

RTF-Steered Binaural MVDR Beamforming Incorporating an External Microphone for Dynamic Acoustic Scenarios

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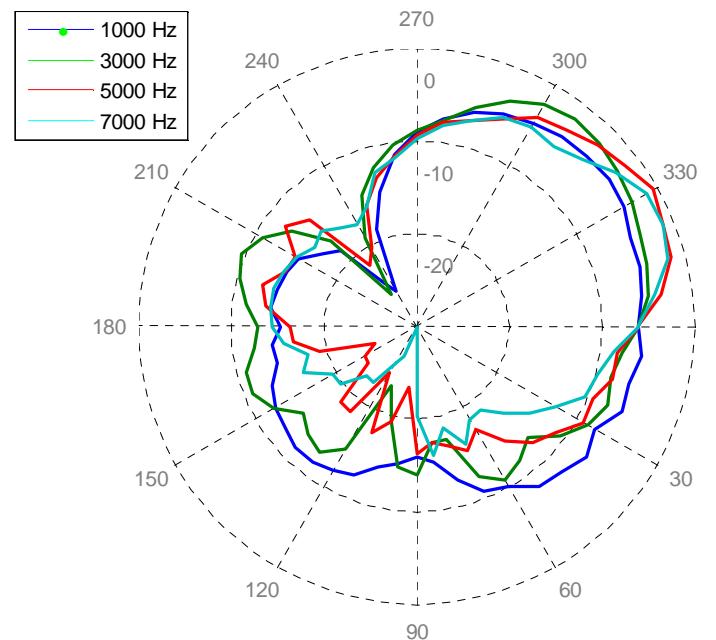
Introduction

- Hearing impaired suffer from a loss of speech understanding in adverse acoustic environments with competing speakers, background noise and reverberation

Multiple microphones available → spatial + spectral processing



Monaural (2-3)



microphones

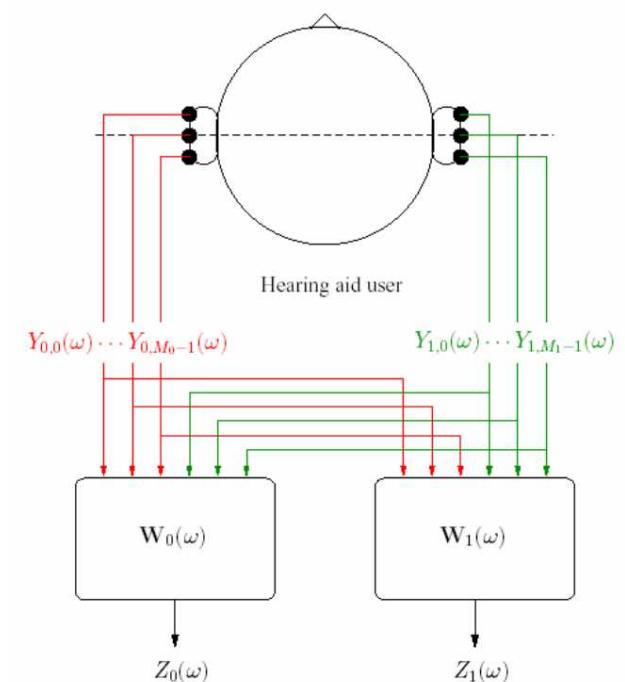
Introduction

□ This presentation:

- **Binaural noise reduction algorithms** based on minimum variance distortionless response (MVDR) beamformer
- Integration with **external microphone(s)** that are spatially separated from the hearing aid microphones

□ Main objectives of algorithms:

- Improve speech intelligibility and avoid signal distortions
- Preserve spatial awareness and directional hearing (binaural cues)



Binaural noise reduction

Minimum-Variance-Distortionless-Response (MVDR) beamformer

Spatial filtering using **all** microphones
(head-mounted and external)

Goal: minimize noise power while preserving speech component in left and right reference microphone signals

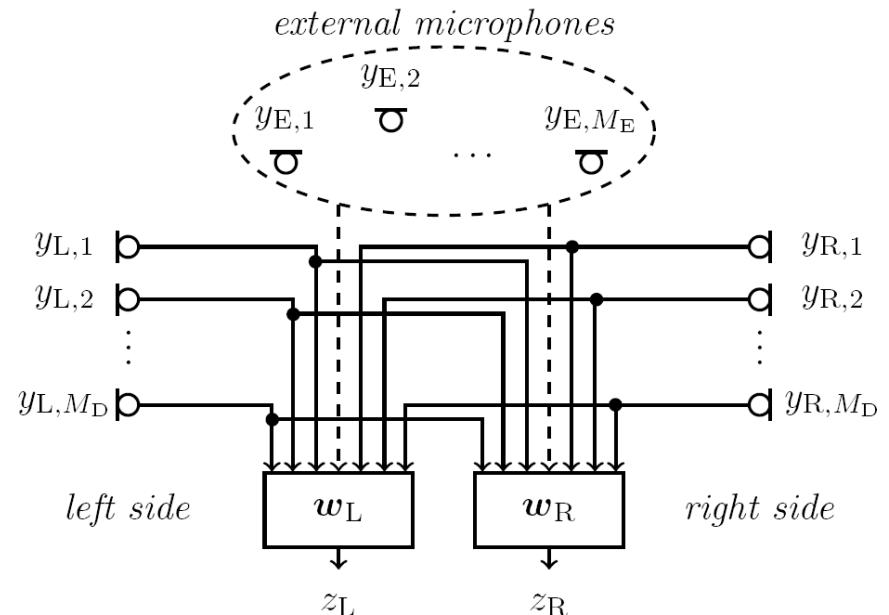
$$\min_{\mathbf{w}_L} \mathcal{E}\{|\mathbf{w}_L^H \mathbf{n}|^2\} \quad \text{subject to} \quad \mathbf{w}_L^H \mathbf{x} = X_{L,1}$$

$$\min_{\mathbf{w}_R} \mathcal{E}\{|\mathbf{w}_R^H \mathbf{n}|^2\} \quad \text{subject to} \quad \mathbf{w}_R^H \mathbf{x} = X_{R,1}$$

↑
noise reduction

↑
distortionless constraint

$$\mathbf{w}_L = \frac{\mathbf{R}_n^{-1} \mathbf{h}_L}{\mathbf{h}_L^H \mathbf{R}_n^{-1} \mathbf{h}_L}, \quad \mathbf{w}_R = \frac{\mathbf{R}_n^{-1} \mathbf{h}_R}{\mathbf{h}_R^H \mathbf{R}_n^{-1} \mathbf{h}_R}$$

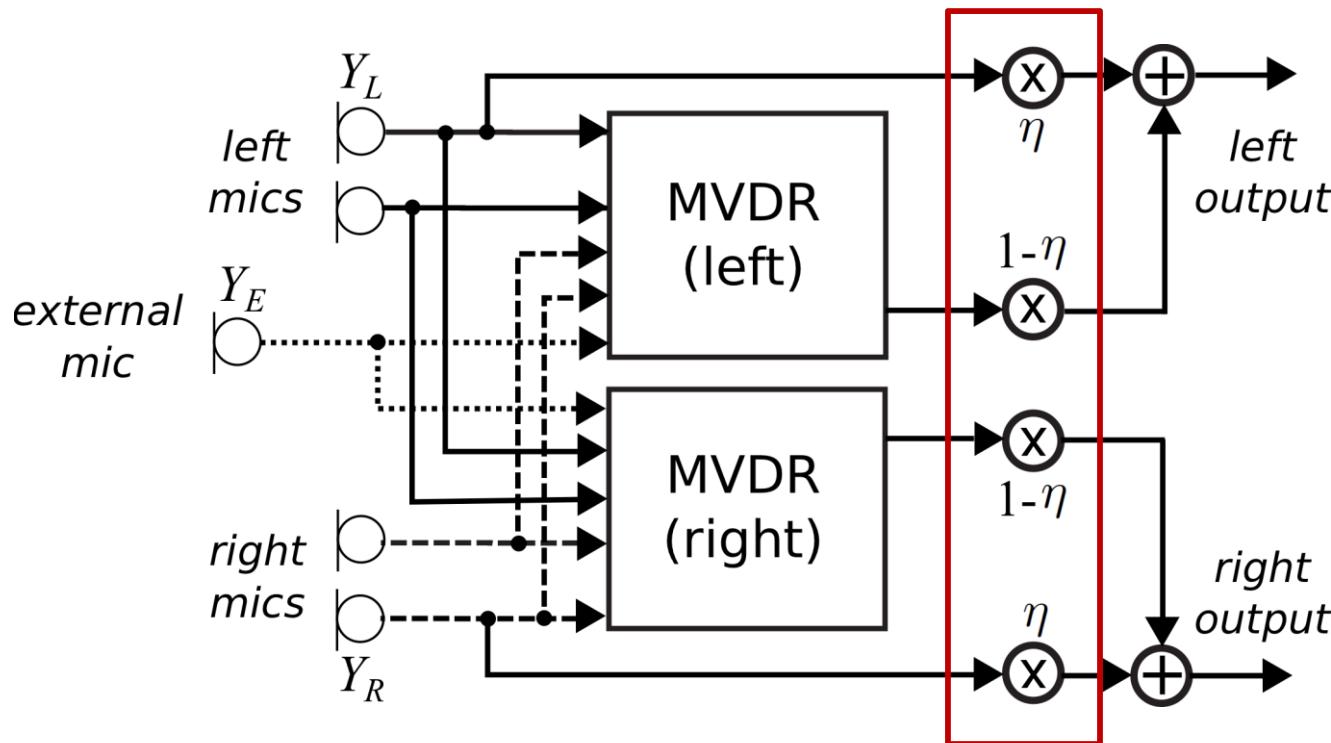


Requires estimate/model of noise covariance matrix (e.g. diffuse) and **estimate/model of relative transfer function (RTF)** of desired speech source

Preserves **binaural cues** of desired source, but distorts binaural cues of noise

Binaural MVDR beamformer with partial noise estimation

- **Goal:** preserve binaural cues of residual noise by **partly mixing** binaural MVDR output signals with reference microphone signals



- $\eta = \mathbf{0}$: binaural MVDR (optimal noise reduction, but no cue preservation)
- $\eta = \mathbf{1}$: reference microphone signals (perfect cue preservation, but no noise reduction)

Note: different procedures available to determine trade-off parameter η (frequency/signal-dependent, psycho-acoustically motivated)

External microphones

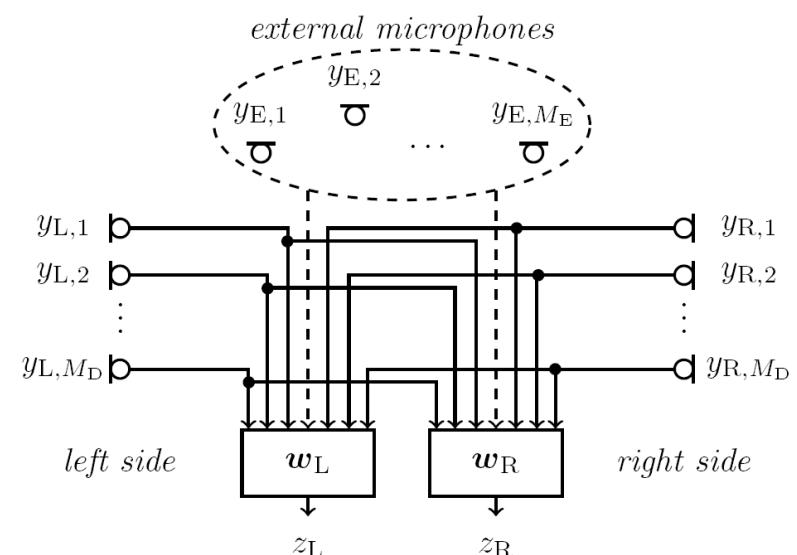
External microphones

- Exploit the availability of one or more external microphones (**acoustic sensor network**) with hearing aids

[Bertrand 2009, Szurley 2016, Yee 2018, Farmani 2018, Kates 2018, Ali 2019, Gößling 2019]

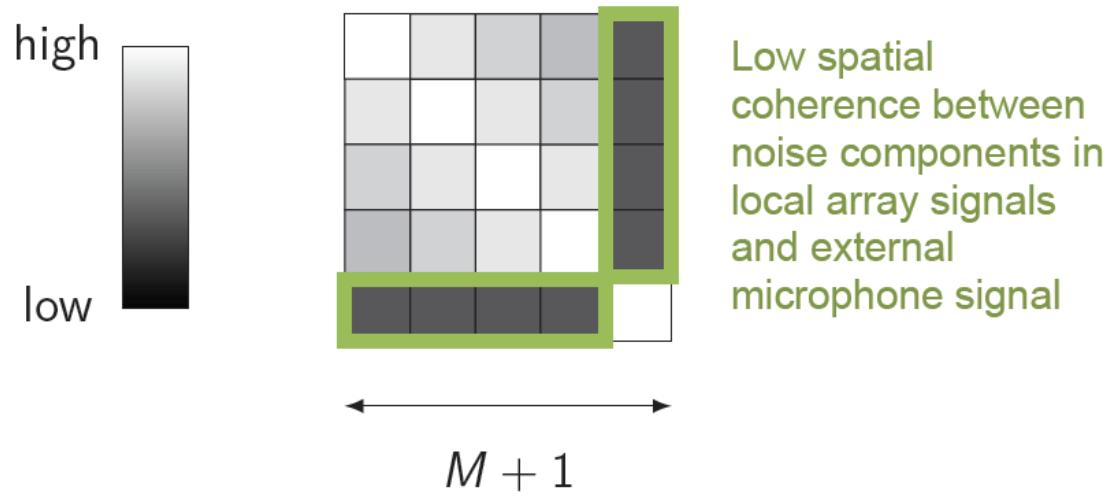
- Integrating external microphone(s) with hearing aid microphones may lead to:
 - Low-complexity method to **estimate relative transfer function (RTF)** vector of target speaker
 - Improved **noise reduction** and **binaural cue preservation** performance

$$\mathbf{w}_L = \frac{\mathbf{R}_n^{-1} \mathbf{h}_L}{\mathbf{h}_L^H \mathbf{R}_n^{-1} \mathbf{h}_L}, \quad \mathbf{w}_R = \frac{\mathbf{R}_n^{-1} \mathbf{h}_R}{\mathbf{h}_R^H \mathbf{R}_n^{-1} \mathbf{h}_R}$$



One external microphone: RTF estimation

- **Estimate RTF vector of target speaker** to steer binaural MVDR beamformer
- **Spatial coherence (SC) method:** assume that noise components in external microphone and HA microphones are uncorrelated, e.g., when external microphone is spatially separated from HA microphones + diffuse noise field
→ correlate HA microphone signals with external microphone signals and normalize by reference element
- **Low computational complexity** with similar (even better in practice) performance than state-of-the-art covariance whitening (CW) approach

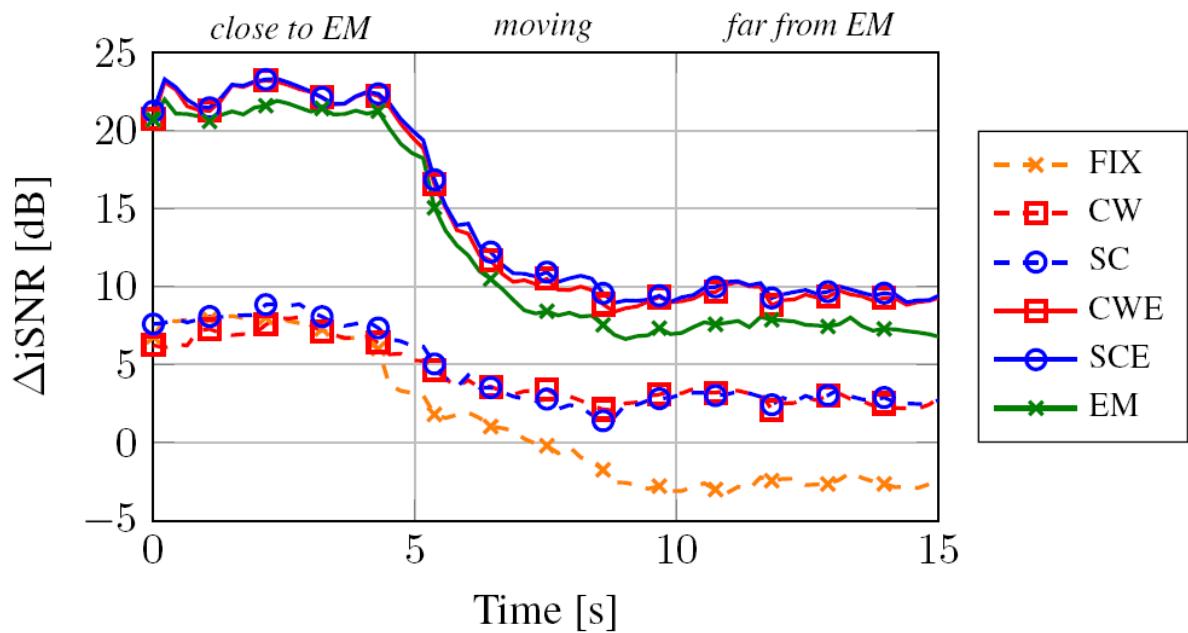
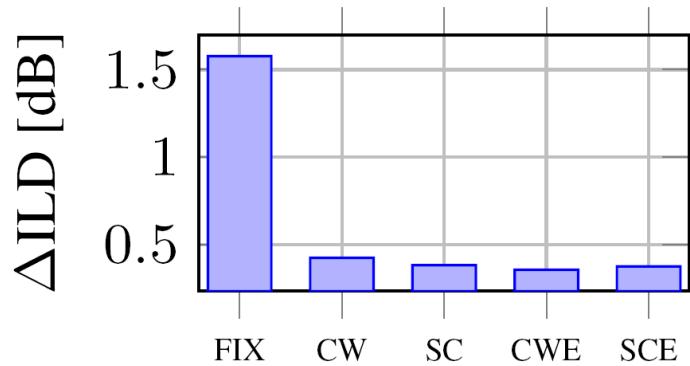
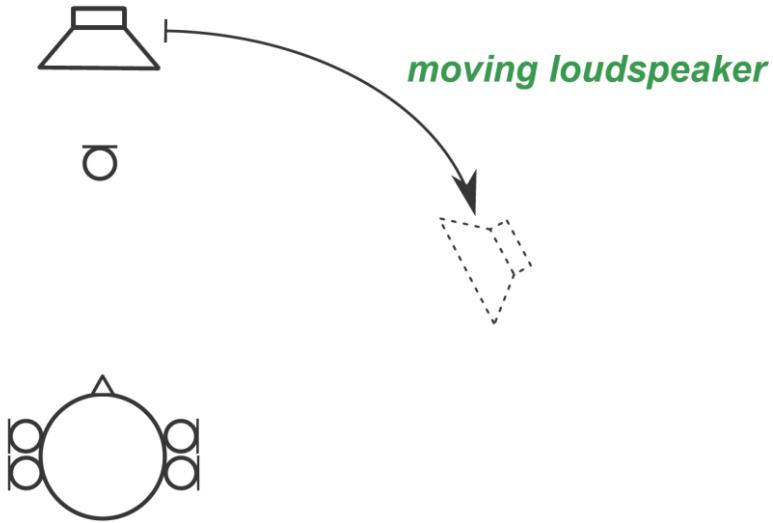


$$\bar{\mathbf{a}}_L^{SCE} = \frac{\bar{\mathbf{R}}_y \mathbf{e}_E}{\mathbf{e}_L^T \bar{\mathbf{R}}_y \mathbf{e}_E}, \quad \bar{\mathbf{a}}_R^{SCE} = \frac{\bar{\mathbf{R}}_y \mathbf{e}_E}{\mathbf{e}_R^T \bar{\mathbf{R}}_y \mathbf{e}_E}$$

$$\bar{\mathbf{w}}_L^{SCE} = \left[\begin{array}{c} \alpha \cdot [\mathbf{I}_{2M}, \mathbf{0}_{2M \times 1}] \bar{\mathbf{w}}_L \\ \alpha(1 + \beta) \cdot \mathbf{e}_E^T \bar{\mathbf{w}}_L \end{array} \right]$$

real-valued bias

One external microphone: Simulation results

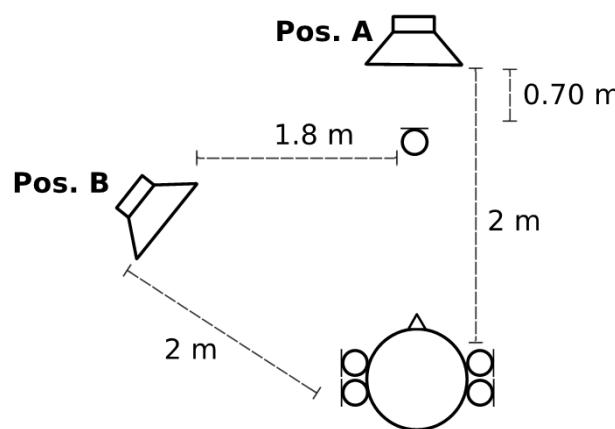


- MVDR with external microphone (SCE) leads to **better SNR** compared to MVDR using only HA microphones (SC, FIX) and external microphone (EM)
- MVDR using estimated RTFs (SCE, SC) **preserves binaural cues of target speaker** compared to fixed MVDR (FIX) and external microphone (EM)

Oldenburg Varechoic Lab ($T_{60} \approx 350\text{ms}$), M=4 + 1 external mic (1.5m/0.5m), moving speaker, pseudo-diffuse babble noise, iSNR=0dB (right HA)
STFT: 32 ms, 50% overlap, sqrt-Hann; SPP in external microphone; smoothing: 100 ms (speech), 1 s (noise)



- Real-world recordings ($T_{60} \approx 300$ ms), **changing speaker position**
- KEMAR with **two BTE hearing aids** (2 mics each) and **one external mic**
- German speaker (10 sec at position A, 10 sec at position B)
- Pseudo-diffuse babble noise



		Filter: only HA microphones	Filter: with external mic
Hearing aid input signals	External mic	MVDR-FIX	eMVDR-SC
		frontal (anechoic)	spatial coherence based RTF estimation
			MVDR
			MVDR-N (partial noise estimation, $\eta=0.2$)

Multiple external microphones

- Each external microphone yields (different) RTF estimate
- **Linear combination/selection** of RTF estimates (per frequency)

$$a_L^{SC-C} = \frac{\mathbf{A}_L^{SC} \mathbf{c}}{\mathbf{e}_L^T \mathbf{A}_L^{SC} \mathbf{c}}$$

1. Input SNR-based selection

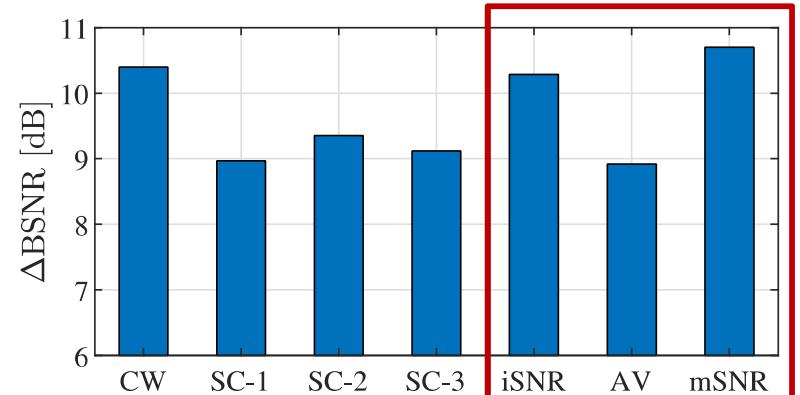
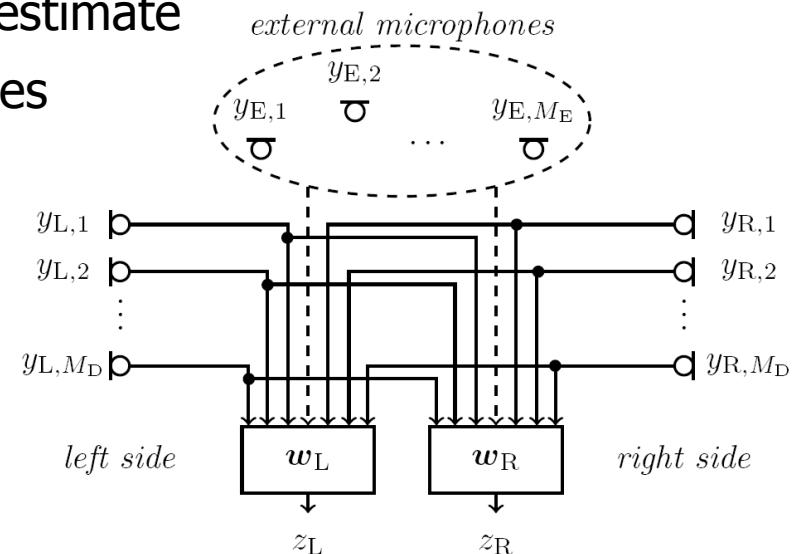
$$\mathbf{c}^{iSNR} = \mathbf{e}_{E,i}, \quad \hat{i} = \arg \max_i \frac{\mathbf{e}_{E,i}^T \mathbf{R}_y \mathbf{e}_{E,i}}{\mathbf{e}_{E,i}^T \mathbf{R}_n \mathbf{e}_{E,i}}$$

2. Simple averaging

$$\mathbf{c}^{AV} = \left[\frac{1}{M_E}, \dots, \frac{1}{M_E} \right]^T$$

3. Output SNR-maximizing combination

$$\mathbf{c}^{mSNR} = \arg \max_{\mathbf{c}} \text{SNR}_{\text{BMVDR,L}}^{\text{out}} = \mathcal{P}\{\mathbf{\Lambda}_2^{-1} \mathbf{\Lambda}_1\}$$



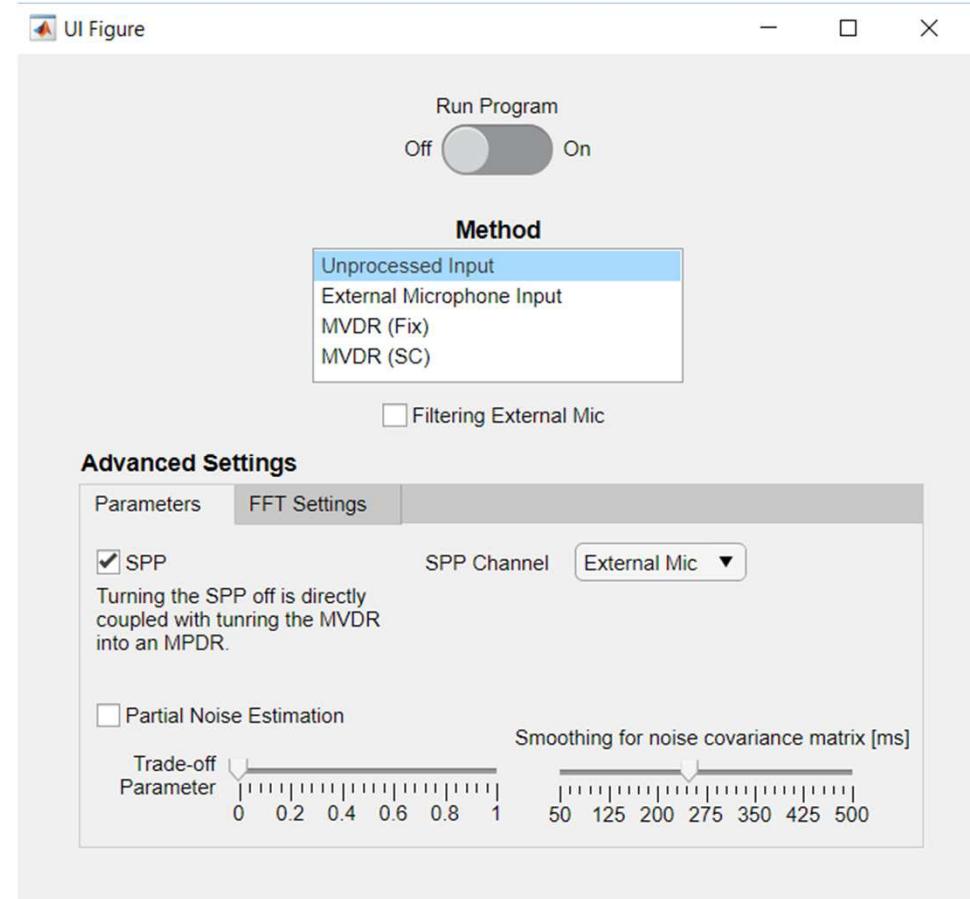
Real-time demonstrator



Hearing aid input:

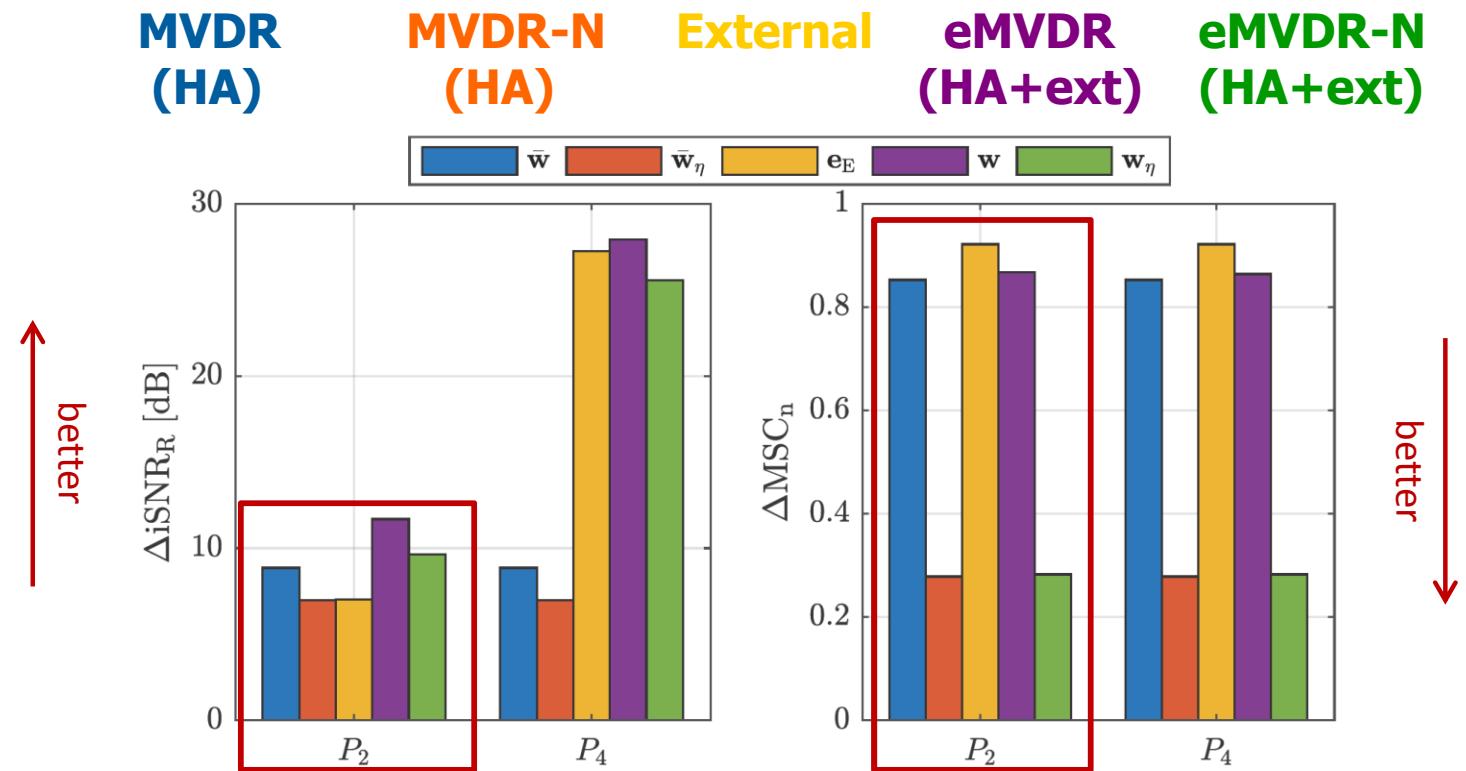
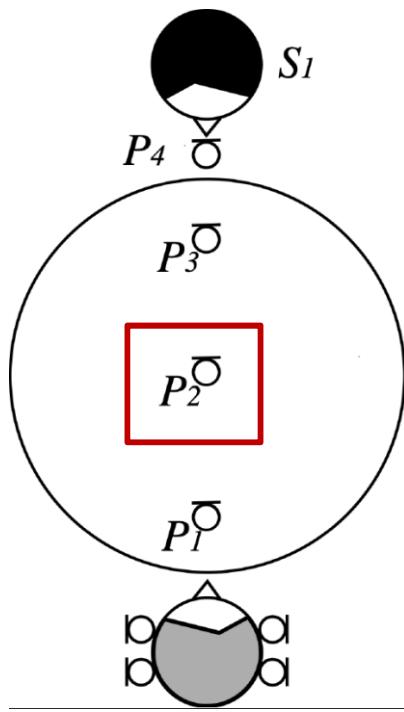


Output signals:



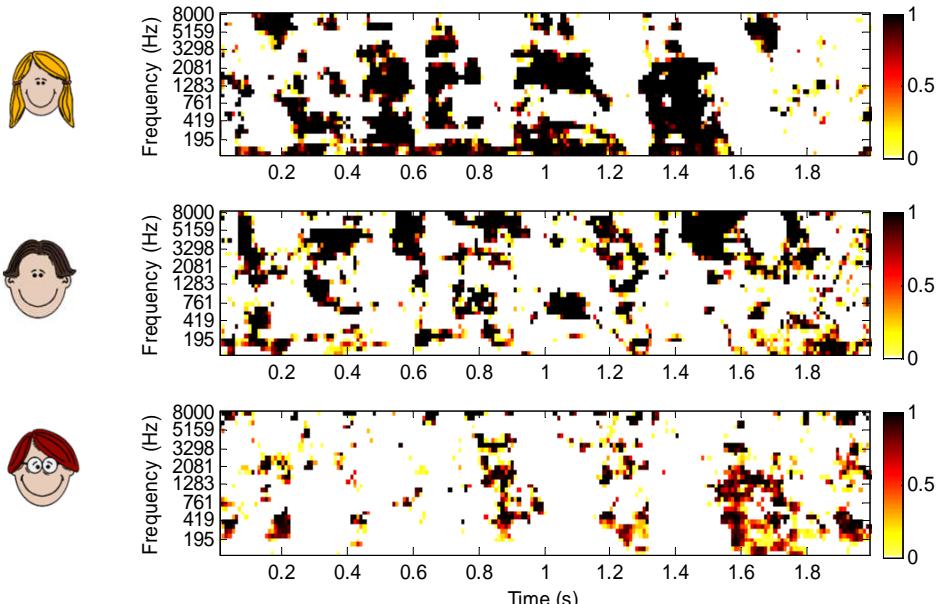
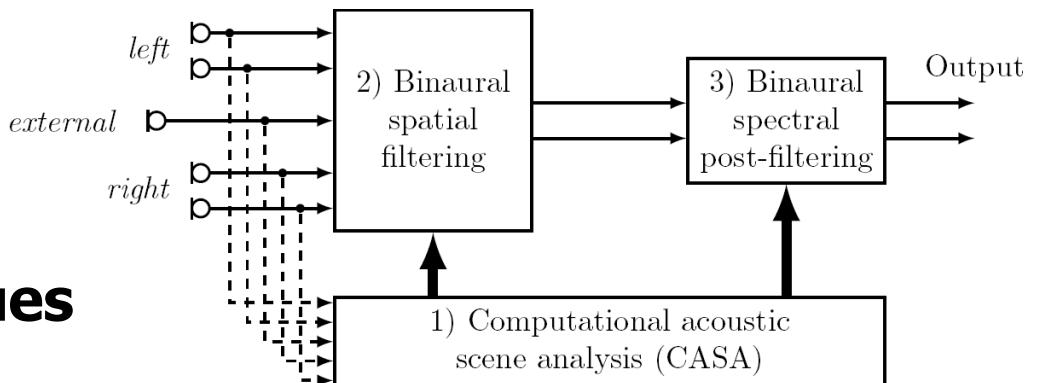
Binaural MVDR-N beamformer

- Including external microphone in **binaural MVDR-N beamformer** leads to:
 - Larger output SNR** for same trade-off parameter η
 - Same output SNR with larger trade-off parameter $\eta \rightarrow$ **better cue preservation**



Starkey database with real-world recordings ($T_{60} \approx 620$ ms), M=4, target speaker S_1 , multi-talker babble noise, 0 dB input iSNR (right hearing aid)
 MVDR: perfectly estimated noise correlation matrix, RTF of target speaker estimated using covariance whitening method

- **Performance analysis** for different acoustic scenarios (interfering speakers)
- **Synchronization/latency issues**
- **Complex and time-varying scenarios:** incorporate computational acoustic scene analysis (CASA) into control path of developed algorithms
- **Subjective evaluation** of binaural speech enhancement algorithms with **HA/CI users** ongoing



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Recent publications

- **S. Doclo, S. Gannot, D. Marquardt, E. Hadad, "Binaural Speech Processing with Application to Hearing Devices", Chapter 18 in *Audio Source Separation and Speech Enhancement* (E. Vincent, T. Virtanen, S. Gannot, eds.), Wiley, 2018.**
- **S. Doclo, W. Kellermann, S. Makino, S. Nordholm, *Multichannel signal enhancement algorithms for assisted listening devices*, IEEE Signal Processing Magazine, vol. 32, no. 2, pp. 18-30, Mar. 2015.**
- D. Marquardt, V. Hohmann, S. Doclo, *Interaural Coherence Preservation in Multi-channel Wiener Filtering Based Noise Reduction for Binaural Hearing Aids*, IEEE/ACM Trans. Audio, Speech and Language Processing, vol. 23, no. 12, pp. 2162-2176, Dec. 2015.
- J. Thiemann, M. Müller, D. Marquardt, S. Doclo, S. van de Par, *Speech Enhancement for Multimicrophone Binaural Hearing Aids Aiming to Preserve the Spatial Auditory Scene*, EURASIP Journal on Advances in Signal Processing, 2016:12, pp. 1-11.
- D. Marquardt, S. Doclo, *Interaural Coherence Preservation in Binaural Hearing Aids using Partial Noise Estimation and Spectral Postfiltering*, IEEE/ACM Trans. Audio, Speech and Language Processing, vol. 26, no. 7, pp. 1257-1270, Jul. 2018.
- N. Gößling, D. Marquardt, S. Doclo, *Performance analysis of the extended binaural MVDR beamformer with partial noise estimation in a homogeneous noise field*, in Proc. Joint Workshop on Hands-free Speech Communication and Microphone Arrays (HSCMA), San Francisco, USA, Mar. 2017, pp. 1-5.
- N. Gößling, S. Doclo, *Relative transfer function estimation exploiting spatially separated microphones in a diffuse noise field*, in Proc. International Workshop on Acoustic Signal Enhancement, Tokyo, Japan, Sep. 2018, pp. 146-150.
- N. Gößling, S. Doclo, *RTF-steered Binaural MVDR Beamforming Incorporating an External Microphone for Dynamic Acoustic Scenarios*, in Proc. IEEE International Conference on Acoustics, Speech and Signal Processing (ICASSP), Brighton, UK, May 2019, pp. 416-420.
- N. Gößling, W. Middelberg, S. Doclo, *RTF-steered Binaural MVDR Beamforming Incorporating Multiple External Microphones*, in Proc. IEEE Workshop on Applications of Signal Processing to Audio and Acoustics (WASPAA), New Paltz, USA, Oct. 2019, pp. 368-372.

<http://www.sigproc.uni-oldenburg.de> → Publications

Questions ?

