

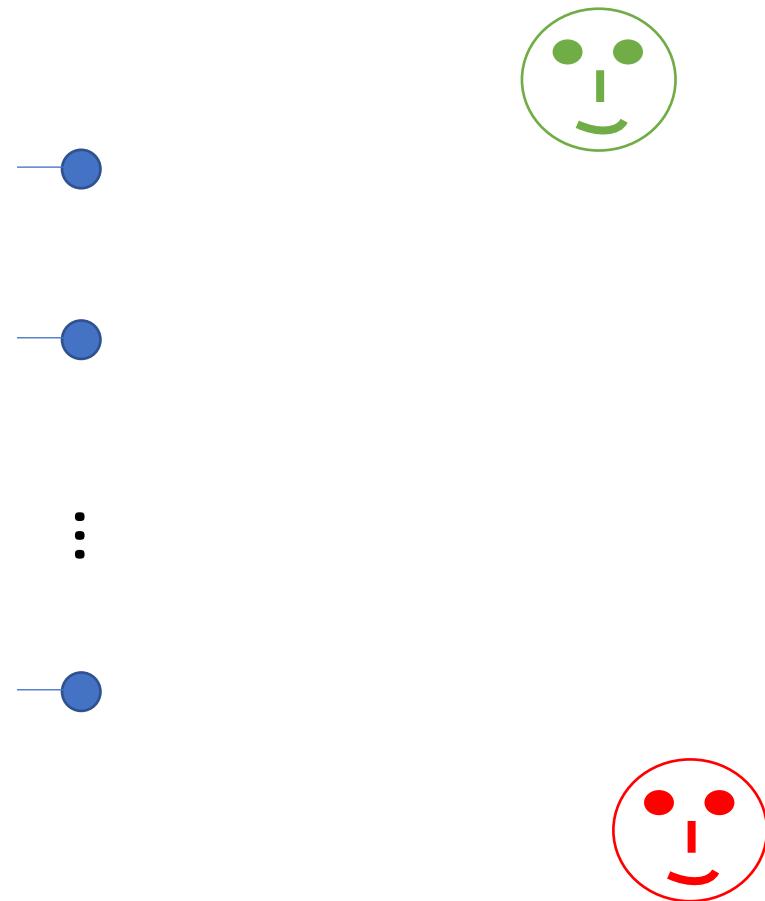
Multi-Microphone Noise Reduction and Source Separation

Andreas Koutrouvelis

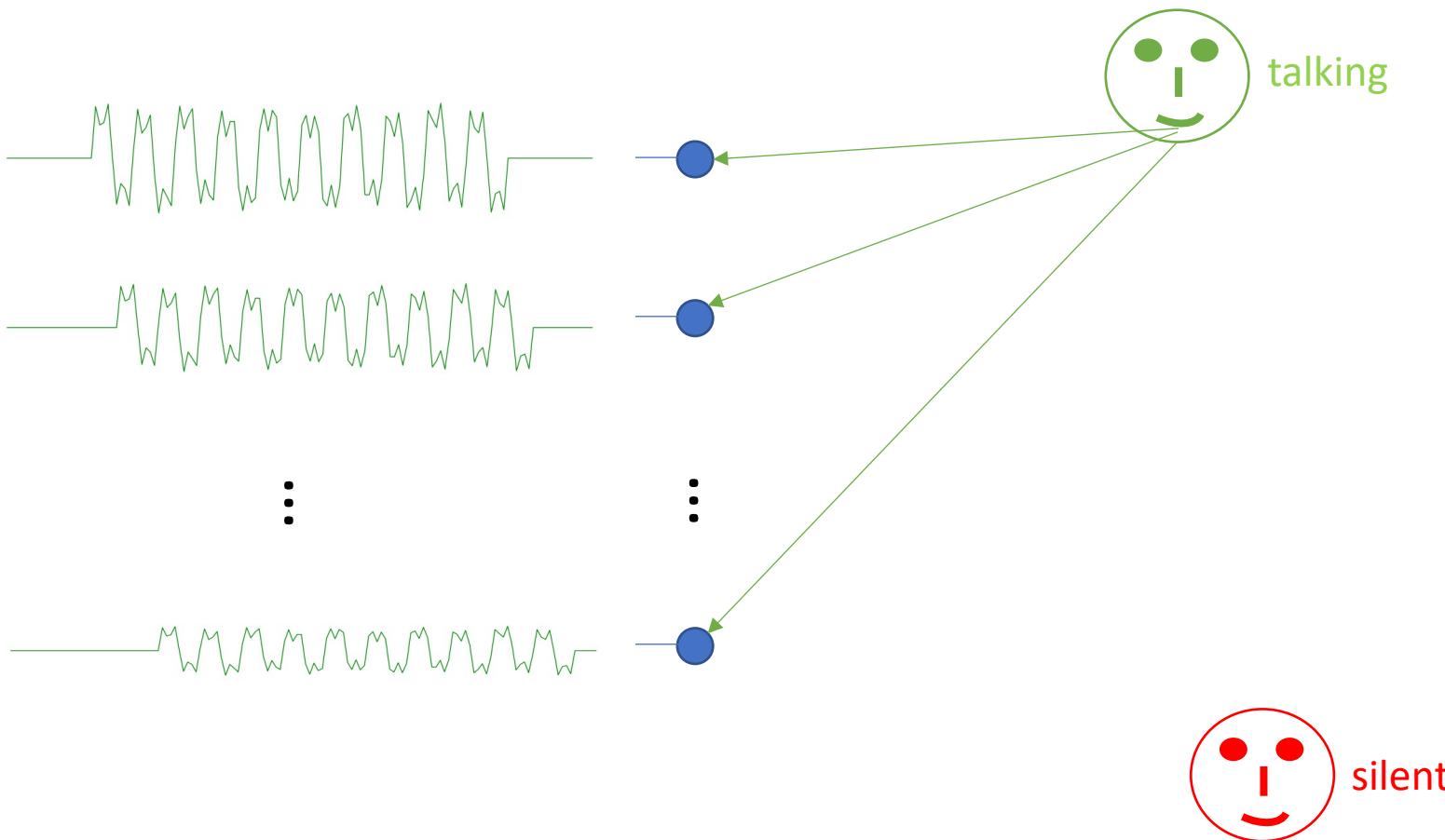
Outline

- Introduction
- Multi-microphone signal model
- Multi-microphone noise reduction
- Multi-microphone source separation
- Estimation of multi-microphone signal model parameters
- Experiments

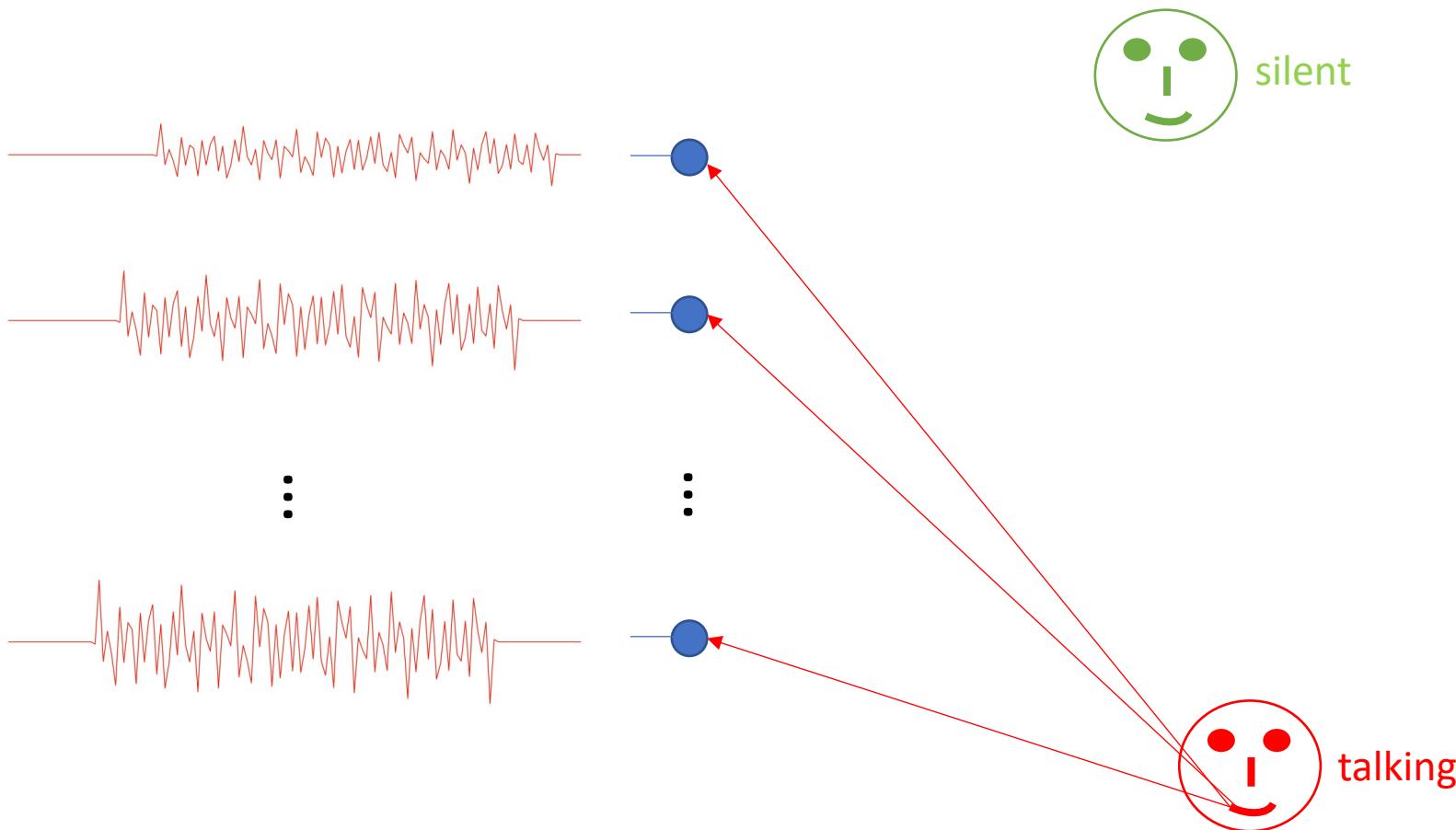
Introduction: a simple acoustic scene



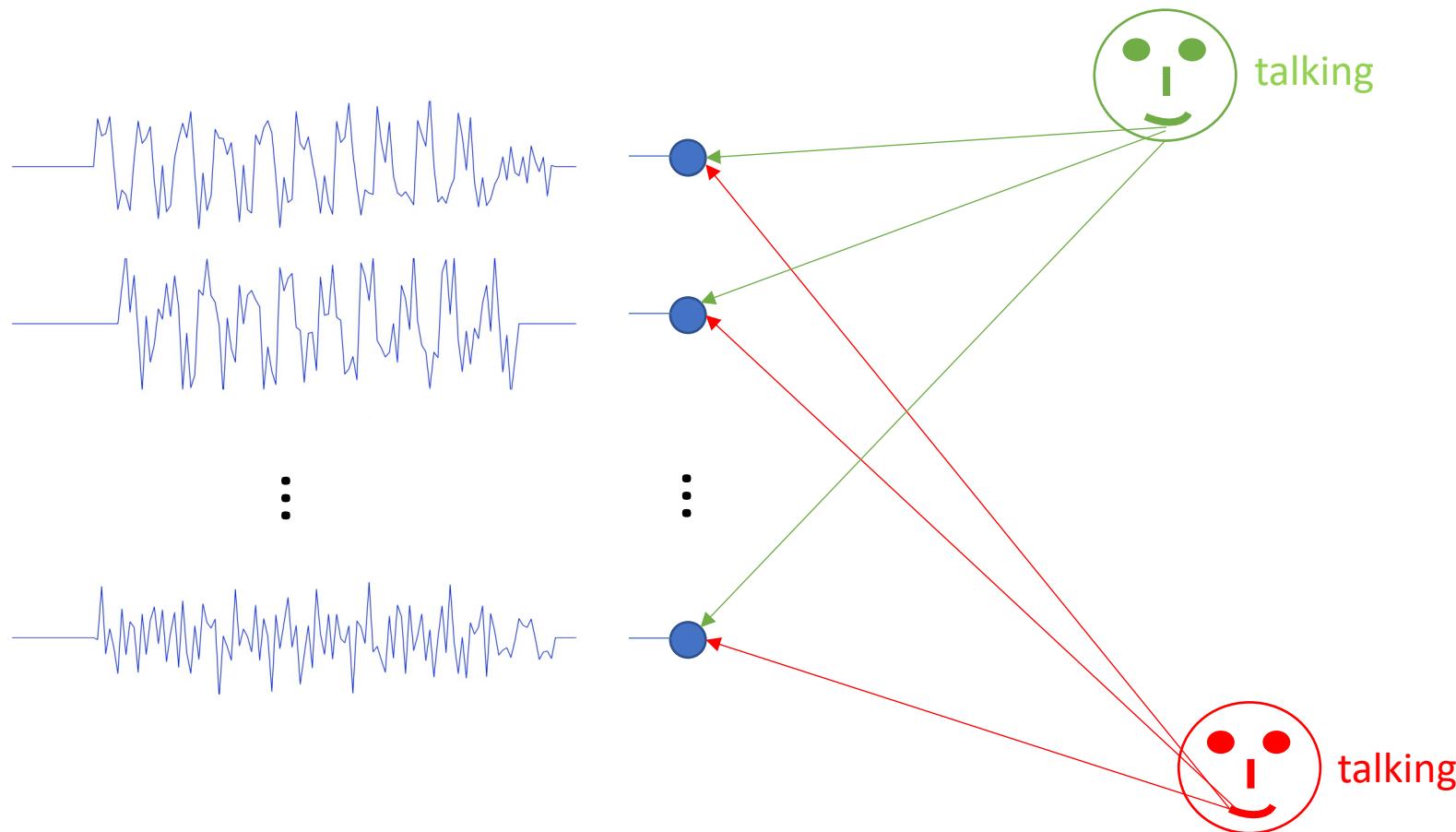
Introduction: a simple acoustic scene



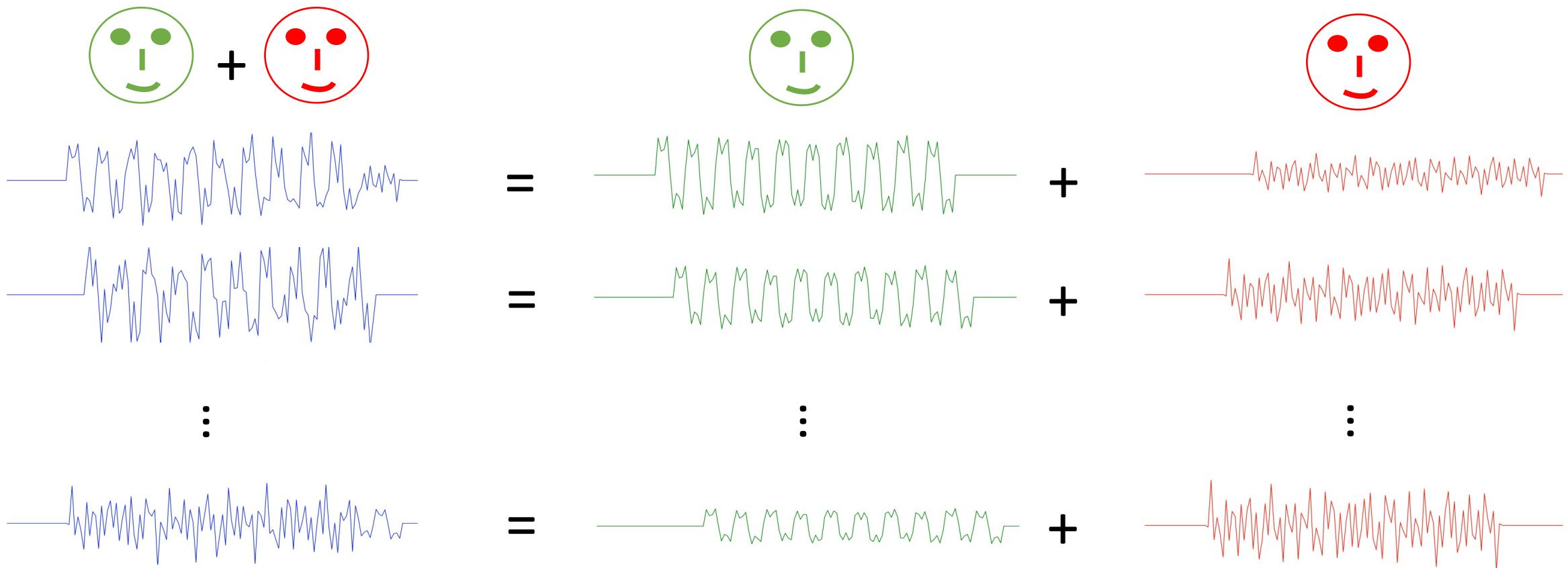
Introduction: a simple acoustic scene



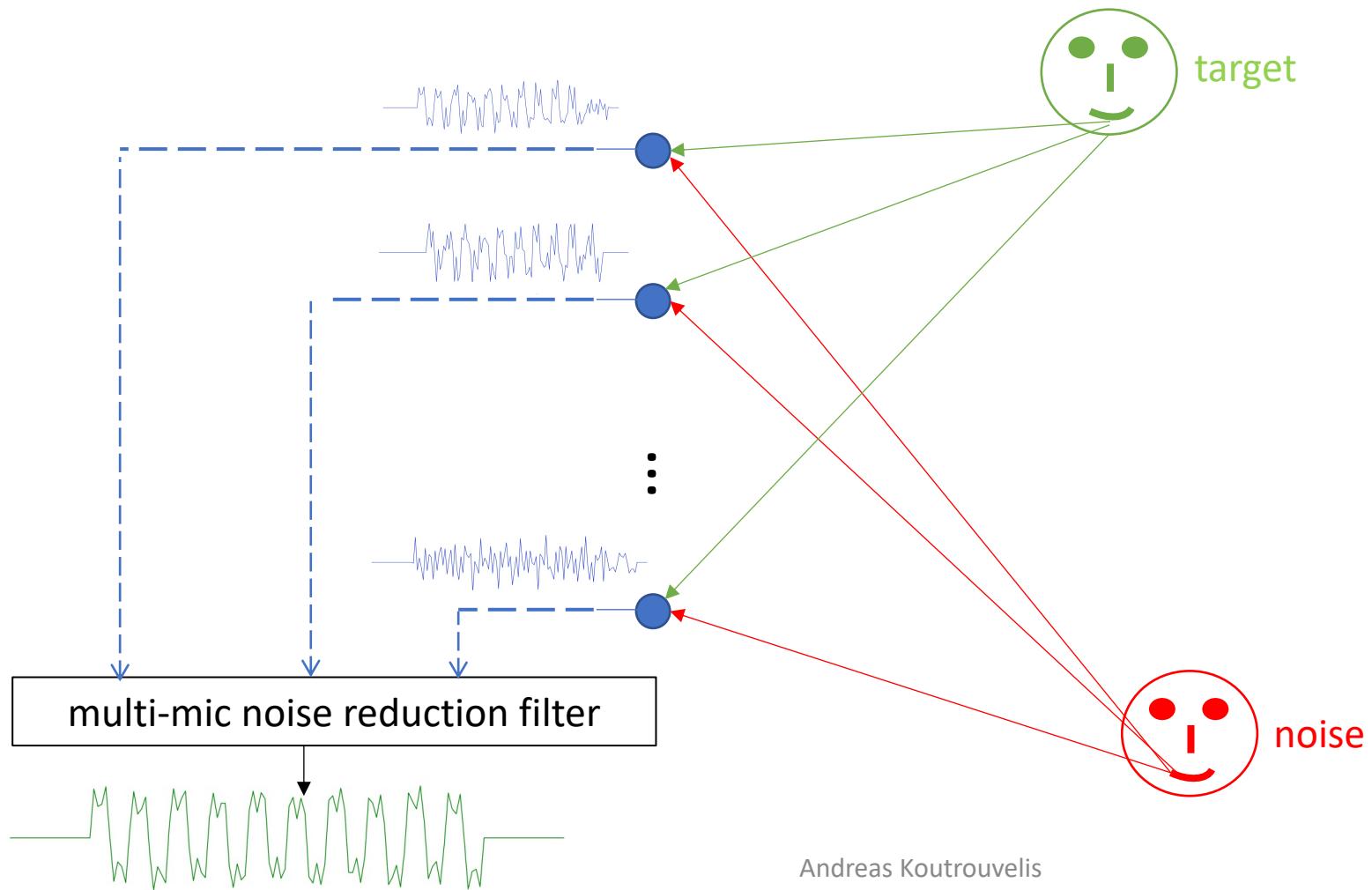
Introduction: a simple acoustic scene



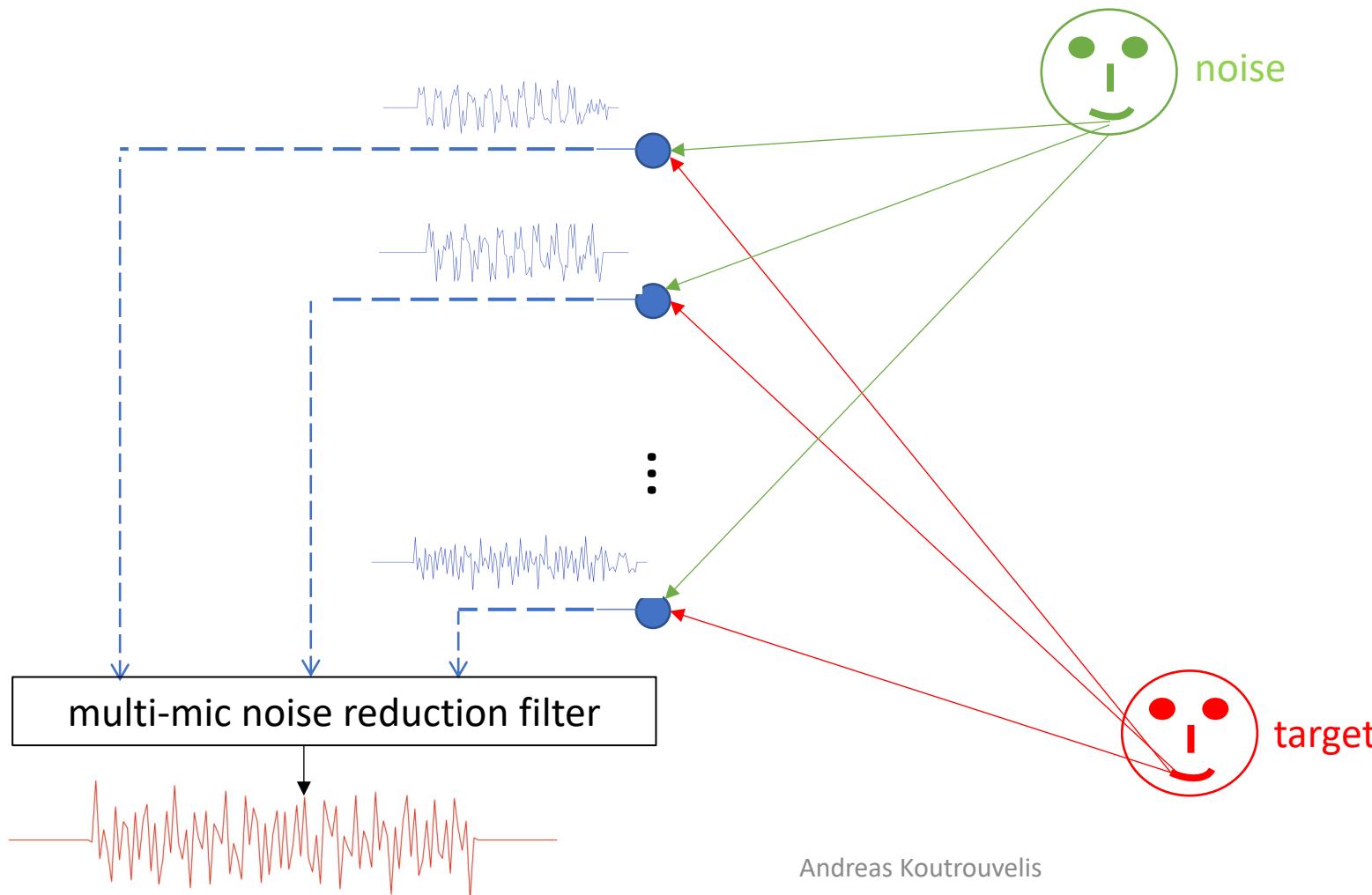
Introduction: a simple acoustic scene



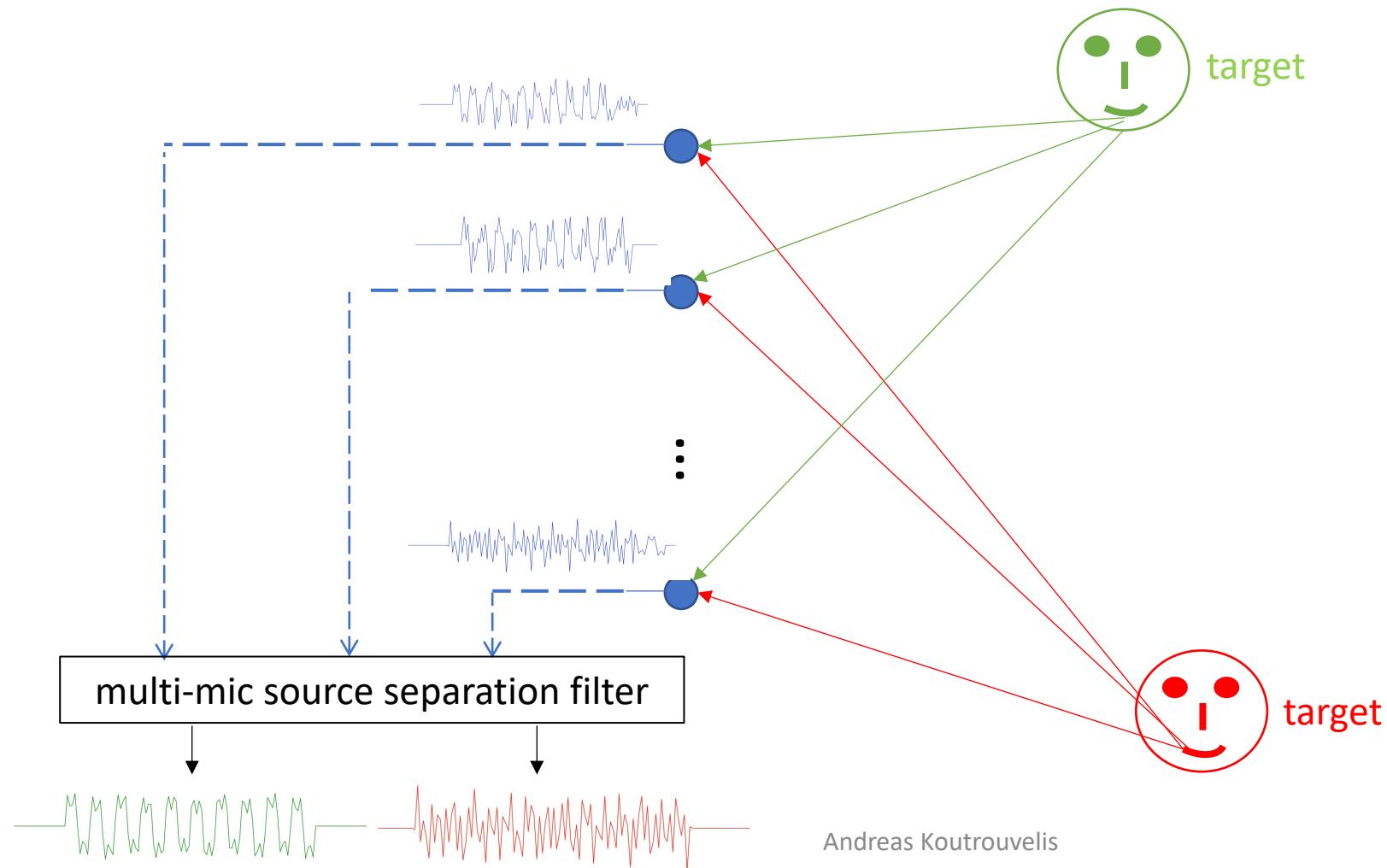
Introduction: multi-microphone noise reduction problem



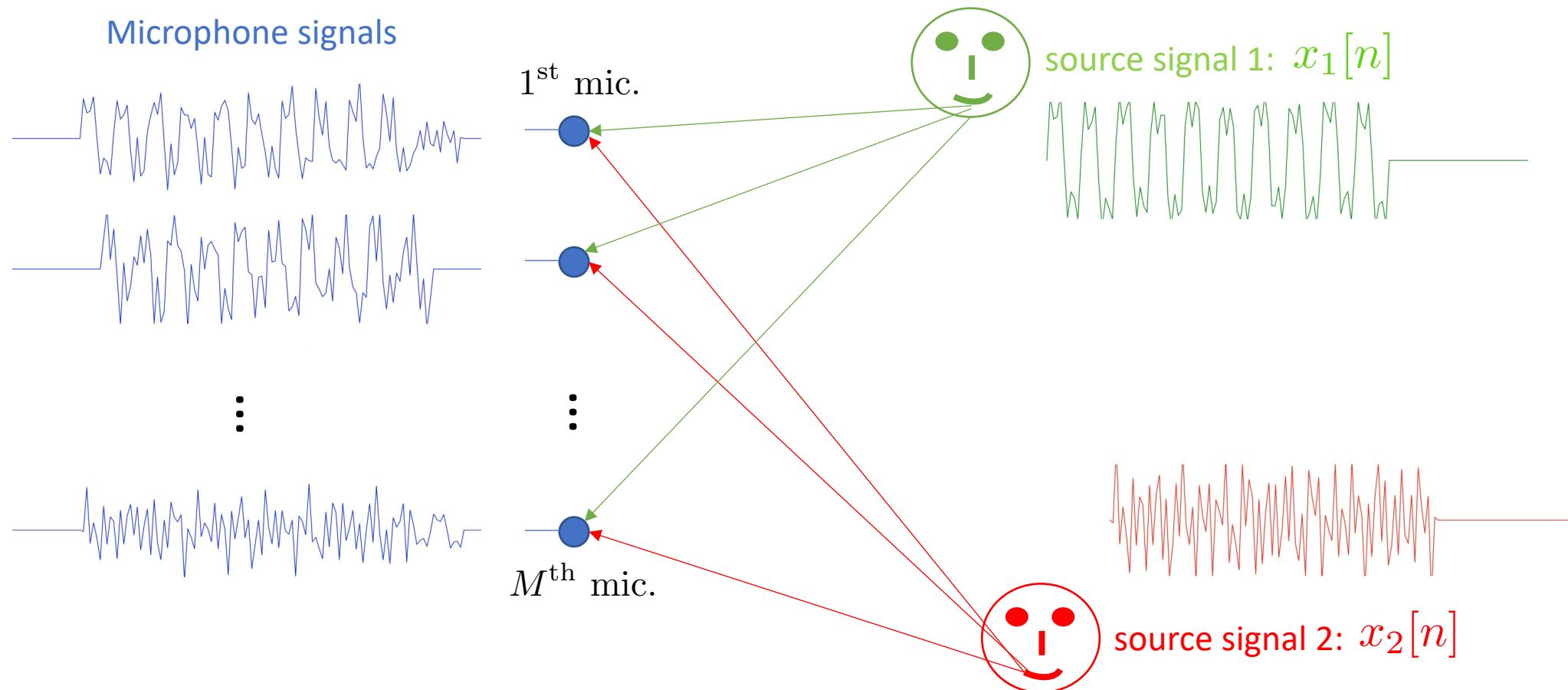
Introduction: multi-microphone noise reduction problem



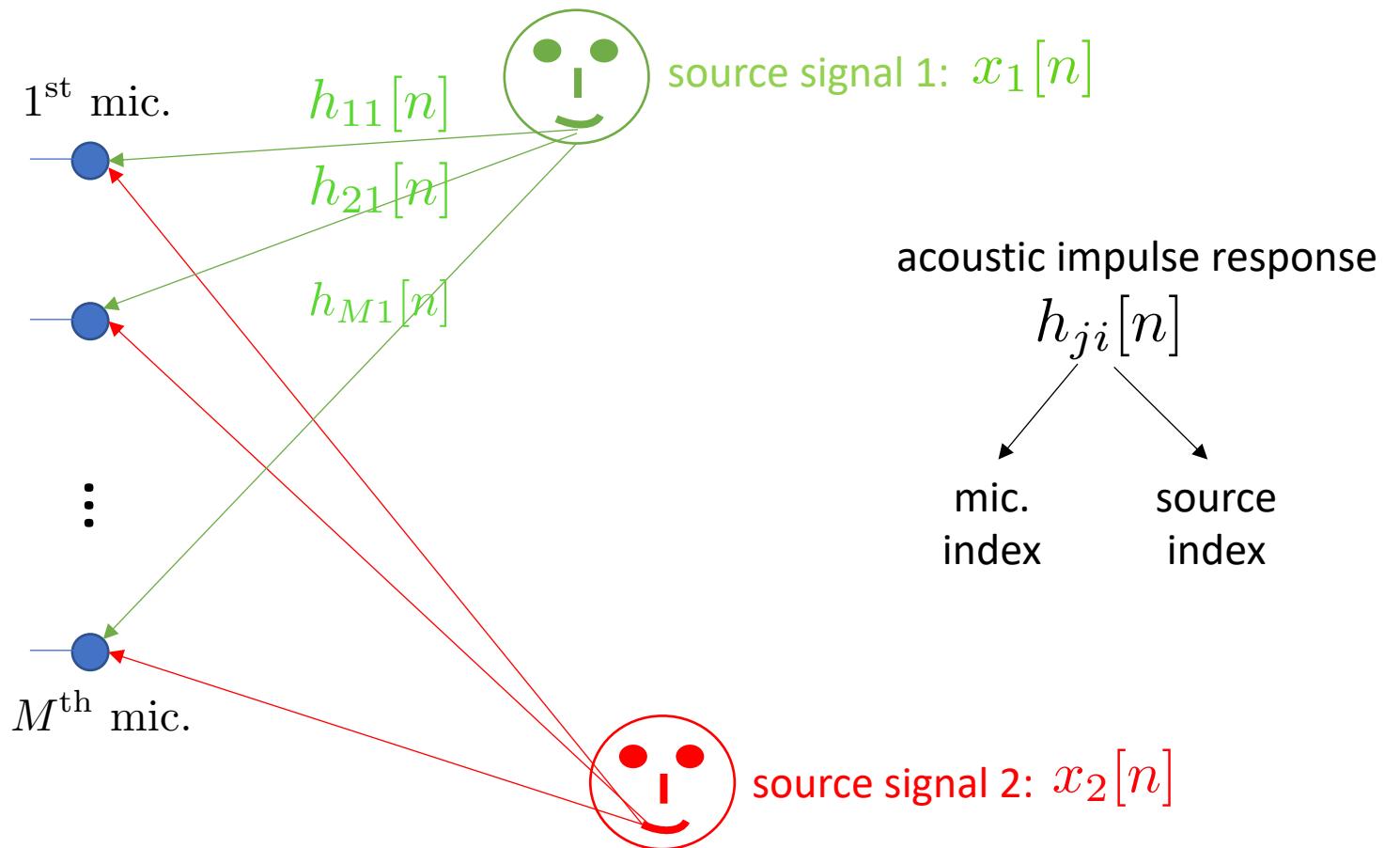
Introduction: source separation problem



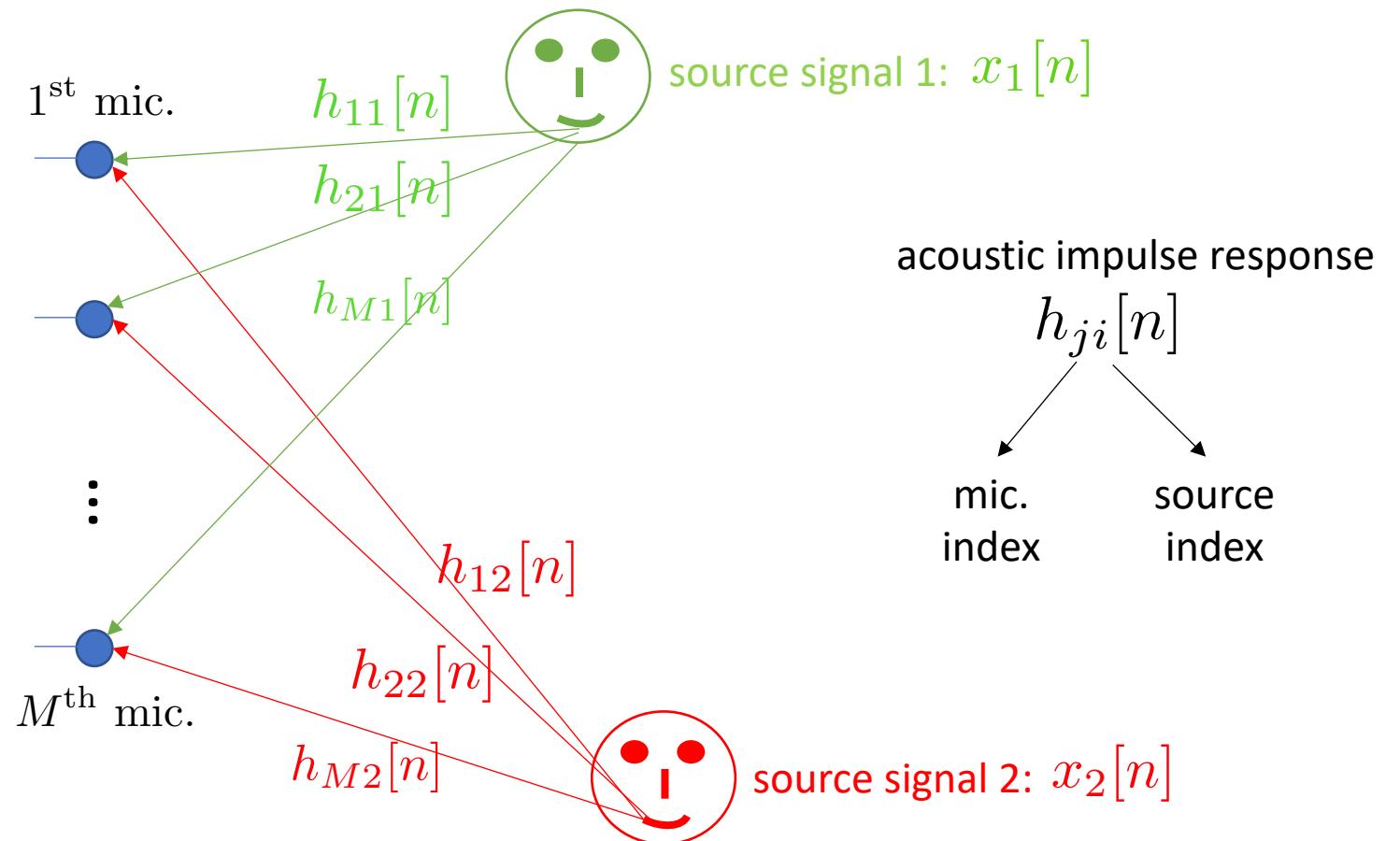
Multi-microphone signal model in time domain



Multi-microphone signal model in time domain



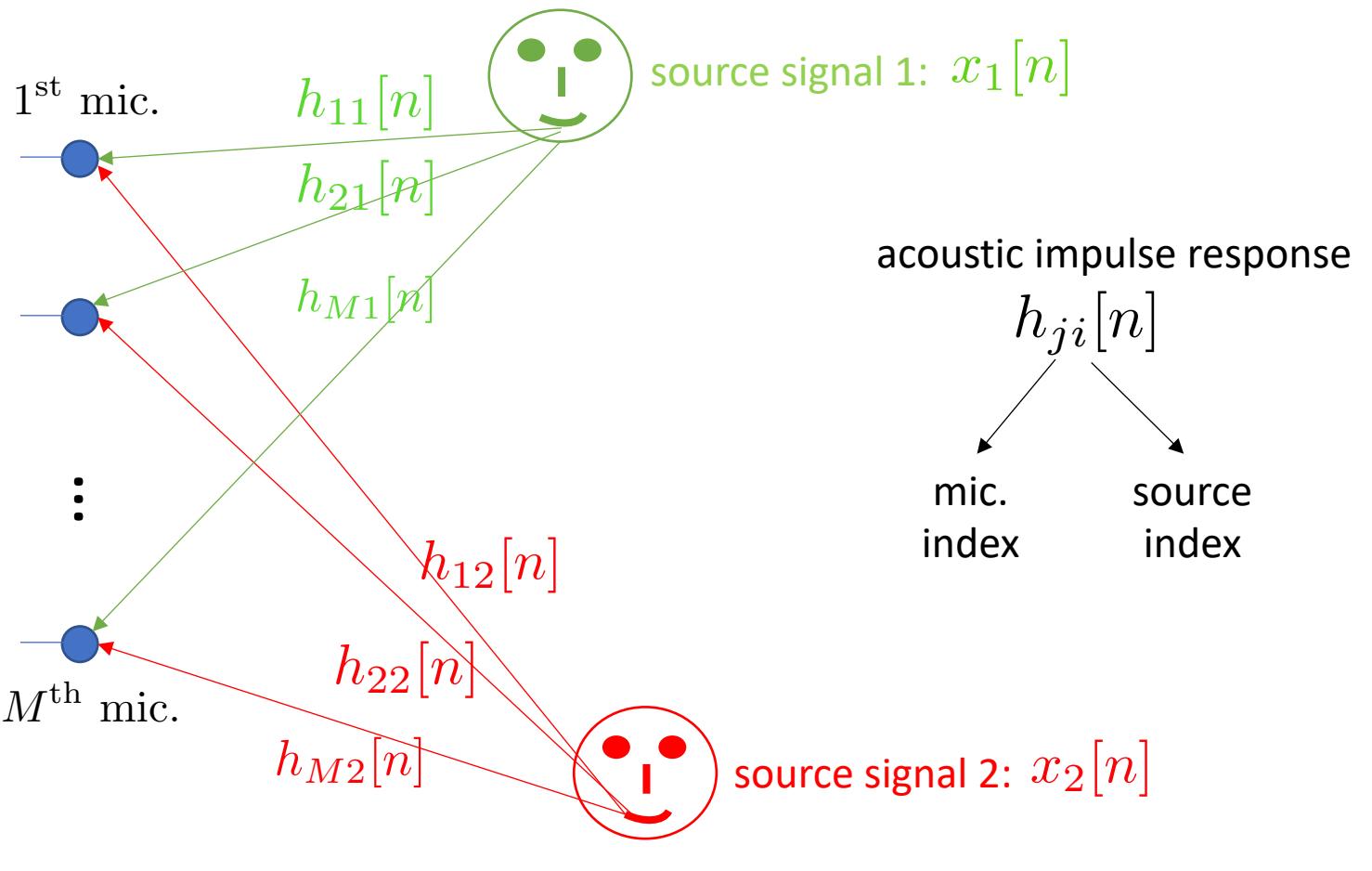
Multi-microphone signal model in time domain



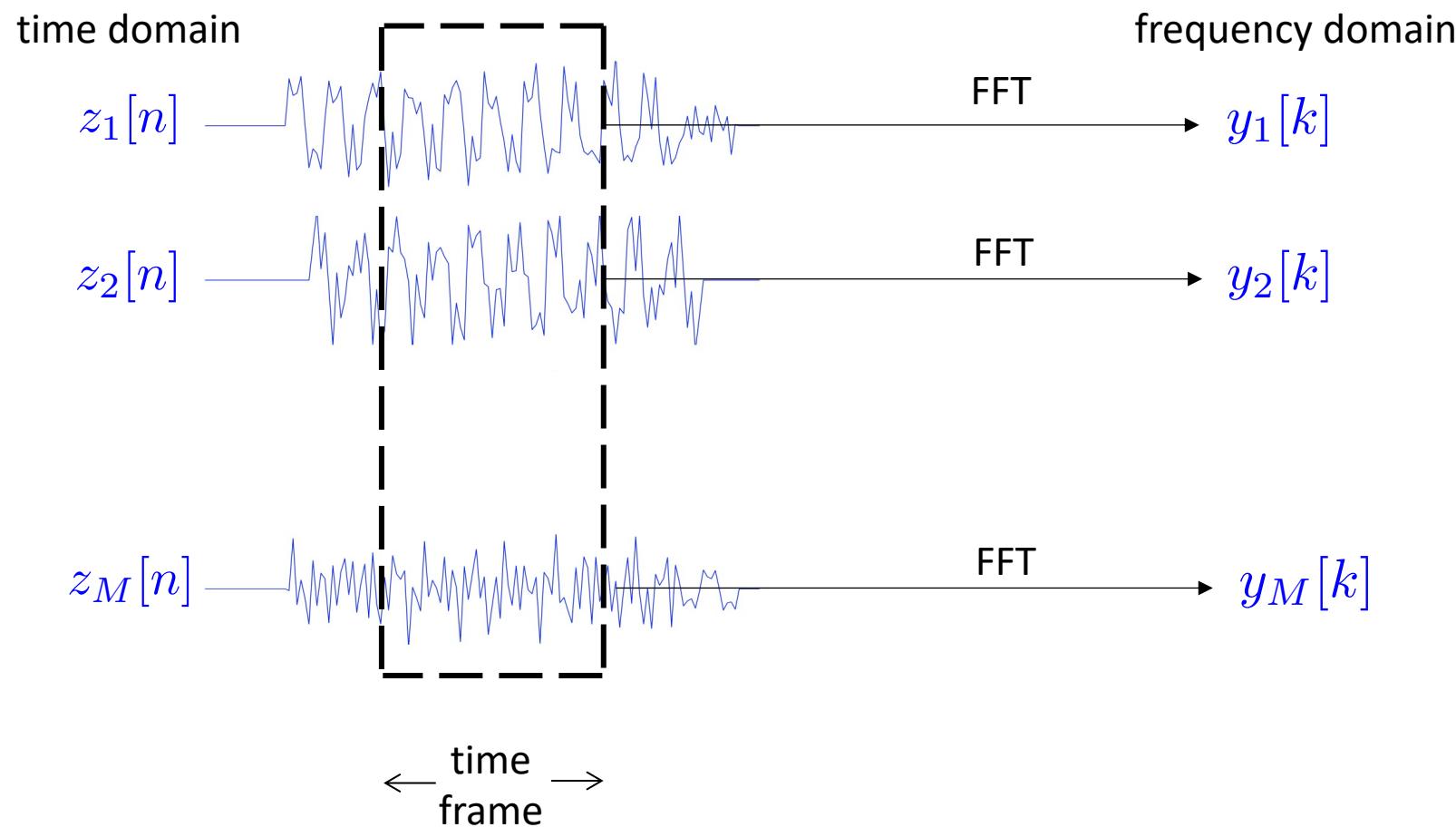
Multi-microphone signal model in time domain

$$\begin{aligned}
 z_1[n] &= x_1[n] * h_{11}[n] \\
 &\quad + x_2[n] * h_{12}[n] + v_1[n] \\
 z_2[n] &= x_1[n] * h_{21}[n] \\
 &\quad + x_2[n] * h_{22}[n] + v_2[n] \\
 \vdots & \\
 z_M[n] &= x_1[n] * h_{M1}[n] \\
 &\quad + x_2[n] * h_{M2}[n] + v_M[n]
 \end{aligned}$$

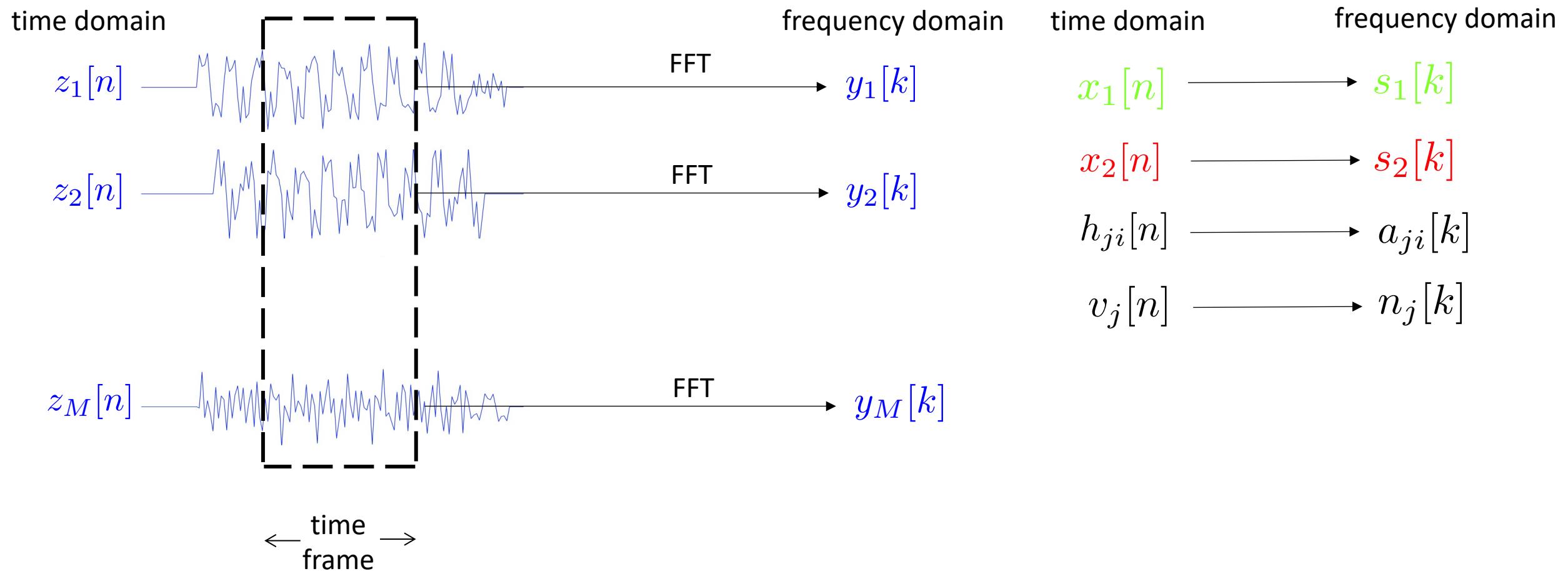
mic. self noise



From time-domain to frequency domain



From time-domain to frequency domain



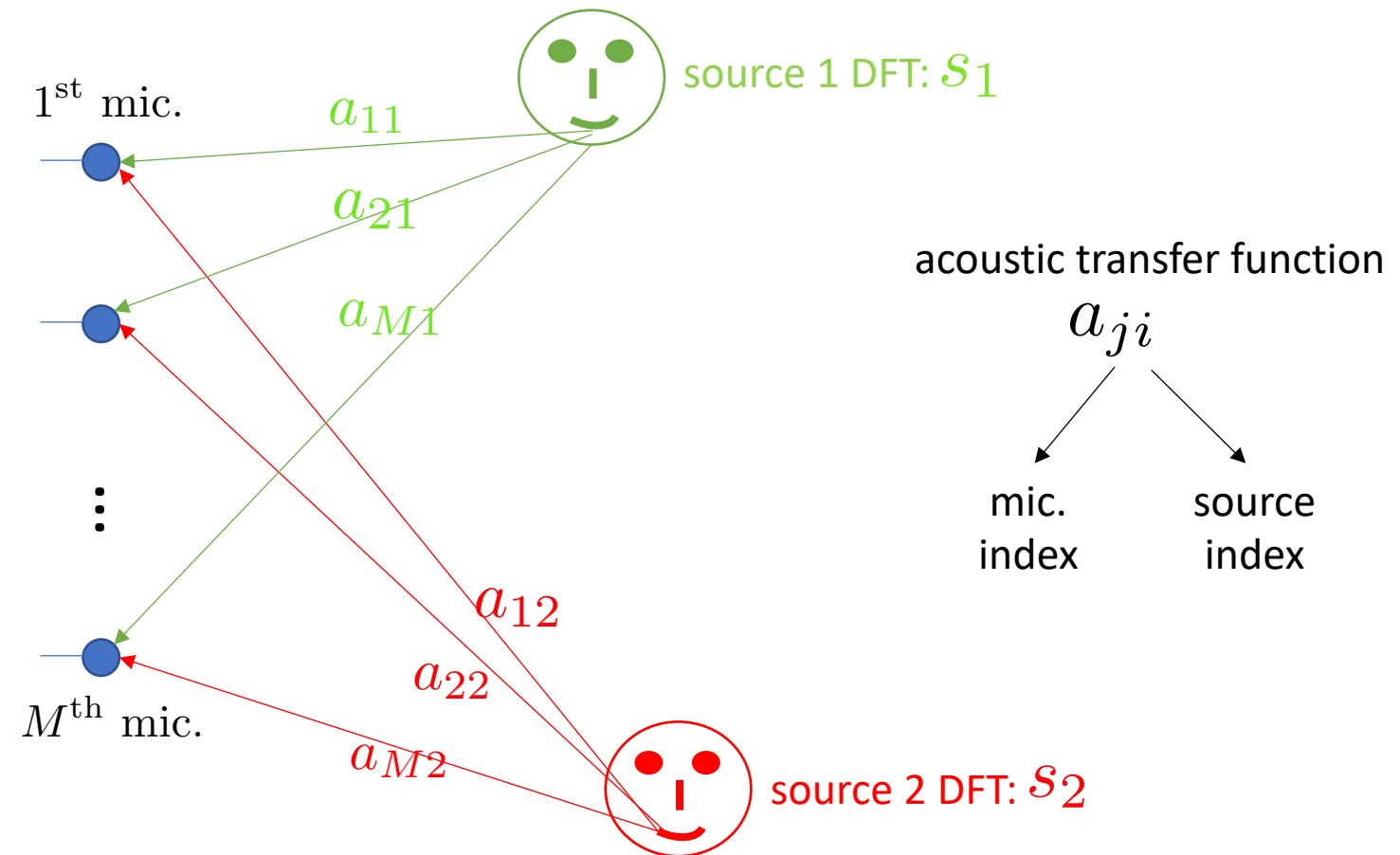
Multi-microphone signal model in frequency domain for a single frequency bin

$$y_1 = s_1 a_{11} + s_2 a_{12} + n_1$$

$$y_2 = s_1 a_{21} + s_2 a_{22} + n_2$$

:

$$y_M = s_1 a_{M1} + s_2 a_{M2} + n_M$$



Multi-microphone signal model in frequency domain for a single frequency bin

$$y_1 = s_1 a_{11} + s_2 a_{12} + n_1$$

Approximation!!!: DFT multiplication means circular convolution in time domain. If the FFT size is large enough the circular convolution (after computing the IFFT) becomes a good approximation of the linear convolution.

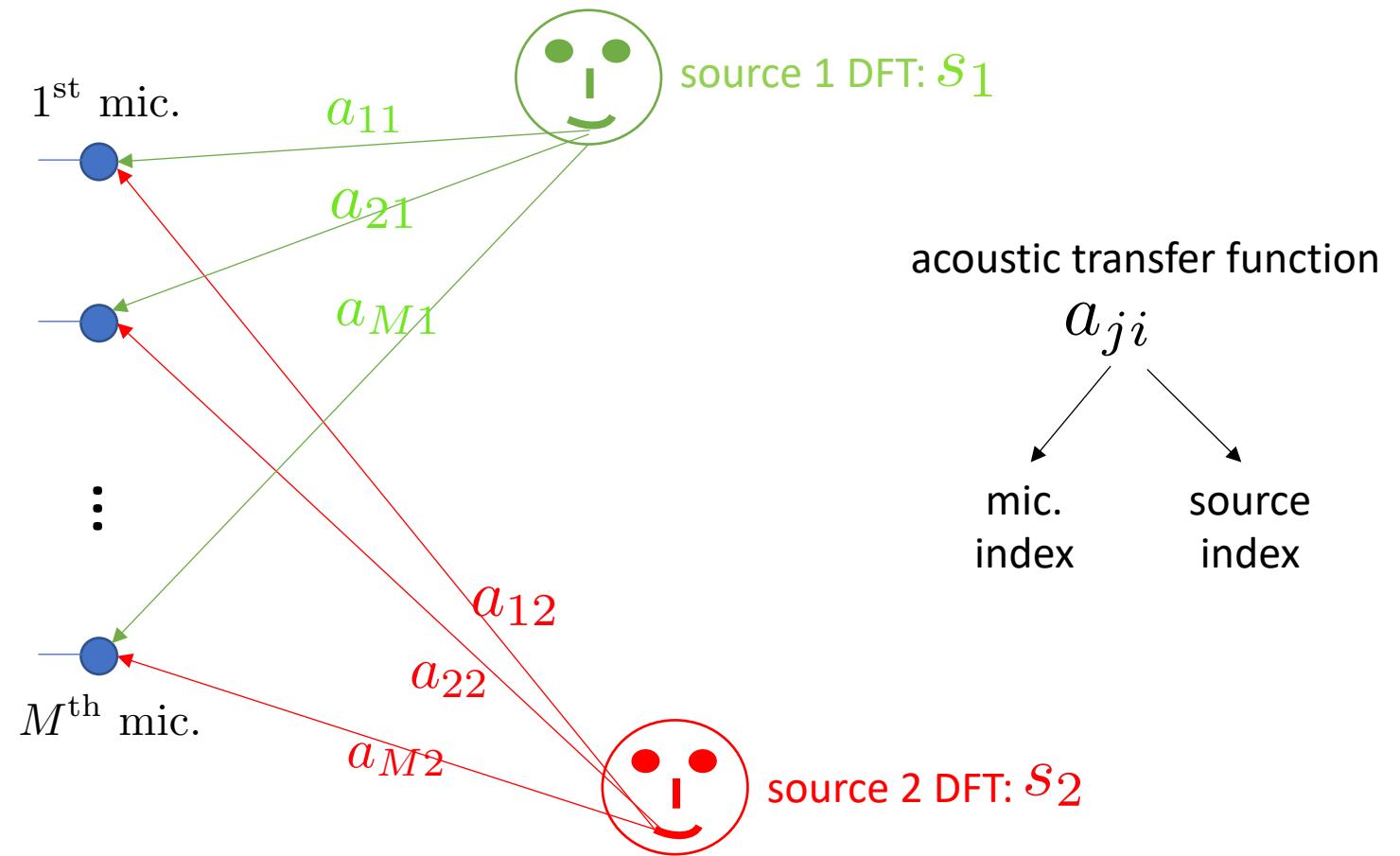
$$y_2 = s_1 a_{21} + s_2 a_{22} + n_2$$

:

$$y_M = s_1 a_{M1} + s_2 a_{M2} + n_M$$

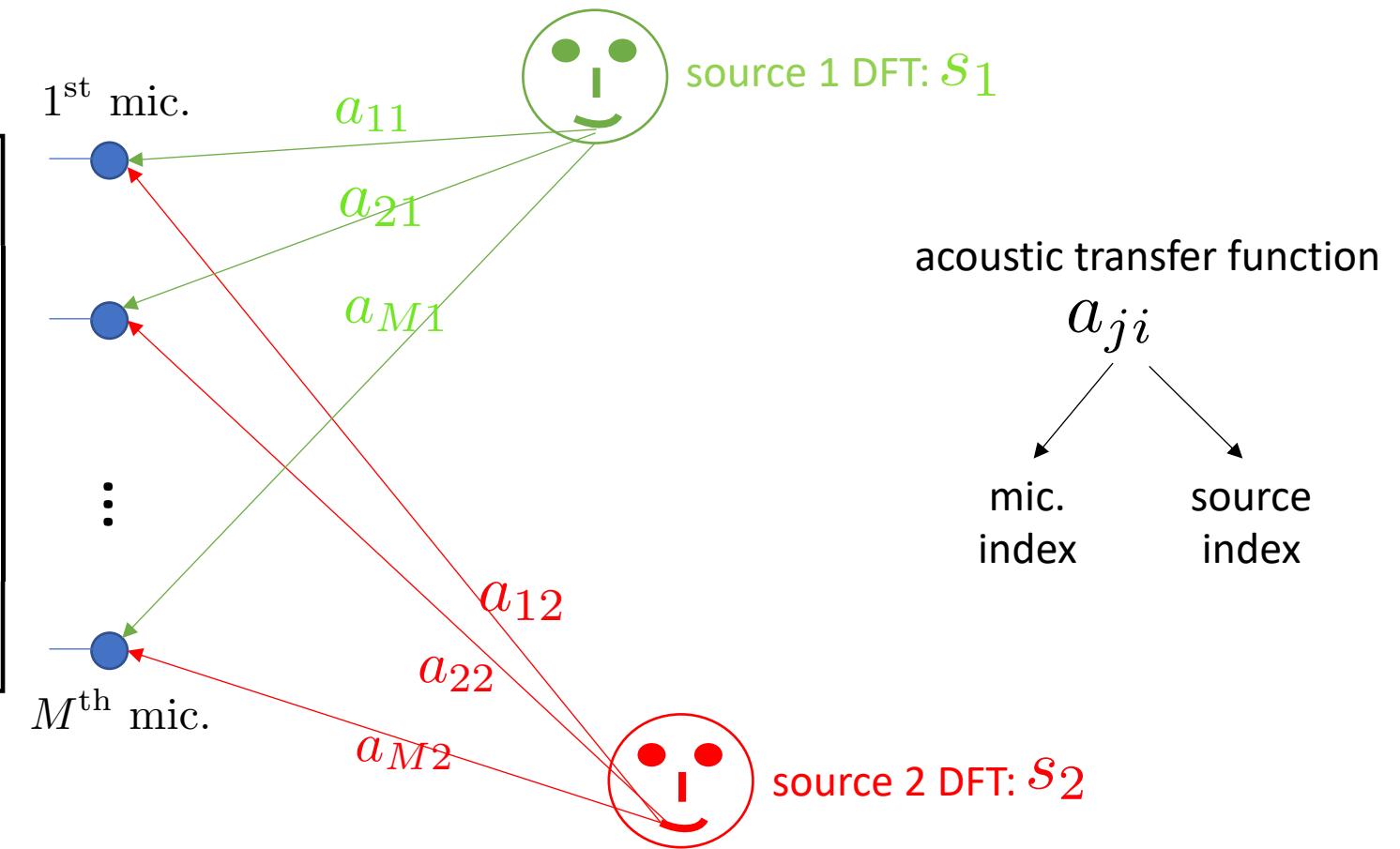
Multi-microphone signal model in frequency domain for a single frequency bin using matrices and vectors

$$\begin{bmatrix} y_1 \\ y_2 \\ \vdots \\ y_M \end{bmatrix} = s_1 \begin{bmatrix} a_{11} \\ a_{21} \\ \vdots \\ a_{M1} \end{bmatrix} + s_2 \begin{bmatrix} a_{12} \\ a_{22} \\ \vdots \\ a_{M2} \end{bmatrix} + \begin{bmatrix} n_1 \\ n_2 \\ \vdots \\ n_M \end{bmatrix}$$



Multi-microphone signal model in frequency domain for a single frequency bin using matrices and vectors

$$\begin{bmatrix} y_1 \\ y_2 \\ \vdots \\ y_M \\ \mathbf{y} \end{bmatrix} = s_1 \begin{bmatrix} a_{11} \\ a_{21} \\ \vdots \\ a_{M1} \\ \mathbf{a}_1 \end{bmatrix} + s_2 \begin{bmatrix} a_{12} \\ a_{22} \\ \vdots \\ a_{M2} \\ \mathbf{a}_2 \end{bmatrix} + \begin{bmatrix} n_1 \\ n_2 \\ \vdots \\ n_M \\ \mathbf{n} \end{bmatrix}$$



bold-face lower-case letters for vectors

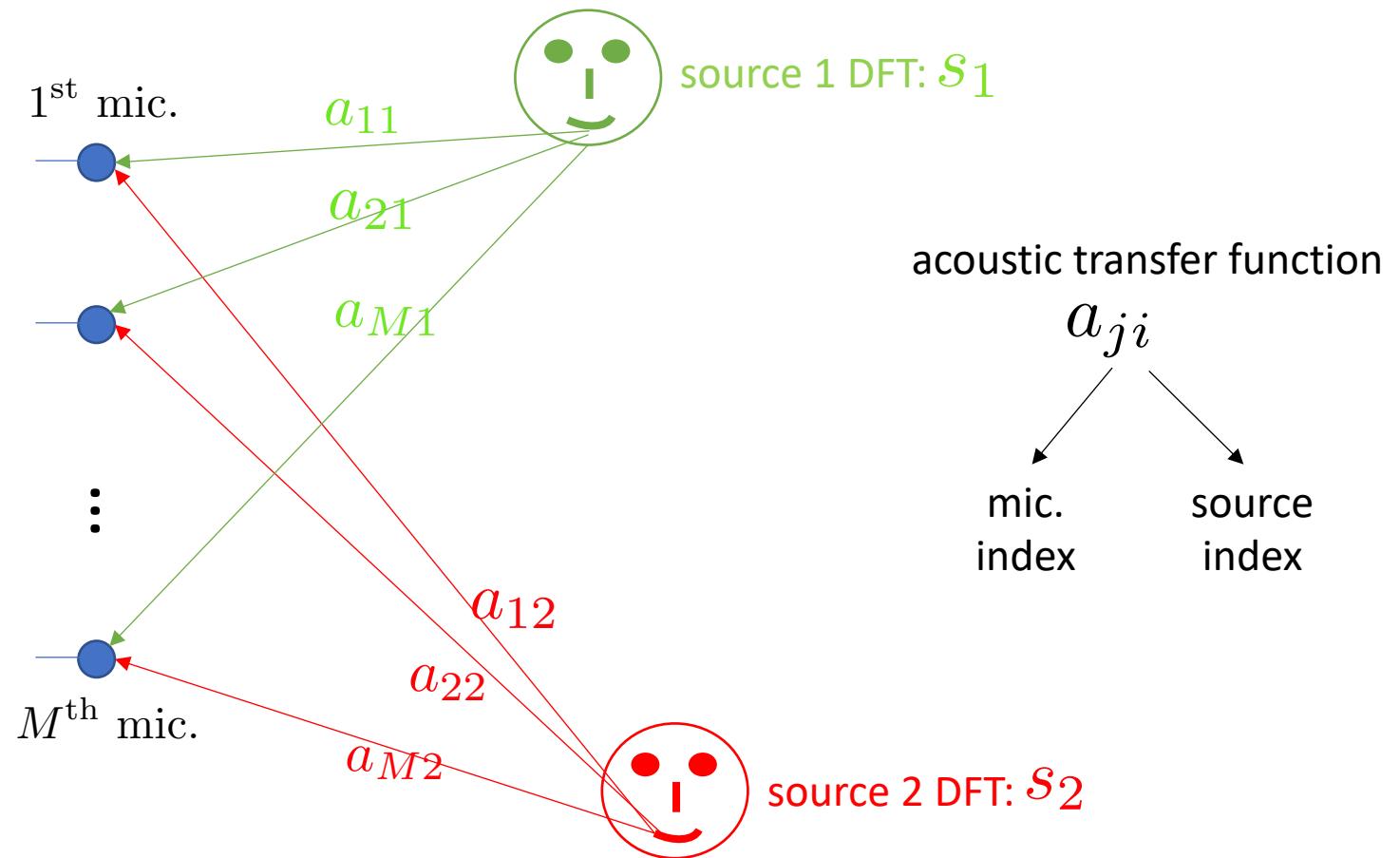
Multi-microphone signal model in frequency domain for a single frequency bin using matrices and vectors

$$\mathbf{y} = [\mathbf{a}_1 \quad \mathbf{a}_2] \begin{bmatrix} \mathbf{s}_1 \\ \mathbf{s}_2 \end{bmatrix} + \mathbf{n}$$

A **S**

$M \times 1 \quad M \times 2 \quad 2 \times 1 \quad M \times 1$

bold-face lower-case letters for vectors
bold-face upper-case letters for matrices



Multi-microphone signal model in frequency domain with r sources

$$\mathbf{y} = \sum_{i=1}^r \mathbf{a}_i s_i + \mathbf{n} \rightarrow \mathbf{y} = \mathbf{As} + \mathbf{n}$$

$M \times r \quad r \times 1 \quad M \times 1$

Multi-microphone signal model in frequency domain with r sources

$$\mathbf{y} = \sum_{i=1}^r \mathbf{a}_i s_i + \mathbf{n} \rightarrow \mathbf{y} = \mathbf{As} + \mathbf{n}$$

$M \times r \quad r \times 1 \quad M \times 1$

↓

$$\mathbf{P}_{\mathbf{y}} = \mathbb{E} [(\mathbf{y} - \mu_{\mathbf{y}})(\mathbf{y} - \mu_{\mathbf{y}})^H] \quad \text{cross-power spectral density matrix}$$

Multi-microphone signal model in frequency domain with r sources

$$\mathbf{y} = \sum_{i=1}^r \mathbf{a}_i s_i + \mathbf{n} \rightarrow \mathbf{y} = \mathbf{As} + \mathbf{n}$$

$M \times r \quad r \times 1 \quad M \times 1$

$\mathbf{P}_y = E[\mathbf{y}\mathbf{y}^H]$ cross-power spectral density matrix if $\mu_y = \mathbf{0}$

```
graph TD; A["y = sumi=1 to r ai si + n"] --> B["y = As + n"]; B --> C["M x r    r x 1    M x 1"]; C --> D["P_y = E[yy^H]    cross-power spectral density matrix if mu_y = 0"];
```

Multi-microphone signal model in frequency domain with r sources

$$\begin{aligned}
 \mathbf{y} &= \sum_{i=1}^r \mathbf{a}_i s_i + \mathbf{n} \rightarrow \mathbf{y} = \mathbf{As} + \mathbf{n} \\
 &\quad \downarrow \\
 &\quad M \times r \quad r \times 1 \quad M \times 1 \\
 \mathbf{P}_y &= E[\mathbf{yy}^H] = \sum_{i=1}^r \underbrace{E[s_i s_i^*]}_{p_i} \mathbf{a}_i \mathbf{a}_i^H + \underbrace{E[\mathbf{nn}^H]}_{\mathbf{P}_n} = \sum_{i=1}^r p_i \mathbf{a}_i \mathbf{a}_i^H + \mathbf{P}_n \\
 &= \mathbf{APA}^H + \mathbf{P}_n
 \end{aligned}$$

Multi-microphone signal model in frequency domain with r sources

$$\begin{aligned}
 \mathbf{y} &= \sum_{i=1}^r \mathbf{a}_i s_i + \mathbf{n} \rightarrow \mathbf{y} = \mathbf{As} + \mathbf{n} \\
 &\quad \downarrow \\
 &\quad M \times r \quad r \times 1 \quad M \times 1 \\
 \mathbf{P}_{\mathbf{y}} &= \mathbb{E} [\mathbf{yy}^H] = \sum_{i=1}^r \underbrace{\mathbb{E} [s_i s_i^*]}_{p_i} \mathbf{a}_i \mathbf{a}_i^H + \underbrace{\mathbb{E} [\mathbf{nn}^H]}_{\mathbf{P}_n} = \sum_{i=1}^r p_i \mathbf{a}_i \mathbf{a}_i^H + \mathbf{P}_n \\
 &= \mathbf{APA}^H + \boxed{\mathbf{P}_n} \quad \xrightarrow{\text{is a diagonal matrix because the microphone self noises are mutually uncorrelated.}}
 \end{aligned}$$

Multi-microphone noise reduction

Multi-microphone noise reduction

- Typically, in multi-microphone noise reduction we have one target. That is,

$$\mathbf{y} = \underbrace{\mathbf{a}_1 s_1}_{\text{target}} + \underbrace{\sum_{i=2}^r \mathbf{a}_i s_i}_{\mathbf{q}: \text{ total noise}} + \mathbf{n}$$

- The goal is to estimate the target s_1 from the noisy DFT coefficients \mathbf{y} . That is,

$$\hat{s}_1 = f(\mathbf{y}),$$

where $f(\cdot)$ is the filter function, where in this presentation is linear. That is,

$$\hat{s}_1 = \mathbf{w}^H \mathbf{y},$$

where \mathbf{W} is the filter vector.

- We want to find a \mathbf{W} such that $\hat{s}_1 = \mathbf{w}^H \mathbf{y} = \underbrace{\mathbf{w}^H \mathbf{a}_1 s_1}_{\approx s_1} + \underbrace{\mathbf{w}^H \mathbf{q}}_{\approx 0}$,

Multi-microphone noise reduction

- Typically, in multi-microphone noise reduction we have one target. That is,

$$\mathbf{y} = \underbrace{\mathbf{a}_1 s_1}_{\text{target}} + \underbrace{\sum_{i=2}^r \mathbf{a}_i s_i}_{\mathbf{q}: \text{ total noise}} + \mathbf{n}$$

- The goal is to estimate the target s_1 from the noisy DFT coefficients \mathbf{y} . That is,

$$\hat{s}_1 = f(\mathbf{y}),$$

where $f(\cdot)$ is the filter function, where in this presentation is linear. That is,

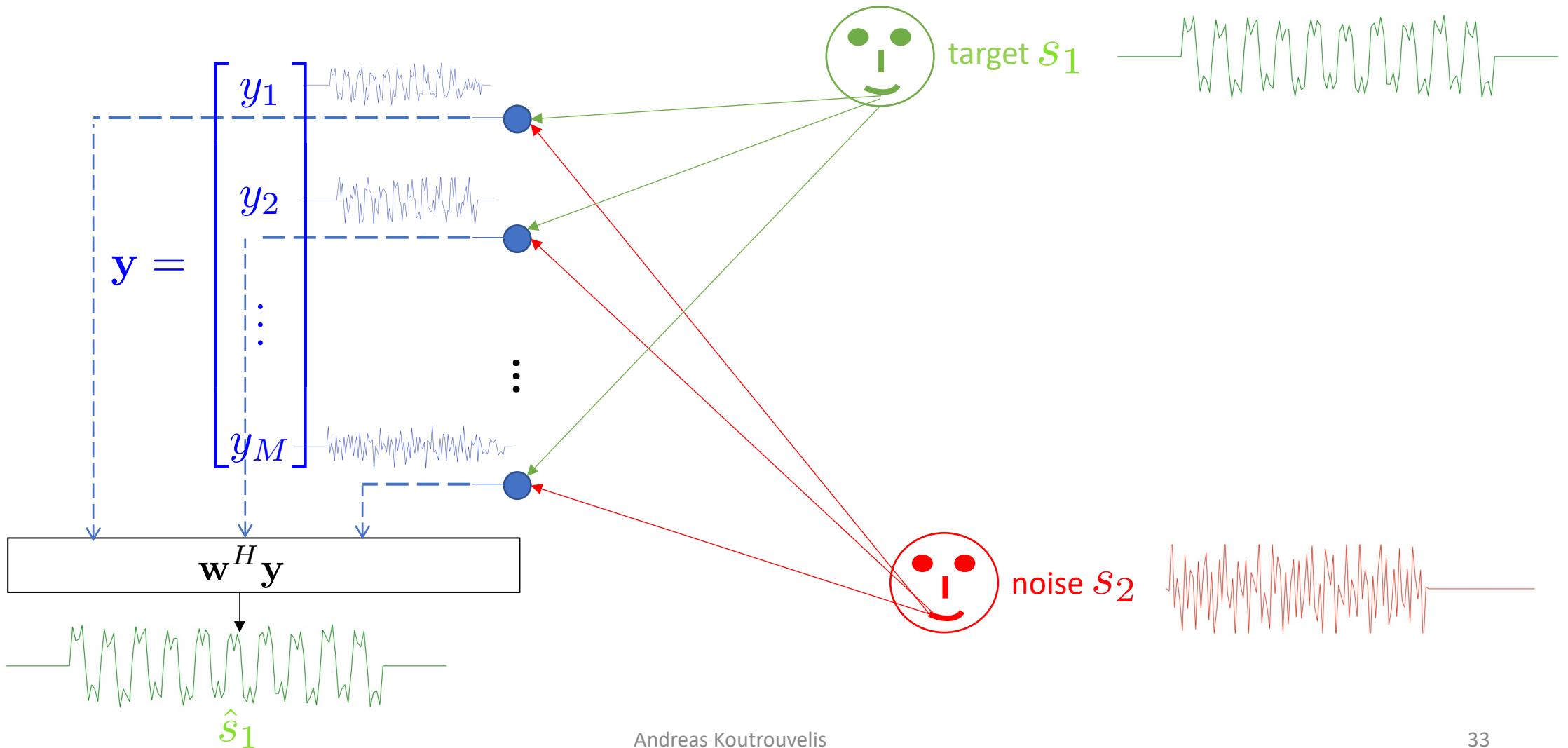
$$\hat{s}_1 = \mathbf{w}^H \mathbf{y},$$

where \mathbf{w} is the filter vector.

- We want to find a \mathbf{w} such that $\hat{s}_1 = \mathbf{w}^H \mathbf{y} = \underbrace{\mathbf{w}^H \mathbf{a}_1 s_1}_{\approx s_1} + \underbrace{\mathbf{w}^H \mathbf{q}}_{\approx 0},$

$$\mathbf{w}^H \mathbf{a}_1 = 1$$

Multi-microphone noise reduction



Multi-microphone noise reduction

- Signal model in freq. domain: $\mathbf{y} = \underbrace{\mathbf{a}_1 s_1}_{\text{target}} + \underbrace{\sum_{i=2}^r \mathbf{a}_i s_i}_{\mathbf{q}: \text{ total noise}} + \mathbf{n}$
- If the total noise \mathbf{q} is stochastic with zero mean (i.e., $E[\mathbf{q}] = \mathbf{0}$) and the target s_1 is deterministic, the variance of the estimator $\hat{s}_1 = \mathbf{w}^H \mathbf{y}$, is given by

$$\begin{aligned} E[(\hat{s}_1 - E[\hat{s}_1])^2] &= E[(\mathbf{w}^H \mathbf{y} - E[\mathbf{w}^H \mathbf{y}])^2] \\ &= E[(\mathbf{w}^H(\mathbf{a}_1 s_1 + \mathbf{q}) - E[\mathbf{w}^H(\mathbf{a}_1 s_1 + \mathbf{q})])^2] \\ &= E[(\mathbf{w}^H \mathbf{a}_1 s_1 + \mathbf{w}^H \mathbf{q} - \mathbf{w}^H \mathbf{a}_1 s_1 - \mathbf{w}^H E[\mathbf{q}])^2] \\ &= E[(\mathbf{w}^H \mathbf{q})^2] = E[\mathbf{w}^H \mathbf{q} \mathbf{q}^H \mathbf{w}] = \mathbf{w}^H \underbrace{E[\mathbf{q} \mathbf{q}^H]}_{\mathbf{P}_q} \mathbf{w} \\ &= \mathbf{w}^H \mathbf{P}_q \mathbf{w} \end{aligned}$$

Multi-microphone noise reduction

- Signal Model in freq. domain: $\mathbf{y} = \underbrace{\mathbf{a}_1 s_1}_{\text{target}} + \sum_{i=2}^r \mathbf{a}_i s_i + \mathbf{q}$
q: total noise
- If the total noise \mathbf{q} is stochastic with zero mean (i.e., $E[\mathbf{q}] = \mathbf{0}$) and the target s_1 is deterministic, the variance of the estimator $\hat{s}_1 = \mathbf{w}^H \mathbf{y}$, is given by

$$\begin{aligned} E[(\hat{s}_1 - E[\hat{s}_1])^2] &= E[(\mathbf{w}^H \mathbf{y} - E[\mathbf{w}^H \mathbf{y}])^2] \\ &= E[(\mathbf{w}^H(\mathbf{a}_1 s_1 + \mathbf{q}) - E[\mathbf{w}^H(\mathbf{a}_1 s_1 + \mathbf{q})])^2] \\ &= E[(\mathbf{w}^H \mathbf{a}_1 s_1 + \mathbf{w}^H \mathbf{q} - \mathbf{w}^H \mathbf{a}_1 s_1 - \mathbf{w}^H E[\mathbf{q}])^2] \\ &= E[(\mathbf{w}^H \mathbf{q})^2] = E[\mathbf{w}^H \mathbf{q} \mathbf{q}^H \mathbf{w}] = \mathbf{w}^H \underbrace{E[\mathbf{q} \mathbf{q}^H]}_{\mathbf{P}_q} \mathbf{w} \\ &= \boxed{\mathbf{w}^H \mathbf{P}_q \mathbf{w}} \rightarrow \text{filter output noise variance/power} \end{aligned}$$

$\mathbf{P}_q = E[\mathbf{q} \mathbf{q}^H]$
is the cross
power spectral
density matrix of
the total noise

Multi-microphone noise reduction

- Multi-microphone noise reduction filters:
 - Spatial filters (DS, MVDR and LCMV)
 - Spatio-temporal filters (MWF)
- Spatial filters (also known as beamformers):
 - Aim at suppressing the noise, while at the same time leaving the target signal undistorted.
- Spatio-temporal filters:
 - Provide more aggressive noise suppression at the expense of some target signal distortions.

Multi-microphone noise reduction: delay and sum (DS) filter (Flanagan 1993)

Estimation of s_1 with an unconstrained quadratic optimization problem also known as least squares.

$$\hat{s}_1 = \arg \min_{s_1} \|\mathbf{y} - \mathbf{a}_1 s_1\|_2^2$$

deterministic



Multi-microphone noise reduction: delay and sum (DS) filter (Flanagan 1993)

Estimation of s_1 with an unconstrained quadratic optimization problem also known as least squares.

$$\hat{s}_1 = \arg \min_{s_1} \|\mathbf{y} - \mathbf{a}_1 s_1\|_2^2 \rightarrow \hat{s}_1 = \frac{1}{\mathbf{a}_1^H \mathbf{a}_1} \mathbf{a}_1^H \mathbf{y}$$

deterministic



Multi-microphone noise reduction: delay and sum (DS) filter (Flanagan 1993)

Estimation of s_1 with an unconstrained quadratic optimization problem also known as least squares.

$$\hat{s}_1 = \arg \min_{s_1} \|\mathbf{y} - \mathbf{a}_1 s_1\|_2^2 \rightarrow \hat{s}_1 = \frac{1}{\mathbf{a}_1^H \mathbf{a}_1} \mathbf{a}_1^H \mathbf{y}$$

deterministic

$$\hat{\mathbf{w}}^H = \frac{1}{\mathbf{a}_1^H \mathbf{a}_1} \mathbf{a}_1^H : \text{ DS filter}$$

Proof:

$$\begin{aligned} \frac{\partial}{\partial \hat{s}_1^*} \|\mathbf{y} - \mathbf{a}_1 \hat{s}_1\|_2^2 &= 0 \rightarrow \frac{\partial}{\partial \hat{s}_1^*} (\mathbf{y} - \mathbf{a}_1 \hat{s}_1)^H (\mathbf{y} - \mathbf{a}_1 \hat{s}_1) = 0 \rightarrow \\ \rightarrow \frac{\partial}{\partial \hat{s}_1^*} [\mathbf{y}^H \mathbf{y} - \mathbf{y}^H \mathbf{a}_1 \hat{s}_1 - \hat{s}_1^* \mathbf{a}_1^H \mathbf{y} + \hat{s}_1^* \hat{s}_1 \mathbf{a}_1^H \mathbf{a}_1] &= 0 \rightarrow -\mathbf{a}_1^H \mathbf{y} + \hat{s}_1 \mathbf{a}_1^H \mathbf{a}_1 = 0 \rightarrow \hat{s}_1 = \frac{1}{\mathbf{a}_1^H \mathbf{a}_1} \mathbf{a}_1^H \mathbf{y} \end{aligned}$$

Multi-microphone noise reduction: delay and sum (DS) filter (Flanagan 1993)

Estimation of s_1 with an unconstrained quadratic optimization problem also known as least squares.

$$\hat{s}_1 = \arg \min_{s_1} \|\mathbf{y} - \mathbf{a}_1 s_1\|_2^2 \rightarrow \hat{s}_1 = \frac{1}{\mathbf{a}_1^H \mathbf{a}_1} \mathbf{a}_1^H \mathbf{y}$$

deterministic

$$\hat{\mathbf{w}}^H = \frac{1}{\mathbf{a}_1^H \mathbf{a}_1} \mathbf{a}_1^H : \text{ DS filter}$$

Estimation of \mathbf{W} with a constrained quadratic optimization problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} \mathbf{w}^H \mathbf{w} \quad \text{s.t.} \quad \mathbf{w}^H \mathbf{a}_1 = 1 \rightarrow \hat{\mathbf{w}} = \frac{1}{\mathbf{a}_1^H \mathbf{a}_1} \mathbf{a}_1$$

Multi-microphone noise reduction: delay and sum (DS) filter (Flanagan 1993)

Estimation of \mathbf{W}
with a constrained
quadratic optimization
problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} \mathbf{w}^H \mathbf{w} \quad \text{s.t.} \quad \mathbf{w}^H \mathbf{a}_1 = 1 \rightarrow \hat{\mathbf{w}} = \frac{1}{\mathbf{a}_1^H \mathbf{a}_1} \mathbf{a}_1$$

Proof: $\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} \mathbf{w}^H \mathbf{w} + 2\Re \left\{ \lambda (\mathbf{w}^H \mathbf{a}_1 - 1) \right\}$

Multi-microphone noise reduction: delay and sum (DS) filter (Flanagan 1993)

Estimation of \mathbf{W}
with a constrained
quadratic optimization
problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} \mathbf{w}^H \mathbf{w} \quad \text{s.t.} \quad \mathbf{w}^H \mathbf{a}_1 = 1 \rightarrow \hat{\mathbf{w}} = \frac{1}{\mathbf{a}_1^H \mathbf{a}_1} \mathbf{a}_1$$

Proof: $\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} \mathbf{w}^H \mathbf{w} + 2\Re \left\{ \lambda \left(\mathbf{w}^H \mathbf{a}_1 - 1 \right) \right\}$

$$\frac{\partial}{\partial \hat{\mathbf{w}}^*} [\hat{\mathbf{w}}^H \hat{\mathbf{w}} + \lambda (1 - \hat{\mathbf{w}}^H \mathbf{a}_1) + \lambda^* (1 - \mathbf{a}_1^H \hat{\mathbf{w}})] = 0 \rightarrow \hat{\mathbf{w}} - \lambda \mathbf{a}_1 = 0 \rightarrow \hat{\mathbf{w}} = \lambda \mathbf{a}_1 \quad \left. \begin{array}{l} \hat{\mathbf{w}}^H \mathbf{a}_1 = 1 \end{array} \right\} \lambda = \frac{1}{\mathbf{a}^H \mathbf{a}}$$

Multi-microphone noise reduction: delay and sum (DS) filter (Flanagan 1993)

Estimation of \mathbf{W}
 with a constrained
 quadratic optimization
 problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} \mathbf{w}^H \mathbf{w} \quad \text{s.t.} \quad \mathbf{w}^H \mathbf{a}_1 = 1 \rightarrow \hat{\mathbf{w}} = \frac{1}{\mathbf{a}_1^H \mathbf{a}_1} \mathbf{a}_1$$

Proof: $\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} \mathbf{w}^H \mathbf{w} + 2\Re \left\{ \lambda \left(\mathbf{w}^H \mathbf{a}_1 - 1 \right) \right\}$

$$\frac{\partial}{\partial \hat{\mathbf{w}}^*} [\hat{\mathbf{w}}^H \hat{\mathbf{w}} + \lambda (1 - \hat{\mathbf{w}}^H \mathbf{a}_1) + \lambda^* (1 - \mathbf{a}_1^H \hat{\mathbf{w}})] = 0 \rightarrow \hat{\mathbf{w}} - \lambda \mathbf{a}_1 = 0 \rightarrow \hat{\mathbf{w}} = \lambda \mathbf{a}_1$$

$$\hat{\mathbf{w}} = \frac{1}{\mathbf{a}_1^H \mathbf{a}_1} \mathbf{a}_1$$

$\hat{\mathbf{w}}^H \mathbf{a}_1 = 1 \quad \lambda = \frac{1}{\mathbf{a}^H \mathbf{a}}$

Multi-microphone noise reduction: minimum variance distortionless response (MVDR) filter (Capon 1969)

Estimation of s_1 with an unconstrained quadratic optimization problem also known as weighted least squares.

$$\hat{s}_1 = \arg \min_{s_1} \|\mathbf{P}_{\mathbf{q}}^{-1/2}(\mathbf{y} - \mathbf{a}_1 s_1)\|_2^2$$

deterministic



Multi-microphone noise reduction: minimum variance distortionless response (MVDR) filter (Capon 1969)

Estimation of s_1 with an unconstrained quadratic optimization problem also known as weighted least squares.

$$\hat{s}_1 = \arg \min_{s_1} \|\mathbf{P}_{\mathbf{q}}^{-1/2}(\mathbf{y} - \mathbf{a}_1 s_1)\|_2^2 \rightarrow \hat{s}_1 = \frac{1}{\mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1} \mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{y}$$

deterministic

$$\hat{\mathbf{w}}^H = \frac{1}{\mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1} \mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} : \text{MVDR filter}$$

Multi-microphone noise reduction: minimum variance distortionless response (MVDR) filter (Capon 1969)

Estimation of s_1 with an unconstrained quadratic optimization problem also known as weighted least squares.

$$\hat{s}_1 = \arg \min_{s_1} \|\mathbf{P}_{\mathbf{q}}^{-1/2}(\mathbf{y} - \mathbf{a}_1 s_1)\|_2^2 \rightarrow \hat{s}_1 = \frac{1}{\mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1} \mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{y}$$

deterministic

$$\hat{\mathbf{w}}^H = \frac{1}{\mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1} \mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} : \text{MVDR filter}$$

Estimation of \mathbf{W} with a constrained quadratic optimization problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} \mathbf{w}^H \mathbf{P}_{\mathbf{q}} \mathbf{w} \quad \text{s.t.} \quad \mathbf{w}^H \mathbf{a}_1 = 1 \rightarrow \hat{\mathbf{w}} = \frac{1}{\mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1} \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1$$

Multi-microphone noise reduction: linearly constrained minimum variance (LCMV) filter (Frost 1972)

Estimation of \mathbf{W}
with a constrained
quadratic optimization
problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} \mathbf{w}^H \mathbf{P}_q \mathbf{w} \quad \text{s.t.} \quad \mathbf{w}^H \boldsymbol{\Lambda} = \mathbf{f}^H$$

Multi-microphone noise reduction: linearly constrained minimum variance (LCMV) filter (Frost 1972)

Estimation of \mathbf{w}
with a constrained
quadratic optimization
problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} \mathbf{w}^H \mathbf{P}_q \mathbf{w} \quad \text{s.t.}$$

$$\mathbf{w}^H \Lambda = \mathbf{f}^H$$

example

$$\mathbf{w}^H \mathbf{a}_1 = 1$$

$$\mathbf{w}^H \mathbf{a}_2 = 0$$

Multi-microphone noise reduction: linearly constrained minimum variance (LCMV) filter (Frost 1972)

Estimation of \mathbf{w}
with a constrained
quadratic optimization
problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} \mathbf{w}^H \mathbf{P}_q \mathbf{w} \quad \text{s.t.}$$

$$\mathbf{w}^H \boldsymbol{\Lambda} = \mathbf{f}^H$$

example

$$\begin{aligned}\mathbf{w}^H \mathbf{a}_1 &= 1 \\ \mathbf{w}^H \mathbf{a}_2 &= 0\end{aligned}$$

$$\mathbf{w}^H \underbrace{[\mathbf{a}_1 \quad \mathbf{a}_2]}_{\boldsymbol{\Lambda}_{M \times 2}} = \underbrace{[1 \quad 0]}_{\mathbf{f}^H_{1 \times 2}}$$

Multi-microphone noise reduction: linearly constrained minimum variance (LCMV) filter (Frost 1972)

Estimation of \mathbf{w}
with a constrained
quadratic optimization $\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} \mathbf{w}^H \mathbf{P}_q \mathbf{w}$ s.t.

$$\mathbf{w}^H \boldsymbol{\Lambda} = \mathbf{f}^H$$

example

$$\begin{aligned}\mathbf{w}^H \mathbf{a}_1 &= 1 \\ \mathbf{w}^H \mathbf{a}_2 &= 0\end{aligned}$$

$$\mathbf{w}^H \underbrace{[\mathbf{a}_1 \quad \mathbf{a}_2]}_{\boldsymbol{\Lambda}_{M \times 2}} = \underbrace{[1 \quad 0]}_{\mathbf{f}^H_{1 \times 2}}$$

M-2 degrees of freedom
for noise reduction

Multi-microphone noise reduction: linearly constrained minimum variance (LCMV) filter (Frost 1972)

Estimation of \mathbf{w}
with a constrained
quadratic optimization
problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} \mathbf{w}^H \mathbf{P}_q \mathbf{w} \quad \text{s.t.}$$

$$\mathbf{w}^H \boldsymbol{\Lambda} = \mathbf{f}^H$$

example

$$\begin{aligned}\mathbf{w}^H \mathbf{a}_1 &= 1 \\ \mathbf{w}^H \mathbf{a}_2 &= 0\end{aligned}$$

$$\mathbf{w}^H \underbrace{[\mathbf{a}_1 \quad \mathbf{a}_2]}_{\boldsymbol{\Lambda}_{M \times 2}} = \underbrace{[1 \quad 0]}_{\mathbf{f}^H_{1 \times 2}}$$

More constraints means more control, but less total noise reduction!

M-2 degrees of freedom
for noise reduction

Multi-microphone noise reduction: linearly constrained minimum variance (LCMV) filter (Frost 1972)

Estimation of \mathbf{w}
with a constrained
quadratic optimization
problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} \mathbf{w}^H \mathbf{P}_q \mathbf{w} \quad \text{s.t.} \quad \mathbf{w}^H \boldsymbol{\Lambda} = \mathbf{f}^H \rightarrow$$

$$\hat{\mathbf{w}} = \mathbf{P}_q^{-1} \boldsymbol{\Lambda} \left(\boldsymbol{\Lambda}^H \mathbf{P}_q^{-1} \boldsymbol{\Lambda} \right)^{-1} \mathbf{f}$$

- The MVDR filter is a special case of the LCMV filter which provides the minimum filter output noise power, but does not control how the noise sources will be suppressed.
- The LCMV optimization problem gives you more control to design the filter vector than the MVDR optimization problem.
- More constraints means more control, but less total noise reduction!

Multi-microphone noise reduction: multi-channel Wiener filter (MWF) (Brandstein 2001)

Estimation of \mathbf{W}
with an unconstrained
LMSE optimization
problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} E [|\mathbf{w}^H \mathbf{y} - s_1|^2]$$

stochastic



Multi-microphone noise reduction: multi-channel Wiener filter (MWF) (Brandstein 2001)

Estimation of \mathbf{W}
with an unconstrained
LMSE optimization
problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} E[|\mathbf{w}^H \mathbf{y} - s_1|^2] \xrightarrow{\text{stochastic}} \hat{\mathbf{w}} = p_1 \mathbf{P}_y^{-1} \mathbf{a}$$

Multi-microphone noise reduction: multi-channel Wiener filter (MWF) (Brandstein 2001)

Estimation of \mathbf{W}
with an unconstrained
LMSE optimization
problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} E[|\mathbf{w}^H \mathbf{y} - s_1|^2] \rightarrow \hat{\mathbf{w}} = p_1 \mathbf{P}_y^{-1} \mathbf{a}$$

↑
stochastic

Proof:

$$\begin{aligned} \frac{\partial}{\partial \hat{\mathbf{w}}^*} \{E[|\hat{\mathbf{w}}^H \mathbf{y} - s_1|^2]\} &= 0 \rightarrow \frac{\partial}{\partial \hat{\mathbf{w}}^*} \{E[(\hat{\mathbf{w}}^H \mathbf{y} - s_1)^* (\hat{\mathbf{w}}^H \mathbf{y} - s_1)]\} = 0 \rightarrow \\ &\rightarrow \frac{\partial}{\partial \hat{\mathbf{w}}^*} \{E[s_1^* s_1 - s_1^* \hat{\mathbf{w}}^H \mathbf{y} - s_1 \mathbf{y}^H \hat{\mathbf{w}} + \hat{\mathbf{w}}^H \mathbf{y} \mathbf{y}^H \hat{\mathbf{w}}]\} = 0 \rightarrow \\ &\rightarrow \frac{\partial}{\partial \hat{\mathbf{w}}^*} \{E[s_1^* s_1] - E[s_1^* \hat{\mathbf{w}}^H \mathbf{y}] - E[s_1 \mathbf{y}^H \hat{\mathbf{w}}] + E[\hat{\mathbf{w}}^H \mathbf{y} \mathbf{y}^H \hat{\mathbf{w}}]\} = 0 \rightarrow \\ &\rightarrow -\frac{\partial}{\partial \hat{\mathbf{w}}^*} E[s_1^* \hat{\mathbf{w}}^H \mathbf{y}] + \frac{\partial}{\partial \hat{\mathbf{w}}^*} E[\hat{\mathbf{w}}^H \mathbf{y} \mathbf{y}^H \hat{\mathbf{w}}] = 0 \rightarrow -\frac{\partial}{\partial \hat{\mathbf{w}}^*} \hat{\mathbf{w}}^H E[s_1^* \mathbf{y}] + \frac{\partial}{\partial \hat{\mathbf{w}}^*} \hat{\mathbf{w}}^H E[\mathbf{y} \mathbf{y}^H] \hat{\mathbf{w}} = 0 \rightarrow \\ &\rightarrow E[s_1^* \mathbf{y}] + \mathbf{P}_y \hat{\mathbf{w}} = 0 \rightarrow E[s_1^* s_1 \mathbf{a}_1 + s_1^* \mathbf{n}] + \mathbf{P}_y \hat{\mathbf{w}} = 0 \rightarrow E[s_1^* s_1] \mathbf{a}_1 + E[s_1^* \mathbf{n}] + \mathbf{P}_y \hat{\mathbf{w}} = 0 \xrightarrow{E[s_1^* \mathbf{n}] = 0} \hat{\mathbf{w}} = p_1 \mathbf{P}_y^{-1} \mathbf{a}_1 \end{aligned}$$

Multi-microphone noise reduction: multi-channel Wiener filter (MWF) (Brandstein 2001)

Estimation of \mathbf{W}
with an unconstrained
LMSE optimization
problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} E [| \mathbf{w}^H \mathbf{y} - s_1 |^2] \xrightarrow{\text{stochastic}} \hat{\mathbf{w}} = p_1 \mathbf{P}_y^{-1} \mathbf{a}$$

Estimation of s_1
with an unconstrained
MSE optimization
problem.

$$\hat{s}_1 = \arg \min_{s_1} E [| \hat{s}_1 - s_1 |^2] = E [s_1 | \mathbf{y}]$$

$$\begin{aligned} s_1 &\sim N(0, p_1) \\ \mathbf{n} &\sim N(\mathbf{0}, \mathbf{P}_n) \end{aligned} \xrightarrow{} \hat{s}_1 = \boxed{p_1 \mathbf{a}_1^H \mathbf{P}_y^{-1} \mathbf{y}} \quad \hat{\mathbf{w}}^H$$

Multi-microphone noise reduction: multi-channel Wiener filter (MWF) (Brandstein 2001)

Estimation of \mathbf{W}
with an unconstrained
LMSE optimization
problem.

$$\begin{aligned} \hat{\mathbf{w}} &= \arg \min_{\mathbf{w}} \mathbb{E} [| \mathbf{w}^H \mathbf{y} - s_1 |^2] \rightarrow \hat{\mathbf{w}} = p_1 \mathbf{P}_{\mathbf{y}}^{-1} \mathbf{a}_1 \\ &\quad \text{stochastic} \\ &= p_1 (p_1 \mathbf{a}_1 \mathbf{a}_1^H + \mathbf{P}_{\mathbf{q}})^{-1} \mathbf{a}_1 \\ &= \frac{p_1}{p_1 + \underbrace{(\mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1)^{-1}}_{\text{MVDR output noise power}}} \underbrace{\frac{1}{\mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1} \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1}_{\text{MVDR filter}} \end{aligned}$$

Multi-microphone noise reduction: multi-channel Wiener filter (MWF) (Brandstein 2001)

Estimation of \mathbf{W}
with an unconstrained
LMSE optimization
problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} E[|\mathbf{w}^H \mathbf{y} - s_1|^2]$$

↑
stochastic

$$\begin{aligned}
 & \hat{\mathbf{w}} = p_1 \mathbf{P}_{\mathbf{y}}^{-1} \mathbf{a}_1 \\
 &= p_1 (p_1 \mathbf{a}_1 \mathbf{a}_1^H + \mathbf{P}_{\mathbf{q}})^{-1} \mathbf{a}_1 \\
 &= \frac{p_1}{p_1 + \underbrace{(\mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1)^{-1}}_{\text{MVDR output noise power}}} \underbrace{\frac{1}{\mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1} \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1}_{\text{MVDR filter}}
 \end{aligned}$$

Sherman Morrison formula
 $\mathbf{P}_{\mathbf{q}}^{-1} - \frac{p_1 \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1 \mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1}}{1 + p_1 \mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1}$

Multi-microphone noise reduction: multi-channel Wiener filter (MWF) (Brandstein 2001)

Estimation of \mathbf{W}
with an unconstrained
LMSE optimization
problem.

$$\begin{aligned} \hat{\mathbf{w}} &= \arg \min_{\mathbf{w}} \mathbb{E} [| \mathbf{w}^H \mathbf{y} - s_1 |^2] \rightarrow \hat{\mathbf{w}} = p_1 \mathbf{P}_{\mathbf{y}}^{-1} \mathbf{a}_1 \\ &= p_1 (p_1 \mathbf{a}_1 \mathbf{a}_1^H + \mathbf{P}_{\mathbf{q}})^{-1} \mathbf{a}_1 \\ &= \frac{p_1}{p_1 + \underbrace{(\mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1)^{-1}}_{\text{MVDR output noise power}}} \underbrace{\frac{1}{\mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1} \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1}_{\text{MVDR filter}} \end{aligned}$$

↑
stochastic

↓

single channel Wiener filter (SWF)

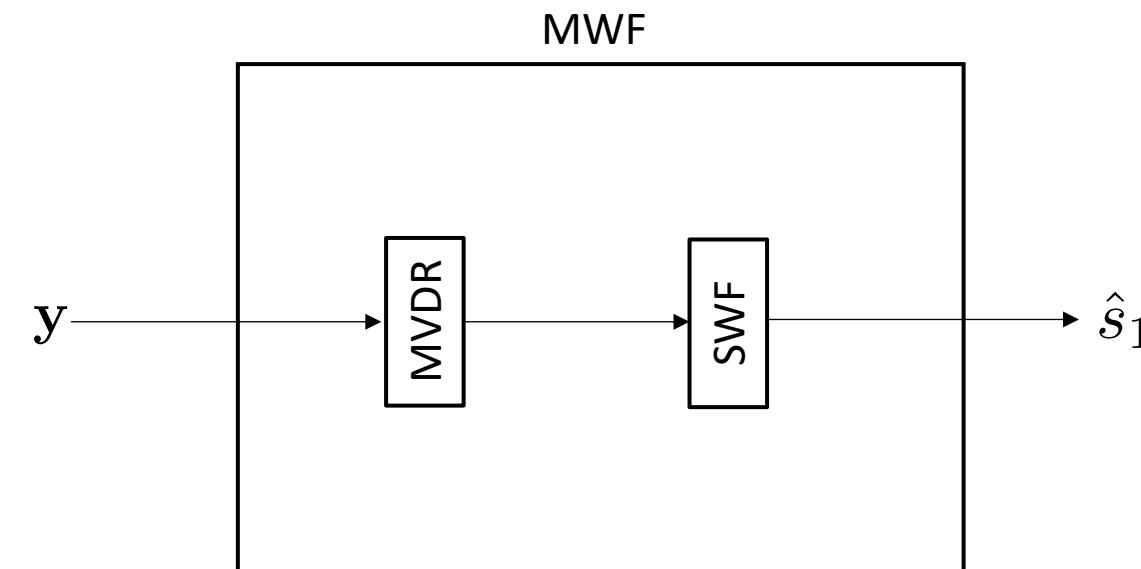
Multi-microphone noise reduction: multi-channel Wiener filter (MWF) (Brandstein 2001)

Estimation of \mathbf{W}
 with an unconstrained
 LMSE optimization
 problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} E [| \mathbf{w}^H \mathbf{y} - s_1 |^2] \rightarrow \hat{\mathbf{w}} = p_1 \mathbf{P}_{\mathbf{y}}^{-1} \mathbf{a}_1$$

stochastic

$$\begin{aligned} &= p_1 (p_1 \mathbf{a}_1 \mathbf{a}_1^H + \mathbf{P}_{\mathbf{q}})^{-1} \mathbf{a}_1 \\ &= \frac{p_1}{p_1 + \underbrace{(\mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1)^{-1}}_{\text{MVDR output noise power}}} \underbrace{\frac{1}{\mathbf{a}_1^H \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1} \mathbf{P}_{\mathbf{q}}^{-1} \mathbf{a}_1}_{\text{MVDR filter}} \end{aligned}$$



single channel Wiener filter (SWF)

Multi-microphone noise reduction: multi-channel Wiener filter (MWF) (Brandstein 2001)

Estimation of \mathbf{W}
with an unconstrained
LMSE optimization
problem.

$$\hat{\mathbf{w}} = \arg \min_{\mathbf{w}} E [|\mathbf{w}^H \mathbf{y} - s_1|^2] \rightarrow \hat{\mathbf{w}} = p_1 \mathbf{P}_{\mathbf{y}}^{-1} \mathbf{a}_1$$

↑
stochastic

- Typically, the MWF filter achieves extra noise reduction compared to the MVDR filter.
- Unlike the MVDR filter, the MWF filter distorts the target signal. This is due to the fact that

$$\mathbf{w}^H \mathbf{a}_1 = p_1 \mathbf{a}_1^H \mathbf{P}_{\mathbf{y}}^{-1} \mathbf{a}_1 \neq 1$$

Multi-microphone source separation

Multi-microphone source separation

- Typically, in multi-microphone source separation all point sources are considered targets. That is,

$$\mathbf{y} = \underbrace{\sum_{i=1}^r \mathbf{a}_i s_i}_{\text{targets}} + \underbrace{\mathbf{n}}_{\text{noise}} \rightarrow \mathbf{y} = \underbrace{\mathbf{A}\mathbf{s}}_{\text{targets}} + \underbrace{\mathbf{n}}_{\text{noise}}$$

- The goal is to estimate the targets \mathbf{S} from the noisy DFT coefficients \mathbf{y} . That is,

$$\hat{\mathbf{s}} = f(\mathbf{y}),$$

where $f(\cdot)$ is the filter function, where in this presentation is linear. That is,

$$\hat{\mathbf{s}} = \mathbf{W}^H \mathbf{y},$$

where \mathbf{W} is the filter matrix.

- We want to find a \mathbf{W} such that $\hat{\mathbf{s}} = \underbrace{\mathbf{W}^H \mathbf{A}\mathbf{s}}_{\approx \mathbf{s}} + \underbrace{\mathbf{W}^H \mathbf{n}}_{\approx 0}$,

Multi-microphone source separation

- Typically, in multi-microphone source separation all point sources are considered targets. That is,

$$\mathbf{y} = \sum_{i=1}^r \underbrace{\mathbf{a}_i s_i}_{\text{targets}} + \underbrace{\mathbf{n}}_{\text{noise}} \rightarrow \mathbf{y} = \underbrace{\mathbf{A}\mathbf{s}}_{\text{targets}} + \underbrace{\mathbf{n}}_{\text{noise}}$$

- The goal is to estimate the targets \mathbf{S} from the noisy DFT coefficients \mathbf{y} . That is,

$$\hat{\mathbf{s}} = f(\mathbf{y}),$$

where $f(\cdot)$ is the filter function, where in this presentation is linear. That is,

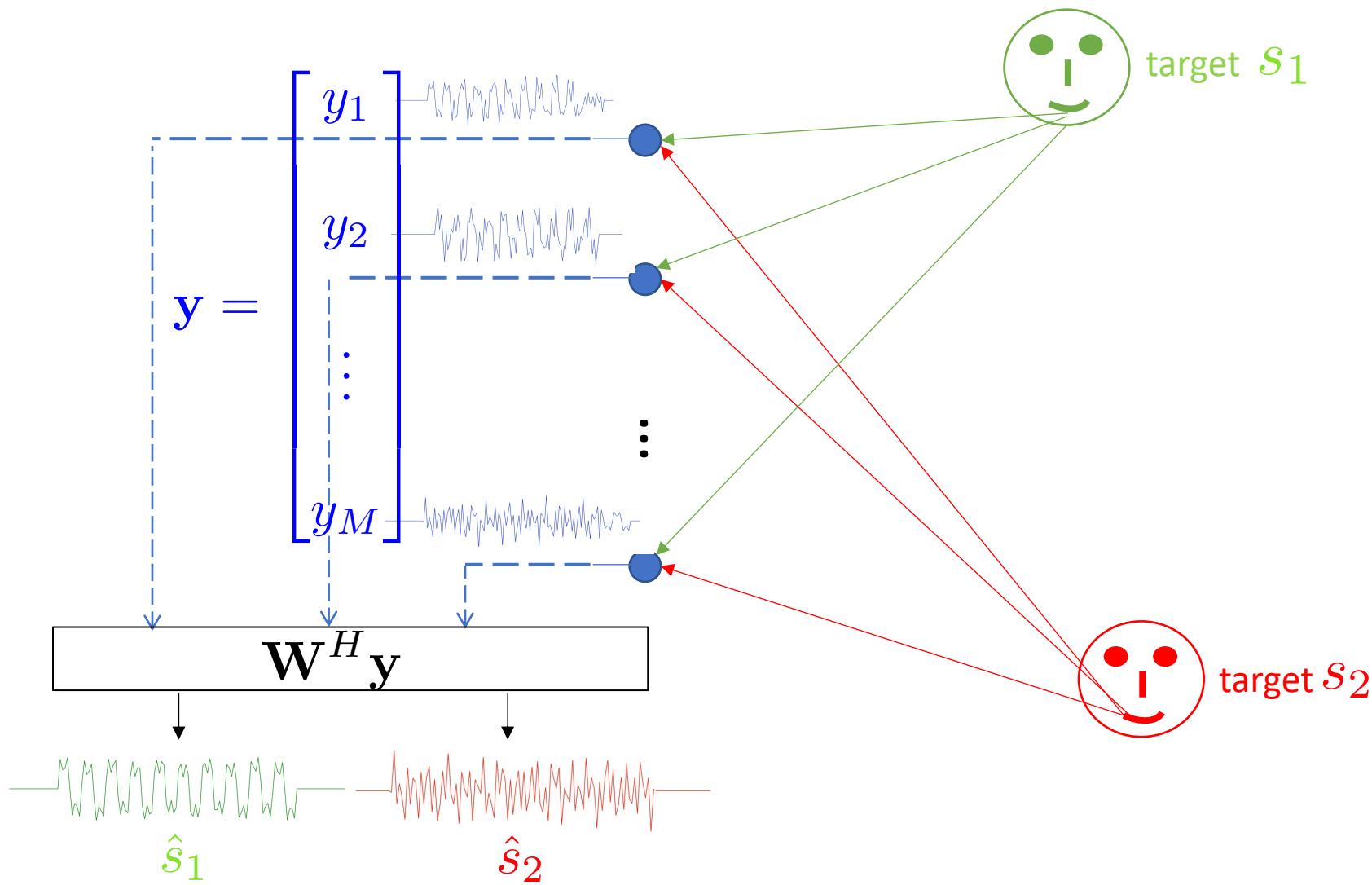
$$\hat{\mathbf{s}} = \mathbf{W}^H \mathbf{y},$$

where \mathbf{W} is the filter matrix.

- We want to find a \mathbf{W} such that $\hat{\mathbf{s}} = \underbrace{\mathbf{W}^H \mathbf{A}\mathbf{s}}_{\approx \mathbf{s}} + \underbrace{\mathbf{W}^H \mathbf{n}}_{\approx 0}$,

$$\hat{\mathbf{s}} = \underbrace{\mathbf{W}^H \mathbf{A}\mathbf{s}}_{\approx \mathbf{s}} + \underbrace{\mathbf{W}^H \mathbf{n}}_{\approx 0}$$

Multi-microphone source separation



Multi-microphone source separation

1st Method
Least squares

$$\hat{\mathbf{s}} = \arg \min_{\mathbf{s}} \|\mathbf{y} - \mathbf{A}\mathbf{s}\|_2^2 \rightarrow \hat{\mathbf{s}} = (\mathbf{A}^H \mathbf{A})^{-1} \mathbf{A}^H \mathbf{y}$$

deterministic

2nd Method
Weighted least
squares

$$\hat{\mathbf{s}} = \arg \min_{\mathbf{s}} \|\mathbf{P}_n^{-\frac{1}{2}} (\mathbf{y} - \mathbf{A}\mathbf{s})\|_2^2 \rightarrow \hat{\mathbf{s}} = (\mathbf{A}^H \mathbf{P}_n^{-1} \mathbf{A})^{-1} \mathbf{A}^H \mathbf{P}_n^{-1} \mathbf{y}$$

deterministic

3d Method
MMSE

$$\hat{\mathbf{s}} = \arg \min_{\hat{\mathbf{s}}} E[\|\hat{\mathbf{s}} - \mathbf{s}\|_2^2] = E[\mathbf{s} | \mathbf{y}]$$

stochastic

$$\frac{\mathbf{s} \sim N(\mathbf{0}, \mathbf{P}) \quad \mathbf{n} \sim N(\mathbf{0}, \mathbf{P}_n)}{\hat{\mathbf{s}} = \mathbf{P} \mathbf{A}^H \mathbf{P}_y^{-1} \mathbf{y}}$$

Multi-microphone source separation

1st Method
Least squares

$$\hat{\mathbf{s}} = \arg \min_{\mathbf{s}} \|\mathbf{y} - \mathbf{A}\mathbf{s}\|_2^2 \rightarrow \hat{\mathbf{s}} = (\mathbf{A}^H \mathbf{A})^{-1} \mathbf{A}^H \mathbf{y}$$

deterministic

$$\mathbf{W}^H \mathbf{A} = \mathbf{I}$$

2nd Method
Weighted least
squares

$$\hat{\mathbf{s}} = \arg \min_{\mathbf{s}} \|\mathbf{P}_n^{-\frac{1}{2}} (\mathbf{y} - \mathbf{A}\mathbf{s})\|_2^2 \rightarrow \hat{\mathbf{s}} = (\mathbf{A}^H \mathbf{P}_n^{-1} \mathbf{A})^{-1} \mathbf{A}^H \mathbf{P}_n^{-1} \mathbf{y}$$

deterministic

3d Method
MMSE

$$\hat{\mathbf{s}} = \arg \min_{\hat{\mathbf{s}}} E[\|\hat{\mathbf{s}} - \mathbf{s}\|_2^2] = E[\mathbf{s} | \mathbf{y}]$$

stochastic

$$\mathbf{s} \sim N(\mathbf{0}, \mathbf{P})$$
$$\mathbf{n} \sim N(\mathbf{0}, \mathbf{P}_n) \xrightarrow{} \hat{\mathbf{s}} = \mathbf{P} \mathbf{A}^H \mathbf{P}_y^{-1} \mathbf{y}$$

Estimation of multi-microphone signal model parameters

Maximum likelihood estimation of \mathbf{P}_y

$$\hat{\mathbf{P}}_y = \arg \min_{\mathbf{P}_y} -\ln p(\{\mathbf{y}_1, \dots, \mathbf{y}_N\} | \mathbf{P}_y)$$

Maximum likelihood estimation of \mathbf{P}_y

$\mathbf{y}_i, i = 1, \dots, N,$ indipendent

$$\hat{\mathbf{P}}_y = \arg \min_{\mathbf{P}_y} - \ln p(\{\mathbf{y}_1, \dots, \mathbf{y}_N\} | \mathbf{P}_y)$$

$$= \arg \min_{\mathbf{P}_y} - \ln \prod_{i=1}^N p(\mathbf{y}_i | \mathbf{P}_y)$$

$$= \arg \min_{\mathbf{P}_y} - \sum_{i=1}^N \ln p(\mathbf{y}_i | \mathbf{P}_y)$$

Maximum likelihood estimation of \mathbf{P}_y

$\mathbf{y}_i, i = 1, \dots, N$, indipendent

iid $\mathbf{y}_i \sim \mathcal{N}(\mathbf{0}, \mathbf{P}_y)$

$$p(\mathbf{y}_i | \mathbf{P}_y) = \frac{1}{|\pi \mathbf{P}_y|} \exp(-\mathbf{y}_i^H \mathbf{P}_y^{-1} \mathbf{y}_i)$$

$$\hat{\mathbf{P}}_y = \arg \min_{\mathbf{P}_y} -\ln p(\{\mathbf{y}_1, \dots, \mathbf{y}_N\} | \mathbf{P}_y)$$

$$= \arg \min_{\mathbf{P}_y} -\ln \prod_{i=1}^N p(\mathbf{y}_i | \mathbf{P}_y)$$

$$= \arg \min_{\mathbf{P}_y} -\sum_{i=1}^N \ln p(\mathbf{y}_i | \mathbf{P}_y)$$

$$= \arg \min_{\mathbf{P}_y} -\sum_{i=1}^N [\ln(1) - \ln(\pi^M) - \ln(|\mathbf{P}_y|) + \ln(\exp(-\mathbf{y}_i^H \mathbf{P}_y^{-1} \mathbf{y}_i))]$$

Maximum likelihood estimation of \mathbf{P}_y

$\mathbf{y}_i, i = 1, \dots, N$, indipendent

$$\hat{\mathbf{P}}_y = \arg \min_{\mathbf{P}_y} - \ln p(\{\mathbf{y}_1, \dots, \mathbf{y}_N\} | \mathbf{P}_y)$$

$$= \arg \min_{\mathbf{P}_y} - \ln \prod_{i=1}^N p(\mathbf{y}_i | \mathbf{P}_y)$$

$$= \arg \min_{\mathbf{P}_y} - \sum_{i=1}^N \ln p(\mathbf{y}_i | \mathbf{P}_y)$$

$$= \arg \min_{\mathbf{P}_y} \sum_{i=1}^N [\ln(|\mathbf{P}_y|) + \mathbf{y}_i^H \mathbf{P}_y^{-1} \mathbf{y}_i]$$

iid $\mathbf{y}_i \sim \mathcal{N}(\mathbf{0}, \mathbf{P}_y)$

$$p(\mathbf{y}_i | \mathbf{P}_y) = \frac{1}{|\pi \mathbf{P}_y|} \exp(-\mathbf{y}_i^H \mathbf{P}_y^{-1} \mathbf{y}_i)$$

Maximum likelihood estimation of \mathbf{P}_y

$\mathbf{y}_i, i = 1, \dots, N,$ indipendent

iid $\mathbf{y}_i \sim \mathcal{N}(\mathbf{0}, \mathbf{P}_y)$

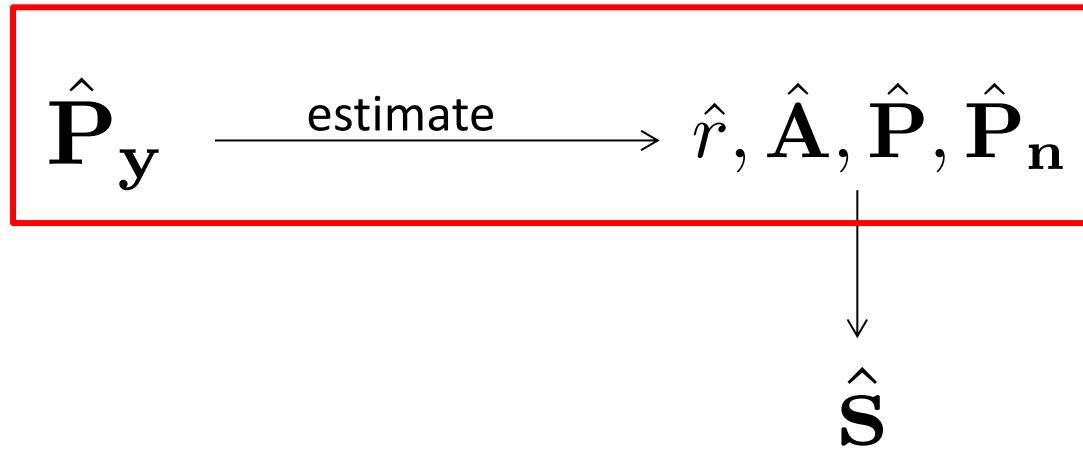
$$p(\mathbf{y}_i | \mathbf{P}_y) = \frac{1}{|\pi \mathbf{P}_y|} \exp(-\mathbf{y}_i^H \mathbf{P}_y^{-1} \mathbf{y}_i)$$

$$\frac{\partial}{\partial \hat{\mathbf{P}}_y} \sum_{i=1}^N \left[\ln(|\hat{\mathbf{P}}_y|) + \mathbf{y}_i^H \hat{\mathbf{P}}_y^{-1} \mathbf{y}_i \right] = \mathbf{0} \rightarrow \sum_{i=1}^N \left[\frac{\partial}{\partial \hat{\mathbf{P}}_y} \ln(|\hat{\mathbf{P}}_y|) + \frac{\partial}{\partial \hat{\mathbf{P}}_y} \mathbf{y}_i^H \hat{\mathbf{P}}_y^{-1} \mathbf{y}_i \right] = \mathbf{0} \rightarrow$$

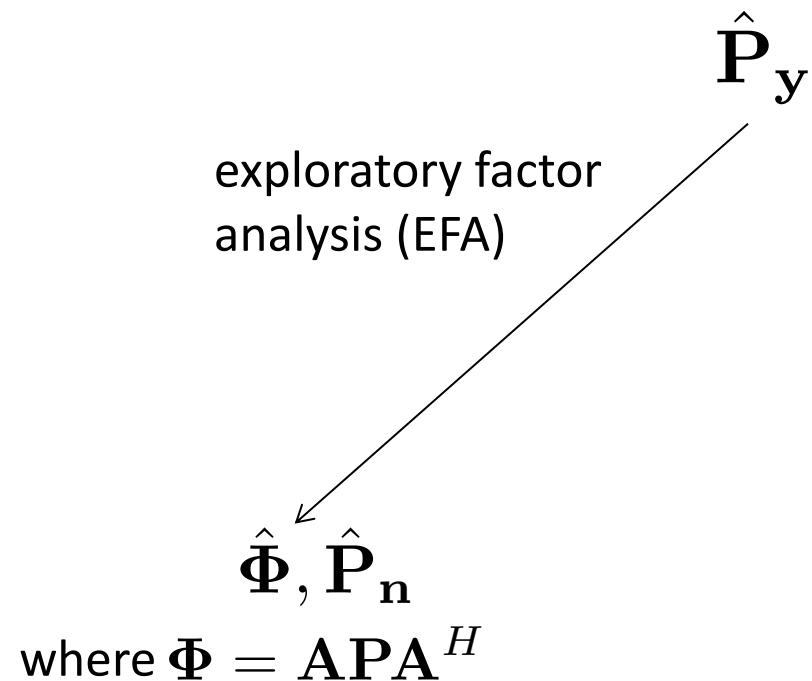
$$\rightarrow \sum_{i=1}^N \left[\hat{\mathbf{P}}_y^{-1} - \hat{\mathbf{P}}_y^{-1} \mathbf{y}_i \mathbf{y}_i^H \hat{\mathbf{P}}_y^{-1} \right] = \mathbf{0} \rightarrow \sum_{i=1}^N \left[\hat{\mathbf{P}}_y^{-1} - \hat{\mathbf{P}}_y^{-1} \mathbf{y}_i \mathbf{y}_i^H \hat{\mathbf{P}}_y^{-1} \right] \hat{\mathbf{P}}_y = \mathbf{0} \hat{\mathbf{P}}_y \rightarrow$$

$$\rightarrow \sum_{i=1}^N \left[\mathbf{I} - \hat{\mathbf{P}}_y^{-1} \mathbf{y}_i \mathbf{y}_i^H \right] = \mathbf{0} \rightarrow \hat{\mathbf{P}}_y = \frac{1}{N} \sum_{i=1}^N \mathbf{y}_i \mathbf{y}_i^H$$

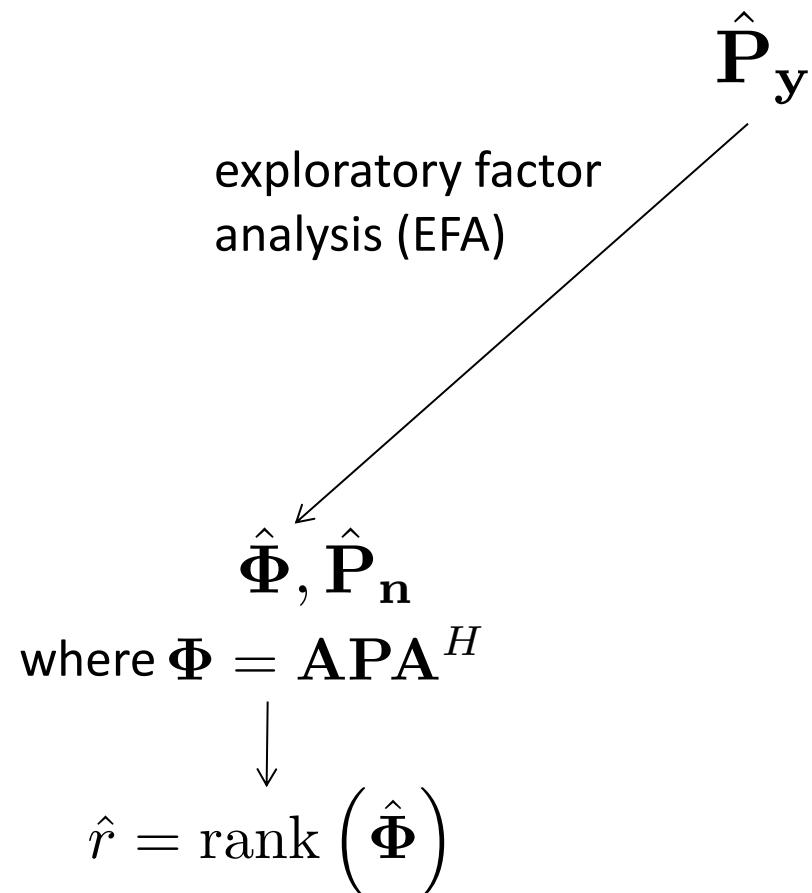
Estimation of multi-microphone signal model parameters $\hat{r}, \hat{\mathbf{A}}, \hat{\mathbf{P}}, \hat{\mathbf{P}}_{\mathbf{n}}$



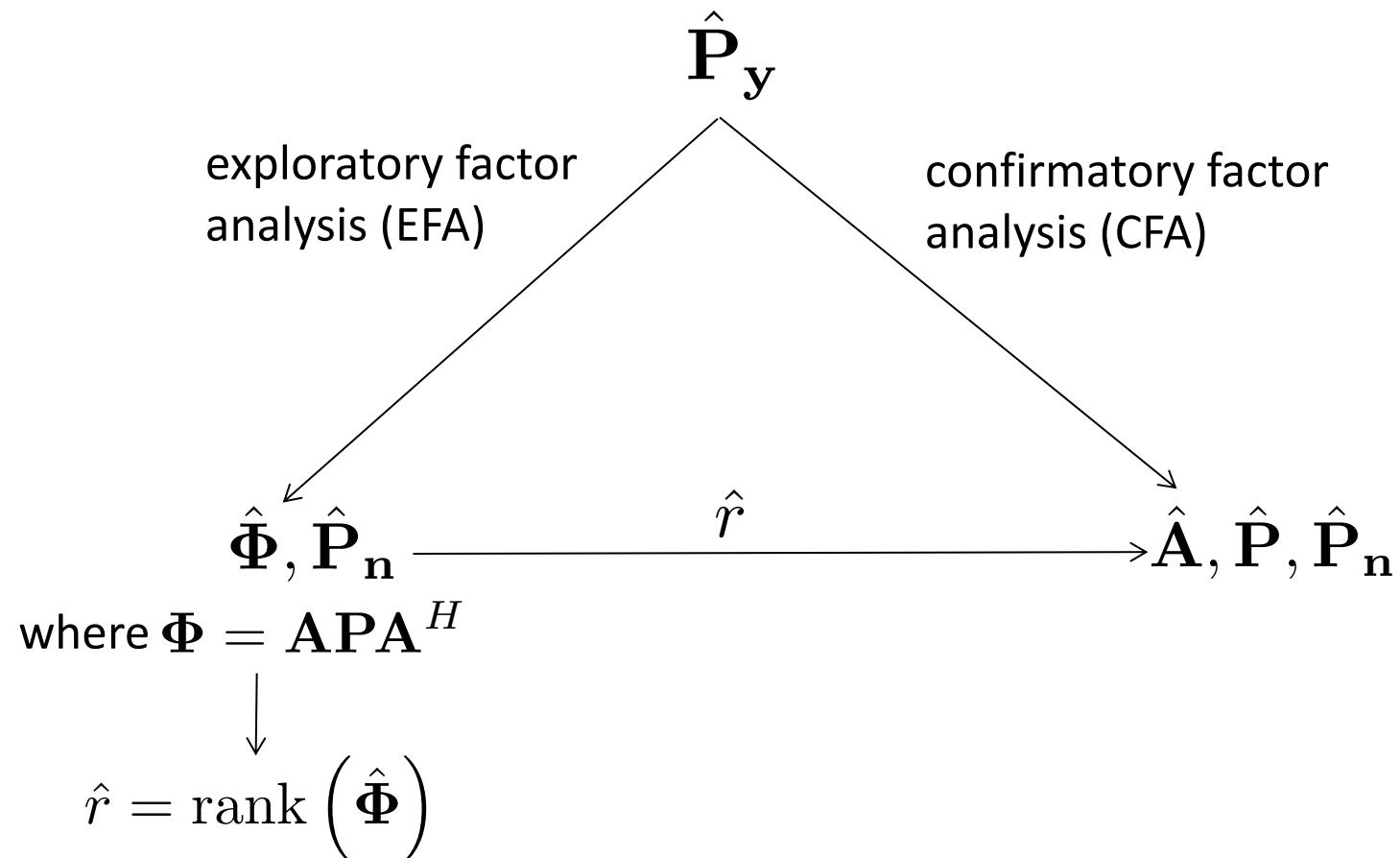
Estimation of multi-microphone signal model parameters $\hat{r}, \hat{\mathbf{A}}, \hat{\mathbf{P}}, \hat{\mathbf{P}}_n$



Estimation of multi-microphone signal model parameters $\hat{r}, \hat{\mathbf{A}}, \hat{\mathbf{P}}, \hat{\mathbf{P}}_n$



Estimation of multi-microphone signal model parameters \hat{r} , $\hat{\mathbf{A}}$, $\hat{\mathbf{P}}$, $\hat{\mathbf{P}}_n$



Exploratory Factor Analysis (EFA) $\hat{\mathbf{P}}_y \xrightarrow{\text{estimate}} \hat{\Phi}, \hat{\mathbf{P}}_n$

Minimum **rank** factor analysis (Lederman 1940, Ten Berge et al. 1991):

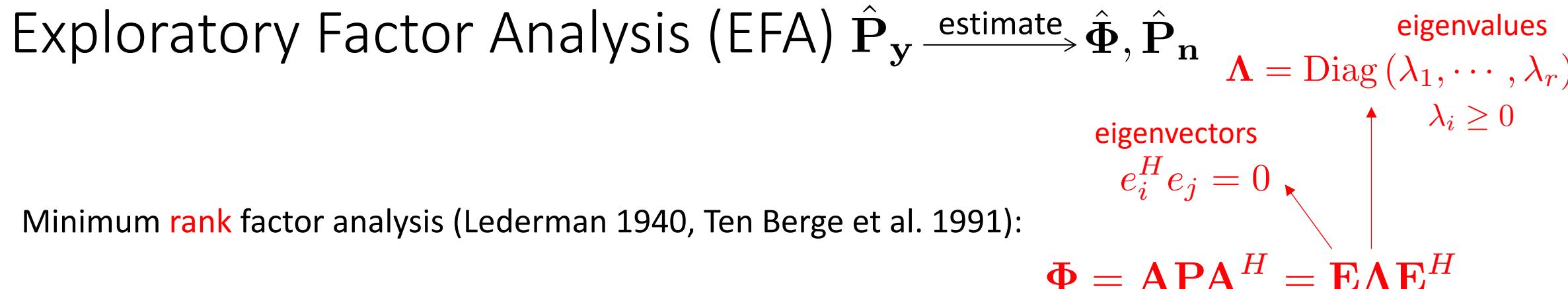
$$\hat{\Phi}, \hat{\mathbf{P}}_n = \arg \min_{\Phi, \mathbf{P}_n} \text{rank}(\Phi)$$

$$\text{s.t. } \hat{\mathbf{P}}_y = \Phi + \mathbf{P}_n$$

$$\Phi \succeq 0$$

$$\mathbf{P}_n = \text{Diag}(p_{n1}, p_{n2}, \dots, p_{n2}) \succeq 0.$$

This is a **non-convex** optimization problem.



$$\hat{\Phi}, \hat{\mathbf{P}}_n = \arg \min_{\Phi, \mathbf{P}_n} \text{rank}(\Phi)$$

$$\text{s.t. } \hat{\mathbf{P}}_y = \Phi + \mathbf{P}_n$$

$$\Phi \succeq 0$$

$$\mathbf{P}_n = \text{Diag}(p_{n1}, p_{n2}, \dots, p_{n2}) \succeq 0.$$

This is a non-convex optimization problem.

Exploratory Factor Analysis (EFA) $\hat{\mathbf{P}}_y \xrightarrow{\text{estimate}} \hat{\Phi}, \hat{\mathbf{P}}_n$

Minimum rank factor analysis (Lederman 1940, Ten Berge et al. 1991):

$$\hat{\Phi}, \hat{\mathbf{P}}_n = \arg \min_{\Phi, \mathbf{P}_n} \text{rank}(\Phi)$$

$$\text{s.t. } \hat{\mathbf{P}}_y = \Phi + \mathbf{P}_n \quad \text{rank}(\Phi) = \|\boldsymbol{\lambda}\|_0 \quad \boldsymbol{\lambda} = \begin{bmatrix} \lambda_1 \\ \vdots \\ \lambda_r \end{bmatrix}$$

$$\Phi \succeq 0$$

$$\mathbf{P}_n = \text{Diag}(p_{n1}, p_{n2}, \dots, p_{n2}) \succeq 0.$$

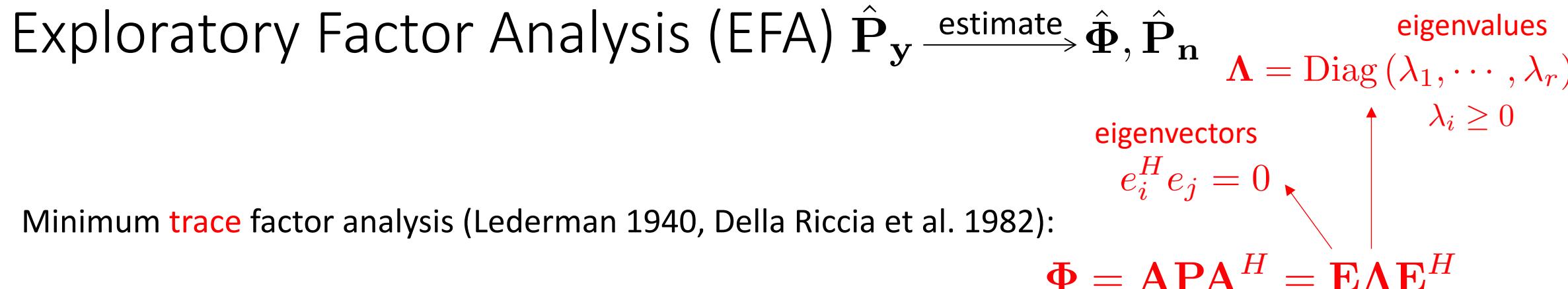
eigenvalues $\Lambda = \text{Diag}(\lambda_1, \dots, \lambda_r)$

eigenvectors $e_i^H e_j = 0$

$\lambda_i \geq 0$

$$\Phi = \mathbf{A} \mathbf{P} \mathbf{A}^H = \mathbf{E} \Lambda \mathbf{E}^H$$

This is a non-convex optimization problem.



Minimum **trace** factor analysis (Lederman 1940, Della Riccia et al. 1982):

$$\hat{\Phi}, \hat{\mathbf{P}}_n = \arg \min_{\Phi, \mathbf{P}_n} \text{trace}(\Phi)$$

$$\text{s.t. } \hat{\mathbf{P}}_y = \Phi + \mathbf{P}_n$$

$$\Phi \succeq 0$$

$$\mathbf{P}_n = \text{Diag}(p_{n1}, p_{n2}, \dots, p_{n2}) \succeq 0.$$

This is a **convex** optimization problem. The global optimum will be obtained for sure.

Exploratory Factor Analysis (EFA) $\hat{\mathbf{P}}_y \xrightarrow{\text{estimate}} \hat{\Phi}, \hat{\mathbf{P}}_n$

$$\Lambda = \text{Diag}(\lambda_1, \dots, \lambda_r)$$

eigenvalues
 $\lambda_i \geq 0$

$$e_i^H e_j = 0$$

$$\Phi = \mathbf{A}\mathbf{P}\mathbf{A}^H = \mathbf{E}\Lambda\mathbf{E}^H$$

Minimum **trace** factor analysis (Lederman 1940, Della Riccia et al. 1982):

$$\hat{\Phi}, \hat{\mathbf{P}}_n = \arg \min_{\Phi, \mathbf{P}_n} \text{trace}(\Phi)$$

s.t. $\hat{\mathbf{P}}_y = \Phi + \mathbf{P}_n$
 $\Phi \succeq 0$

$$\mathbf{P}_n = \text{Diag}(p_{n1}, p_{n2}, \dots, p_{n2}) \succeq 0.$$

$$\text{trace}(\Phi) = \sum_{i=1}^M \lambda_i = \|\boldsymbol{\lambda}\|_1 \quad \boldsymbol{\lambda} = \begin{bmatrix} \lambda_1 \\ \vdots \\ \lambda_r \end{bmatrix}$$

This is a **convex** optimization problem. The global optimum will be obtained for sure.

Exploratory Factor Analysis (EFA) $\hat{\mathbf{P}}_y \xrightarrow{\text{estimate}} \hat{\Phi}, \hat{\mathbf{P}}_n$

Minimum **trace** factor analysis (Lederman 1940, Della Riccia et al. 1982):

$$\hat{\Phi}, \hat{\mathbf{P}}_n = \arg \min_{\Phi, \mathbf{P}_n} \text{trace}(\Phi)$$

$$\text{s.t. } \hat{\mathbf{P}}_y = \Phi + \mathbf{P}_n$$

$$\Phi \succeq 0$$

$$\mathbf{P}_n = \text{Diag}(p_{n1}, p_{n2}, \dots, p_{n2}) \succeq 0.$$

$\hat{r} = \text{rank}(\hat{\Phi})$: is not a good estimate of number of sources

Scree Test (Cattell 1966) $\hat{\Phi} \xrightarrow{\text{estimate}} \hat{r}$

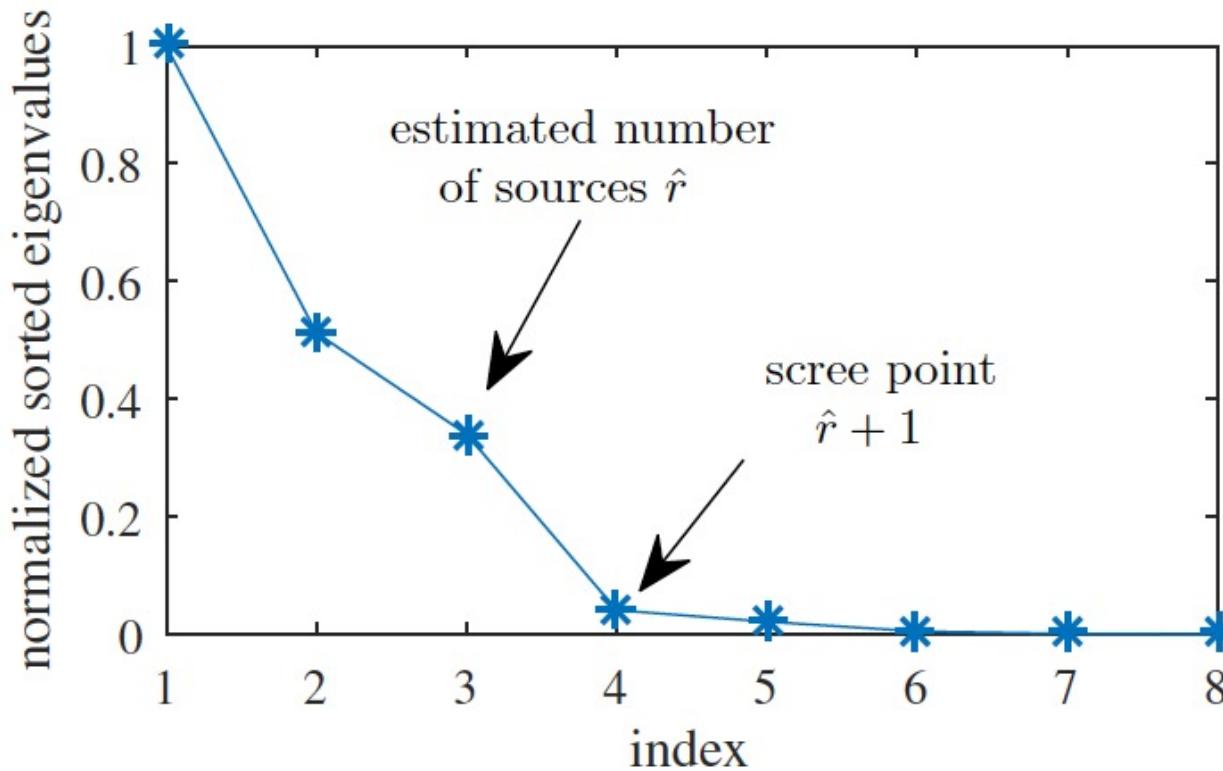
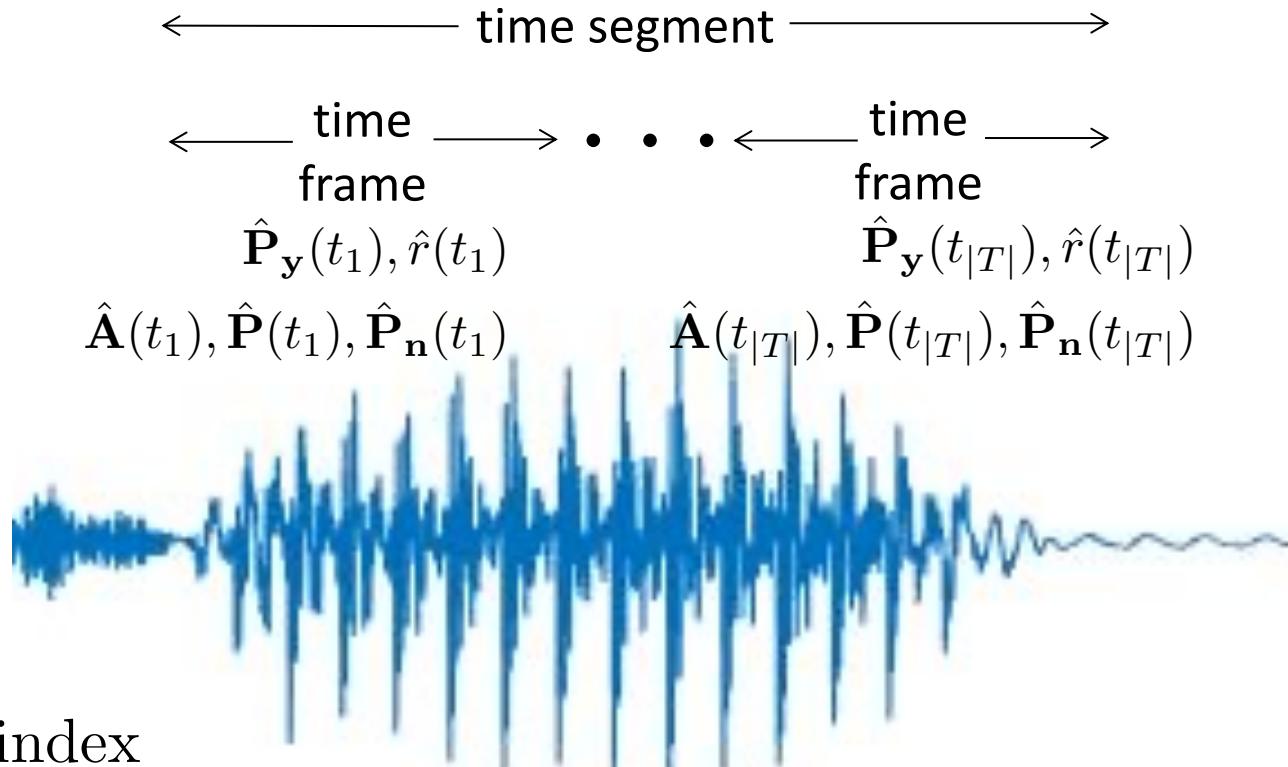


Figure: Normalized eigenvalues of $\hat{\Phi}$ in descending order. Source: Koutrouvelis et al. 2019

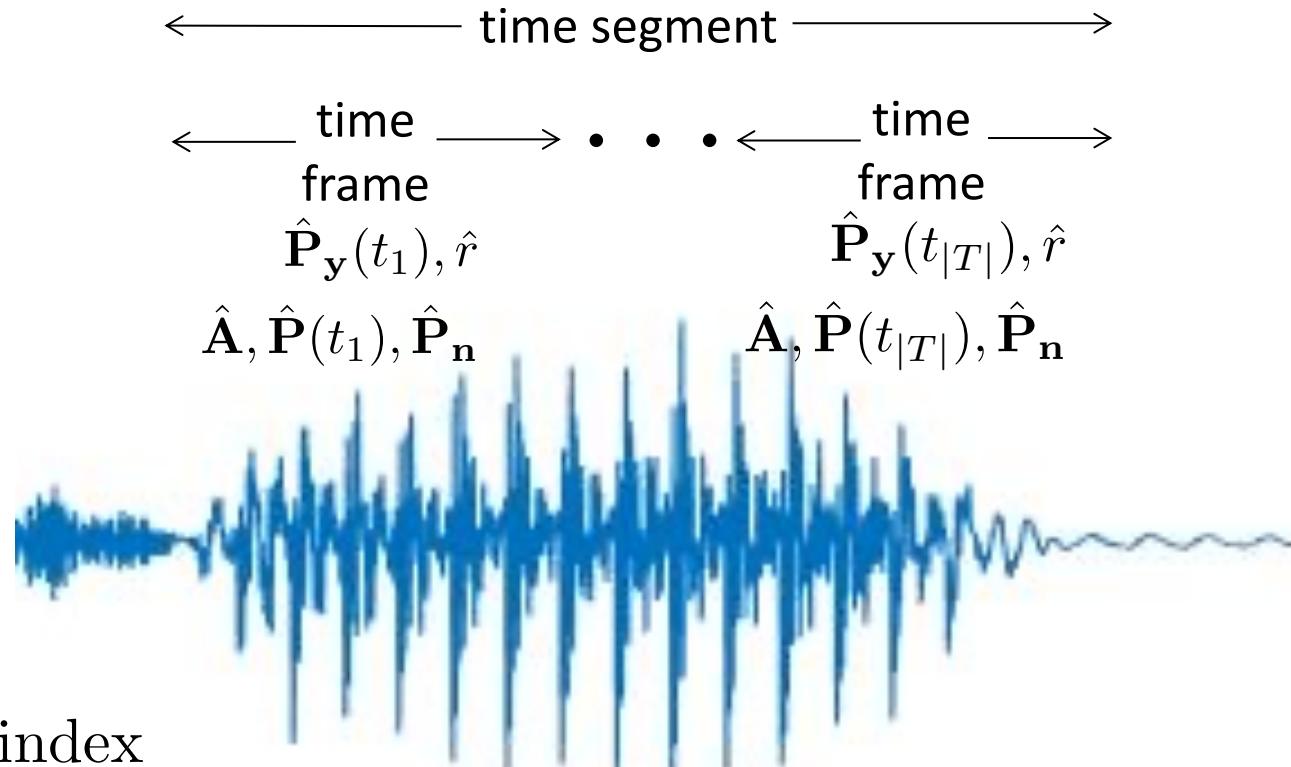
Confirmatory Factor Analysis (CFA) $\hat{\mathbf{P}}_{\mathbf{y}}, \hat{r} \xrightarrow{\text{estimate}} \hat{\mathbf{A}}, \hat{\mathbf{P}}, \hat{\mathbf{P}}_{\mathbf{n}}$



t : time frame index

\mathcal{T} : set of all time frames within the time segment, $t \in \mathcal{T}$

Confirmatory Factor Analysis (CFA) $\hat{\mathbf{P}}_{\mathbf{y}}, \hat{r} \xrightarrow{\text{estimate}} \hat{\mathbf{A}}, \hat{\mathbf{P}}, \hat{\mathbf{P}}_{\mathbf{n}}$



t : time frame index

\mathcal{T} : set of all time frames within the time segment, $t \in \mathcal{T}$

Confirmatory Factor Analysis (CFA) $\hat{\mathbf{P}}_{\mathbf{y}}, \hat{r} \xrightarrow{\text{estimate}} \hat{\mathbf{A}}, \hat{\mathbf{P}}, \hat{\mathbf{P}}_{\mathbf{n}}$
 (Jöreskog 1969, Parra et al. 2000, Koutrouvelis et al. 2019)

$$\begin{aligned} \hat{\mathbf{A}}, \{\hat{\mathbf{P}}(t) : t \in \mathcal{T}\}, \hat{\mathbf{P}}_{\mathbf{n}} = & \underset{\substack{\mathbf{A}, \mathbf{P}_{\mathbf{n}}, \\ \{\mathbf{P}(t) : t \in \mathcal{T}\}}}{\arg \min} \sum_{\forall \tau \in \mathcal{T}} F(\hat{\mathbf{P}}_{\mathbf{y}}(\tau), \mathbf{P}_{\mathbf{y}}(\tau)) \\ \text{s.t. } & \mathbf{P}_{\mathbf{y}}(t) = \mathbf{A} \mathbf{P}(t) \mathbf{A}^H + \mathbf{P}_{\mathbf{n}}, \quad \forall t \in \mathcal{T} \\ & \mathbf{P}(t) = \text{Diag}(p_1(t), p_2(t), \dots, p_M(t)) \succeq 0 \\ & \mathbf{P}_{\mathbf{n}} = \text{Diag}(p_{n1}, p_{n2}, \dots, p_{nM}) \succeq 0 \end{aligned}$$

$$F(\hat{\mathbf{P}}_{\mathbf{y}}, \mathbf{P}_{\mathbf{y}}) = \begin{cases} (\text{ML}): \log |\mathbf{P}_{\mathbf{y}}| + \text{tr} \left(\hat{\mathbf{P}}_{\mathbf{y}} \mathbf{P}_{\mathbf{y}}^{-1} \right), \\ (\text{LS}): \frac{1}{2} \|\mathbf{P}_{\mathbf{y}} - \hat{\mathbf{P}}_{\mathbf{y}}\|_F^2, \\ (\text{GLS}): \frac{1}{2} \|\hat{\mathbf{P}}_{\mathbf{y}}^{-\frac{1}{2}} (\mathbf{P}_{\mathbf{y}} - \hat{\mathbf{P}}_{\mathbf{y}}) \hat{\mathbf{P}}_{\mathbf{y}}^{-\frac{1}{2}}\|_F^2. \end{cases} \quad (\text{Mulaik 2009})$$

Confirmatory Factor Analysis (CFA) $\hat{\mathbf{P}}_{\mathbf{y}}, \hat{r} \xrightarrow{\text{estimate}} \hat{\mathbf{A}}, \hat{\mathbf{P}}, \hat{\mathbf{P}}_{\mathbf{n}}$
 (Jöreskog 1969, Parra et al. 2000, Koutrouvelis et al. 2019)

$$\begin{aligned} \hat{\mathbf{A}}, \{\hat{\mathbf{P}}(t) : t \in \mathcal{T}\}, \hat{\mathbf{P}}_{\mathbf{n}} = & \underset{\substack{\mathbf{A}, \mathbf{P}_{\mathbf{n}}, \\ \{\mathbf{P}(t) : t \in \mathcal{T}\}}}{\arg \min} \sum_{\forall \tau \in \mathcal{T}} F(\hat{\mathbf{P}}_{\mathbf{y}}(\tau), \mathbf{P}_{\mathbf{y}}(\tau)) \\ \text{s.t. } & \mathbf{P}_{\mathbf{y}}(t) = \mathbf{A} \mathbf{P}(t) \mathbf{A}^H + \mathbf{P}_{\mathbf{n}}, \quad \forall t \in \mathcal{T} \\ & \mathbf{P}(t) = \text{Diag}(p_1(t), p_2(t), \dots, p_M(t)) \succeq 0 \\ & \mathbf{P}_{\mathbf{n}} = \text{Diag}(p_{n1}, p_{n2}, \dots, p_{nM}) \succeq 0 \end{aligned}$$

$$\hat{\mathbf{A}} \approx \mathbf{AT}^{-1}, \hat{\mathbf{P}}(t) \approx \mathbf{TP}(t)\mathbf{T}^H$$

$$\mathbf{T} = \color{red}{\mathbf{\Pi S}}$$

permutation, scaling

Confirmatory Factor Analysis (CFA) $\hat{\mathbf{P}}_{\mathbf{y}}, \hat{r} \xrightarrow{\text{estimate}} \hat{\mathbf{A}}, \hat{\mathbf{P}}, \hat{\mathbf{P}}_{\mathbf{n}}$
 (Jöreskog 1969, Parra et al. 2000, Koutrouvelis et al. 2019)

$$\hat{\mathbf{A}}, \{\hat{\mathbf{P}}(t) : t \in \mathcal{T}\}, \hat{\mathbf{P}}_{\mathbf{n}} = \arg \min_{\substack{\mathbf{A}, \mathbf{P}_{\mathbf{n}}, \\ \{\mathbf{P}(t) : t \in \mathcal{T}\}}} \sum_{\forall \tau \in \mathcal{T}} F(\hat{\mathbf{P}}_{\mathbf{y}}(\tau), \mathbf{P}_{\mathbf{y}}(\tau))$$

s.t. $\mathbf{P}_{\mathbf{y}}(t) = \mathbf{A}\mathbf{P}(t)\mathbf{A}^H + \mathbf{P}_{\mathbf{n}}, \forall t \in \mathcal{T}$

$$\mathbf{P}(t) = \text{Diag}(p_1(t), p_2(t), \dots, p_M(t)) \succeq 0$$

$$\mathbf{P}_{\mathbf{n}} = \text{Diag}(p_{n1}, p_{n2}, \dots, p_{nM}) \succeq 0$$

$$a_{\rho i} = 1, \text{ for } i = 1, \dots, r.$$

$$\hat{\mathbf{A}} \approx \mathbf{A}\mathbf{T}^{-1}, \hat{\mathbf{P}}(t) \approx \mathbf{T}\mathbf{P}(t)\mathbf{T}^H$$

$$\mathbf{T} = \color{red}{\boldsymbol{\Pi}}$$

permutation

We select a
reference
microphone
 ρ

Confirmatory Factor Analysis (CFA) $\hat{\mathbf{P}}_{\mathbf{y}}, \hat{r} \xrightarrow{\text{estimate}} \hat{\mathbf{A}}, \hat{\mathbf{P}}, \hat{\mathbf{P}}_{\mathbf{n}}$
 (Jöreskog 1969, Parra et al. 2000, Koutrouvelis et al. 2019)

$$\hat{\mathbf{A}}, \{\hat{\mathbf{P}}(t) : t \in \mathcal{T}\}, \hat{\mathbf{P}}_{\mathbf{n}} = \arg \min_{\substack{\mathbf{A}, \mathbf{P}_{\mathbf{n}}, \\ \{\mathbf{P}(t) : t \in \mathcal{T}\}}} \sum_{\forall \tau \in \mathcal{T}} F(\hat{\mathbf{P}}_{\mathbf{y}}(\tau), \mathbf{P}_{\mathbf{y}}(\tau))$$

$$\text{s.t. } \mathbf{P}_{\mathbf{y}}(t) = \mathbf{A}\mathbf{P}(t)\mathbf{A}^H + \mathbf{P}_{\mathbf{n}}, \quad \forall t \in \mathcal{T}$$

$$\mathbf{P}(t) = \text{Diag}(p_1(t), p_2(t), \dots, p_M(t)) \succeq 0$$

$$\mathbf{P}_{\mathbf{n}} = \text{Diag}(p_{n1}, p_{n2}, \dots, p_{nM}) \succeq 0$$

$$a_{\rho i} = 1, \text{ for } i = 1, \dots, r.$$

$$\hat{\mathbf{A}} \approx \mathbf{A}\mathbf{T}^{-1}, \hat{\mathbf{P}}(t) \approx \mathbf{T}\mathbf{P}(t)\mathbf{T}^H$$

$$\mathbf{T} = \color{red}{\boldsymbol{\Pi}}$$

permutation

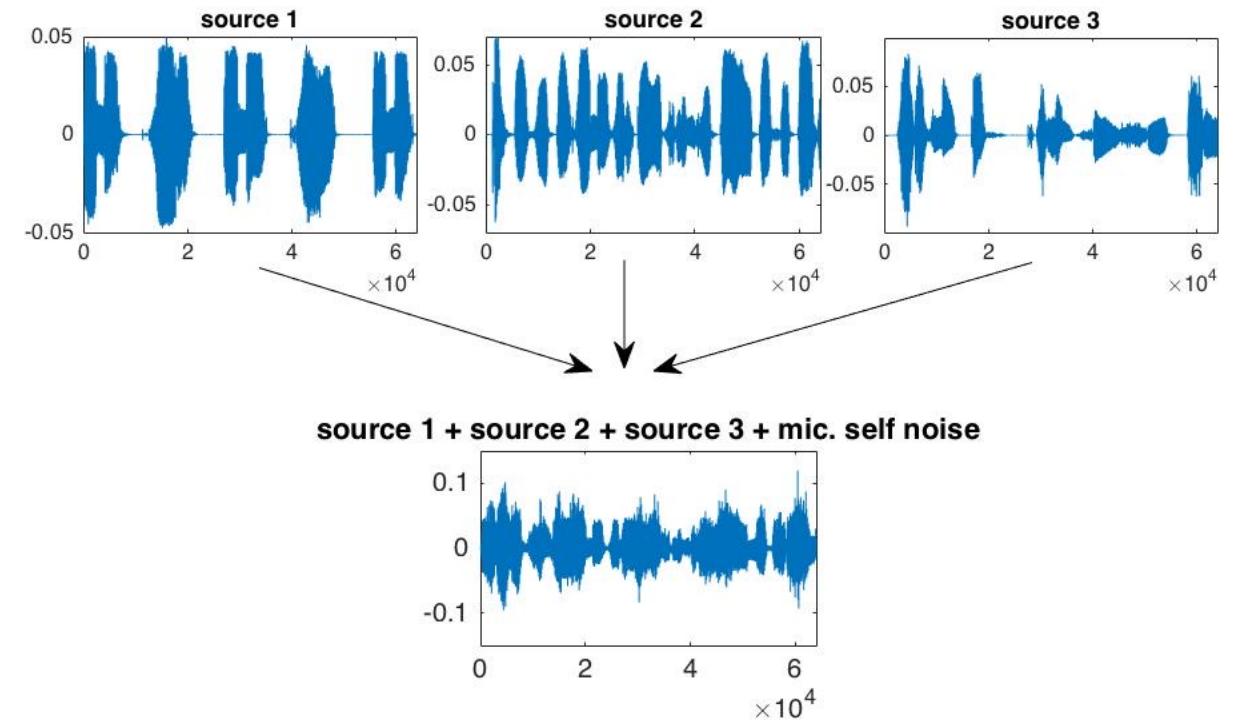
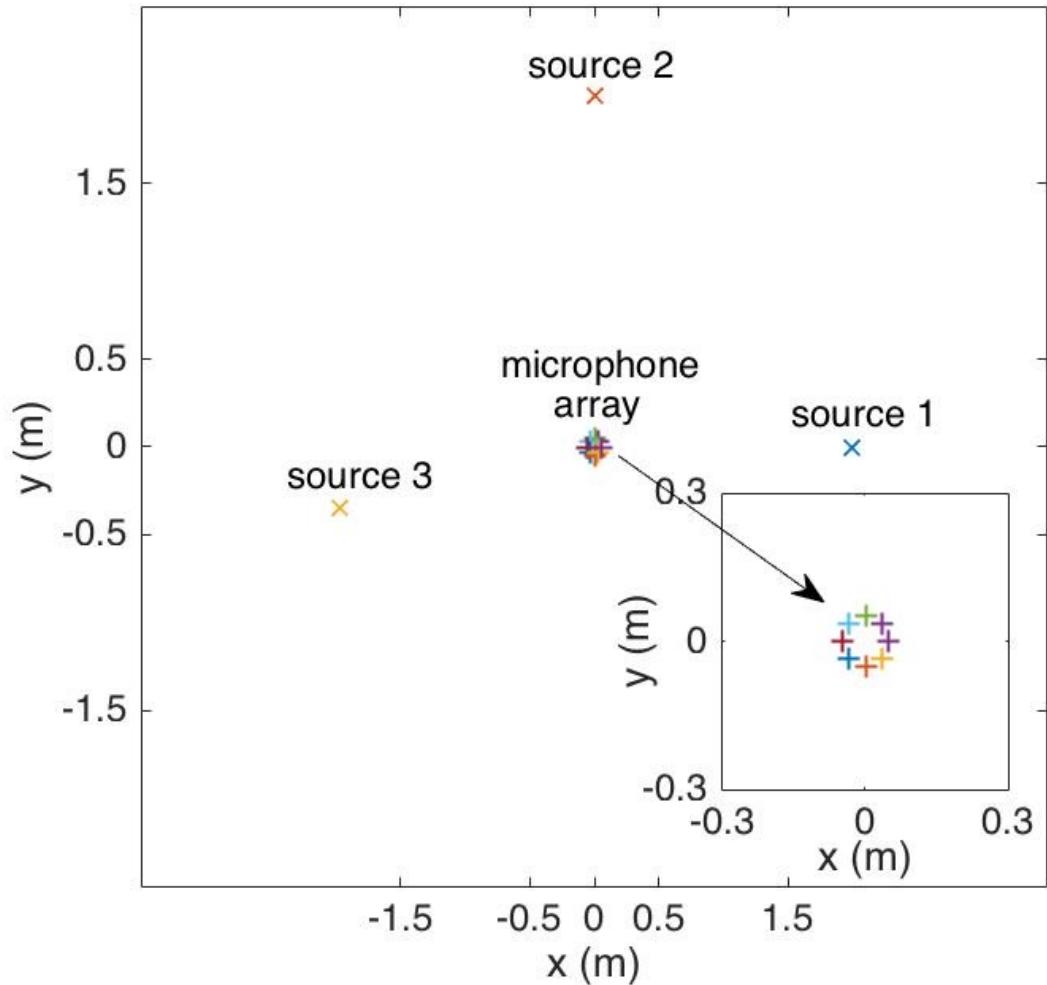
Relative acoustic
transfer functions

Permutation ambiguities

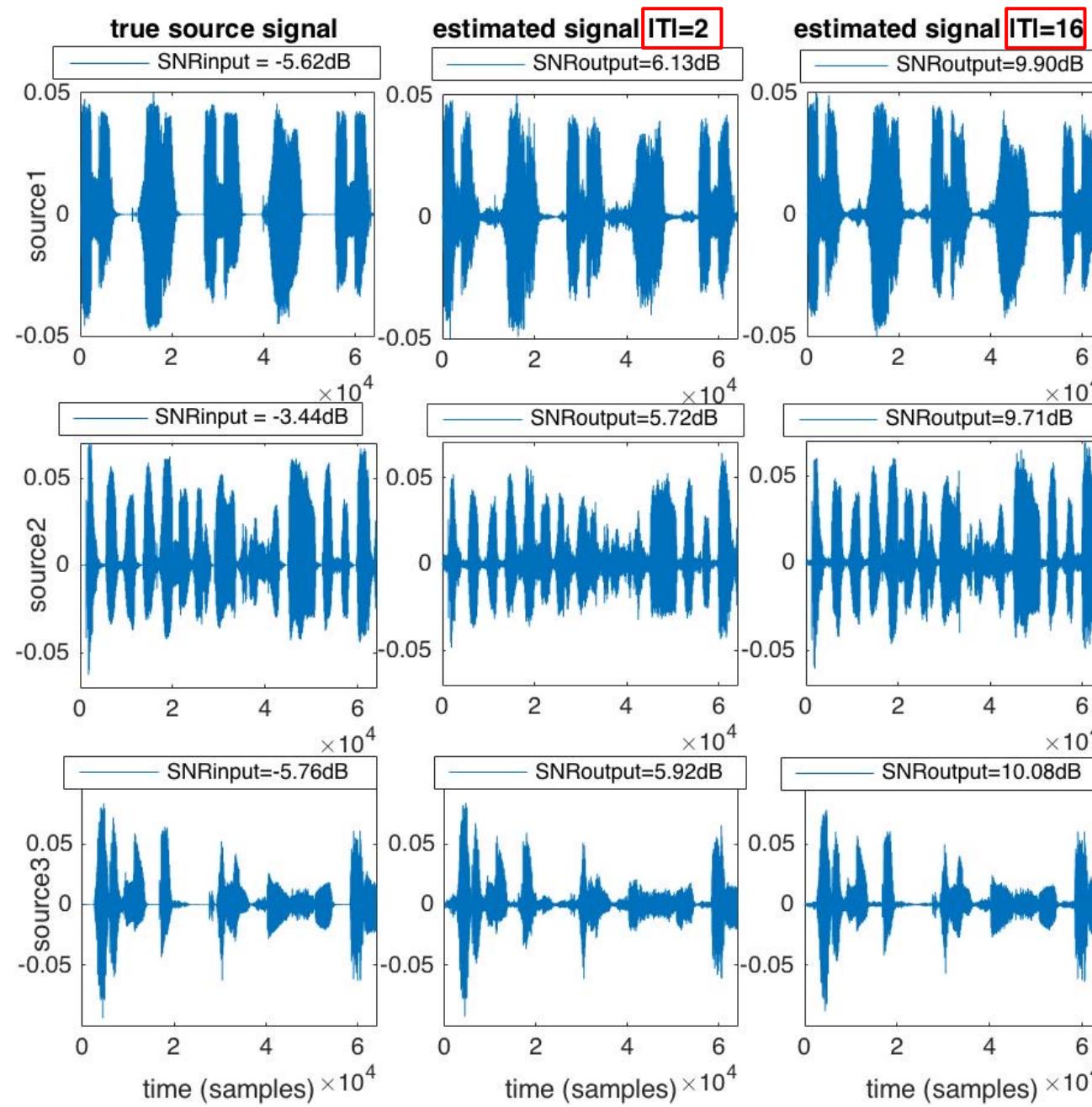
Every time-frequency bin may have a different permutation ambiguity (i.e., a different $\Pi(t, k)$). There are several methods on how to solve the permutation ambiguities over time-frequency bins, e.g.:

- R.Mukai, H. Sawada, S. Araki, and S. Makino, “Frequency-domain blind source separation of many speech signals using near-field and far-field models,” EURASIP J. Appl. Signal Process., vol. 2006, no. 1, pp. 1–13, 2006.
- D. Nion, K. Mokios, N. D. Sidiropoulos, and A. Potamianos, “Batch and adaptive parafac-based blind separation of convolutive speech mixtures,” IEEE Trans. Audio, Speech, Lang. Process., vol. 18, no. 6, pp. 1193–1207, Aug. 2010.

Experiments



Experiments



Typically, an increase in $|\mathcal{T}|$ implies a better performance

Estimation using
EFA + CFA +
MMSE

References

- J.L.Fanagan,A.C.Surendran, and E.E.Jan, "Spatially selective sound capture for speech and audio processing," *ELSEVIER Speech Commun.*, vol. 13, no. 1-2, pp. 207–222, Oct. 1993.
- J. Capon, "High-resolution frequency-wavenumber spectrum analysis," *Proc. of the IEEE*, vol. 57, no. 8, pp. 1408–1418, Aug. 1969.
- O. L. Frost III, "An algorithm for linearly constrained adaptive array processing," *Proc. of the IEEE*, vol. 60, no. 8, pp. 926–935, Aug. 1972.
- M. Brandstein and D. Ward (Eds.), *Microphone Arrays: Signal Processing Techniques and Applications*. New York, NY, USA: Springer, 2001.
- L. Parra and C. Spence, "Convulsive blind separation of non-stationary sources," *IEEE Trans. Audio, Speech, Language Process.*, vol. 8, no. 3, pp. 320–327, 2000.
- D. N. Lawley and A. E. Maxwell, *Factor Analysis as a Statistical Method*. London Butterworths, 1963.
- K. G. Jöreskog, "A general approach to confirmatory maximum likelihood factor analysis," *Psychometrika*, vol. 34, no. 2, pp. 183–202, 1969.
- K. G. Jöreskog, "Simultaneous factor analysis in several populations," *Psychometrika*, vol. 36, no. 4, pp. 409–426, 1971.
- S. A. Mulaik, *Foundations of factor analysis*. CRC press, 2009.
- K. G. Jöreskog, "Factor analysis by generalized least squares," *Psychometrika*, vol. 37, no. 3, pp. 243–260, 1972.
- K. G. Jöreskog and D. N. Lawley, "New methods in maximum likelihood factor analysis," *British J. Math. Statist. Psycol.*, vol. 21, pp. 85–96, 1968.
- A. M. Sardarabadi and A. J. van der Veen, "Complex factor analysis and extensions," *IEEE Trans. on Signal Process.*, vol. 66, no. 4, pp. 954–967, 2018.
- A. I. Koutrouvelis, R. C. Hendriks, R. Heusdens, and J. Jensen, "Robust joint estimation of multi-microphone signal model parameters," *IEEE/ACM Trans. Audio, Speech, Language Process.*, vol. 27, no. 7, pp. 1136–1150, 2019.
- A.I. Koutrouvelis, R.C. Hendriks, R. Heusdens and J. Jensen, "Estimation of Sensor Array Signal Model Parameters Using Factor Analysis," EUSIPCO 2019.
- J. M. Ten Berge and H. A. Kiers, "A numerical approach to the approximate and the exact minimum rank of a covariance matrix," *Psychometrika*, vol. 56, no. 2, pp. 309–315, 1991.
- G. Della Riccia and A. Shapiro, "Minimum rank and minimum trace of covariance matrices," *Psychometrika*, vol. 47, no. 4, pp. 443–448, 1982.
- W. Ledermann, "On a problem concerning matrices with variable diagonal elements," *Proc. Roy. Soc. Edinburgh*, vol. 60, pp. 1–17, 1940.
- R. B. Cattell, "The scree test for the number of factors," *Multivariate behavioral research*, vol. 1, no. 2, pp. 245–276, 1966.