel Source Separation

- The problem of the Delay-and-Sum Beamformer
 - The beamforming pattern of DAS beamformer is not sharp enough!

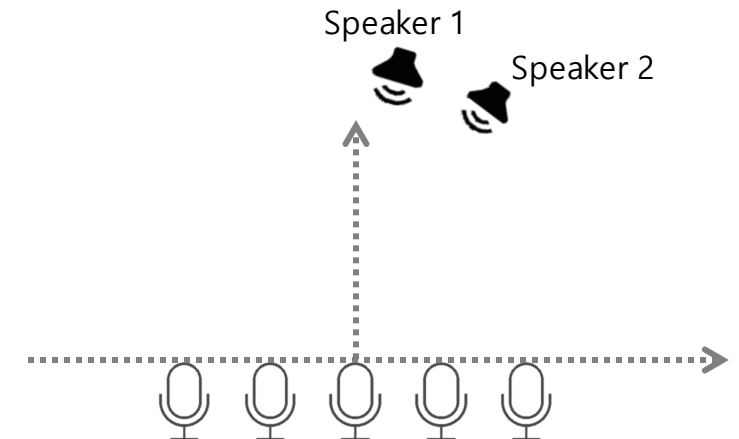


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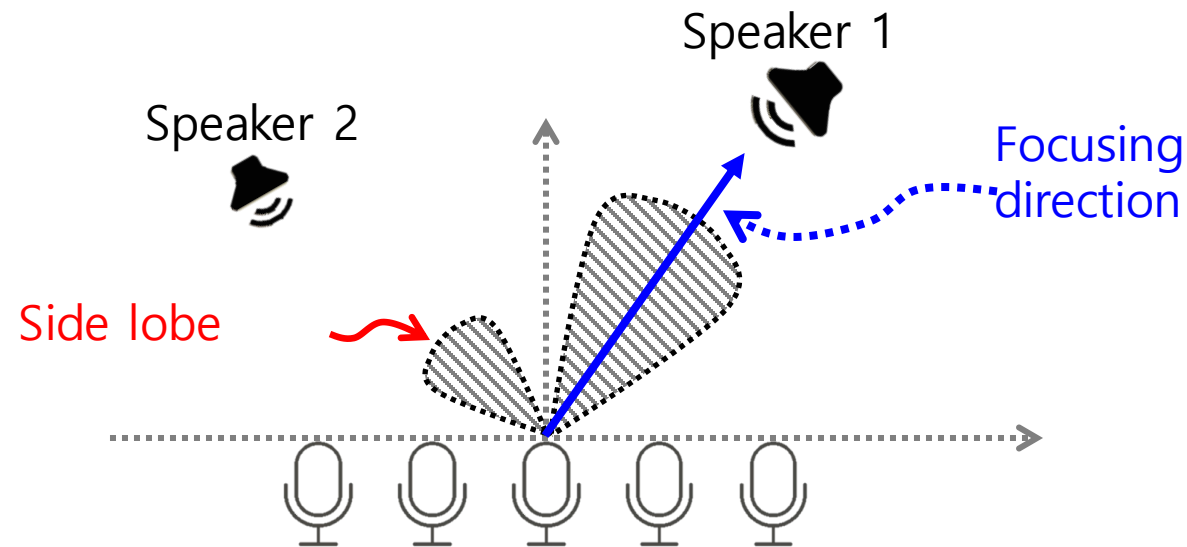


Delay-and-Sum
beamformer



Multi-channel Source Separation

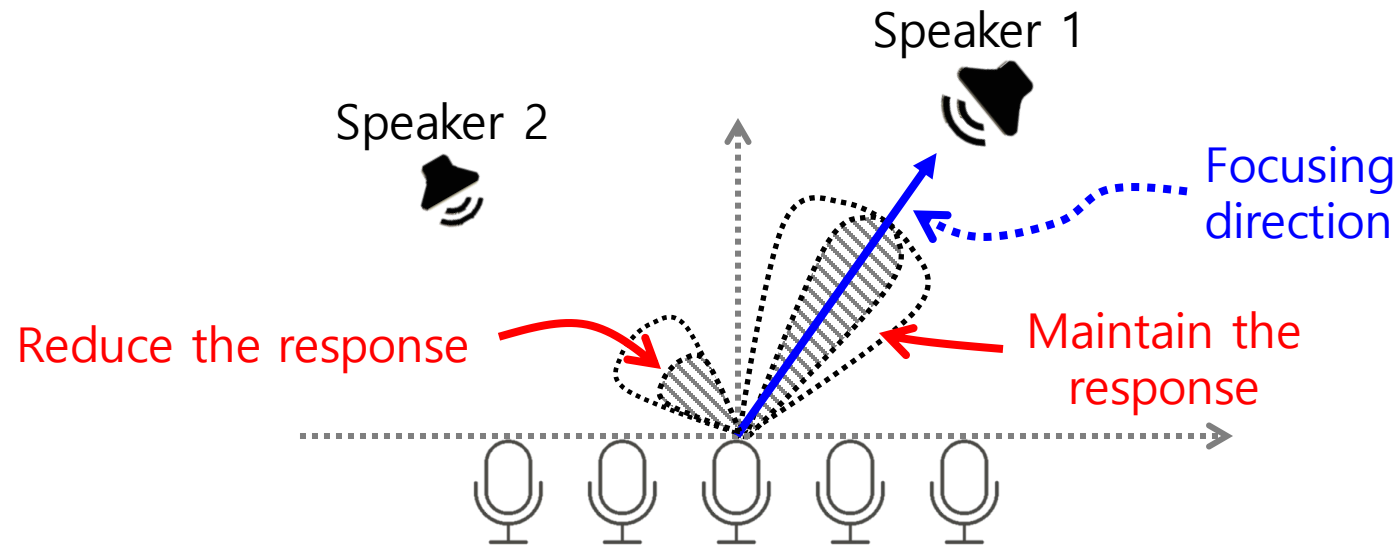
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- How to reduce the response for the speaker 2?  **MVDR beamformer!**
when focusing on the speaker 1...

Multi-channel Source Separation

- MVDR beamformer
 - Design a spatial filter w that only responds to a specific direction



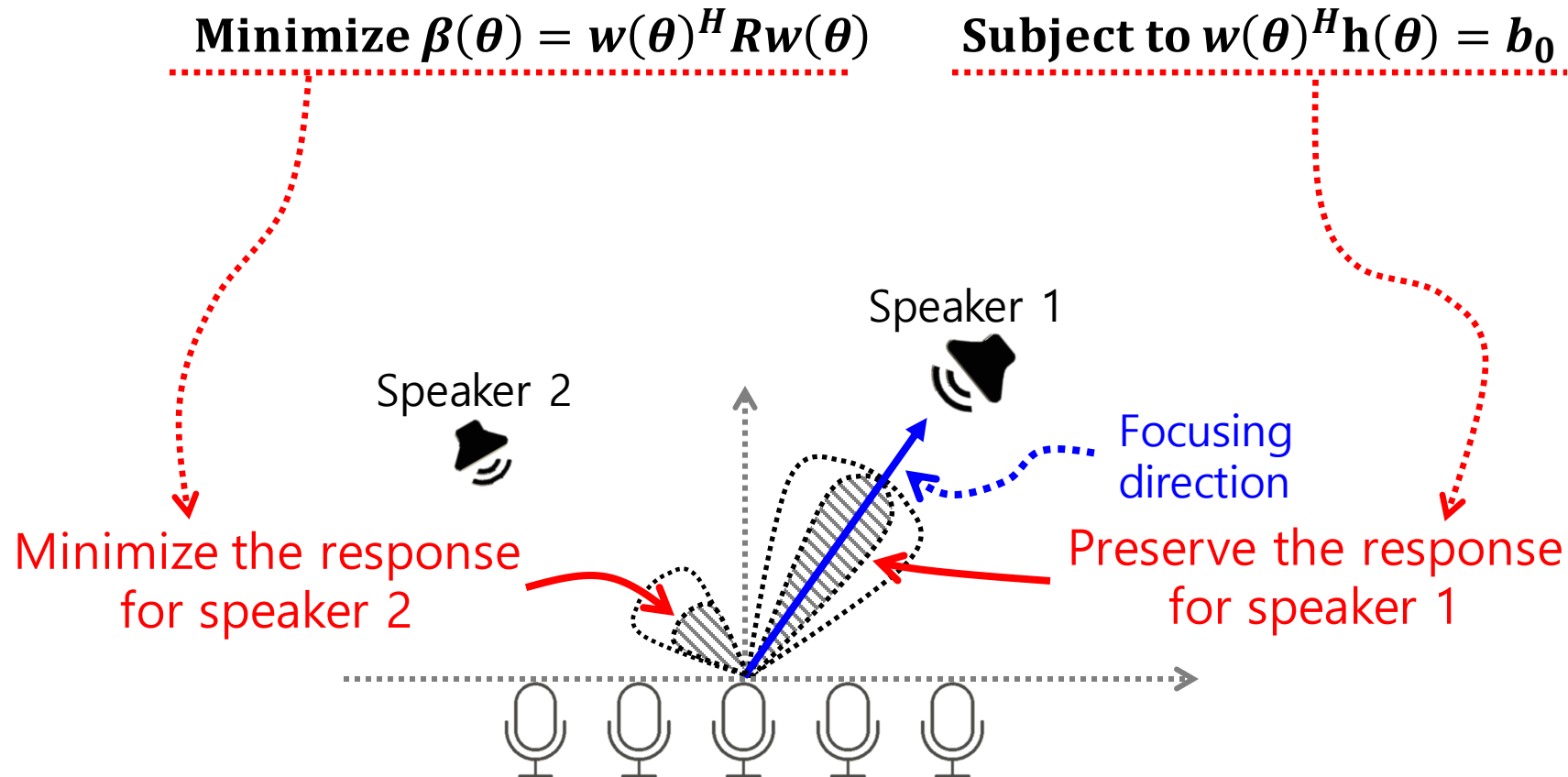
- Design a spatial filter w that minimizes the total beamforming power but maintain the response in the aiming angle

$$\text{Minimize } \beta = w^H E[x(\omega)x(\omega)^H] w \quad \text{Subject to } w^H \mathbf{h} = b_0$$

Plane wave model 

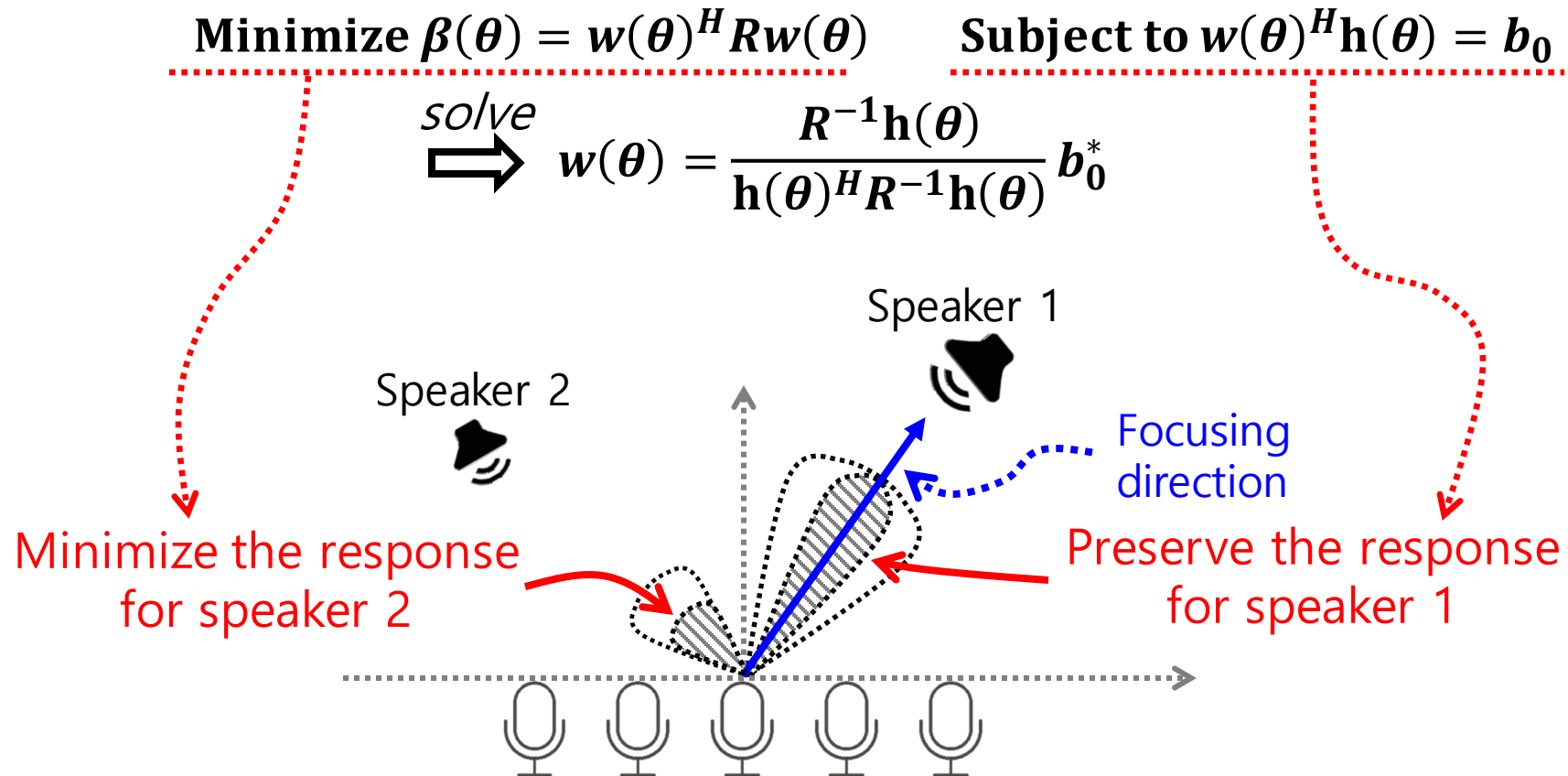
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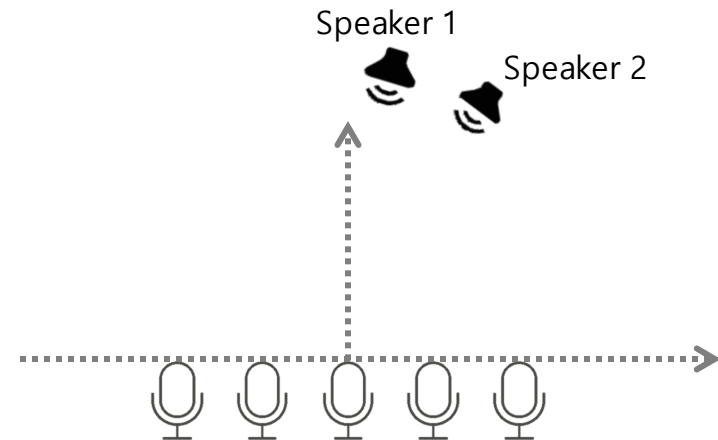
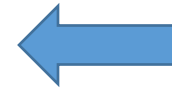
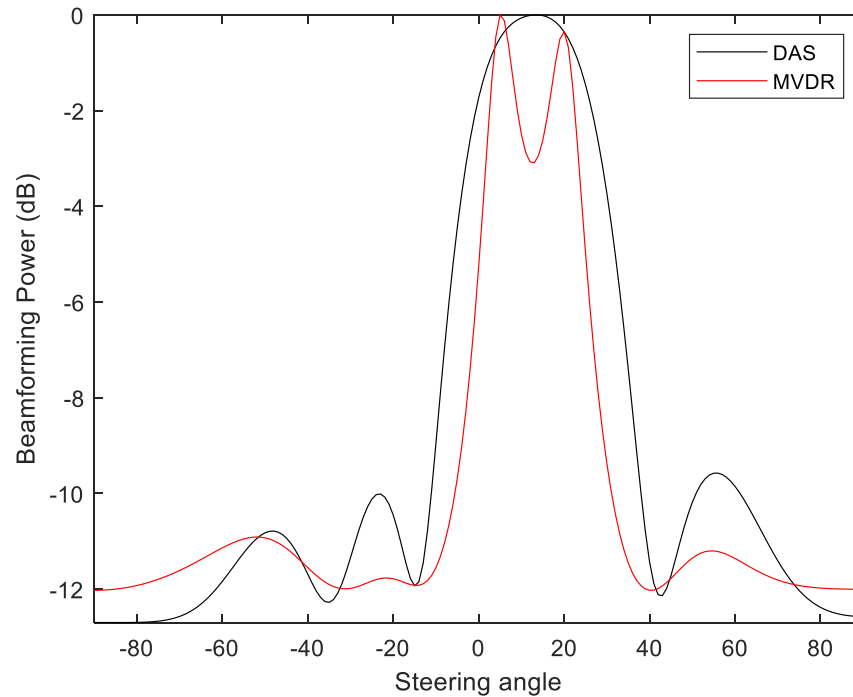
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Multi-channel Source Separation

- MVDR beamformer



Multi-channel Source Separation

- The limitation of the beamformer!
 - We have to know the direction-of-arrival (DoA) of the sound

$$y(\boldsymbol{\theta}, t) = \underset{\substack{\uparrow \\ \text{Convolution}}}{w(\boldsymbol{\theta}, -t)^H} * x(t) \quad \longleftrightarrow_{\text{FFT \& iFFT}} \quad y(\boldsymbol{\theta}, \omega) = \underset{\substack{\text{Beamforming} \\ \text{output}}}{w(\boldsymbol{\theta}, \omega)^H} \underset{\substack{\text{Measured} \\ \text{pressure signals}}}{x(\omega)}$$

Weight (Spatial filter)

- However, It is have to find out the accurate DoA...

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