

DNN-based Speech Mask Estimation for Eigenvector Beamforming

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Motivation

- Boost beamforming performance using NNs
- Replace Direction-Of-Arrival estimate by a speech mask
- Use speech mask to construct the MVDR, GSC and GEV Beamformers, and a Postfilter
- Speech mask can be learned from eigenvector features







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Why use a speech mask?

- Direction-Of-Arrival estimate:
 - Direct-path steering vector
 - Target leakage may occur



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- Speech mask:
 - Multi-path steering vector (models reverberation)
 - Sufficient to construct Beamformer + Postfilter
 - Existing estimation approaches: use magnitude features [Erdogan et al., 2016] and [Heymann et al., 2016]



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Our idea:

- Use eigenvector features
- Exploit spatial information
- Independent from array geometry and signal energy

Outline



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- 1. System Model
- 2. Super-directive Beamforming: MVDR, GSC, GEV
- 3. Speech Mask Estimation
- 4. Experiments





System Model

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- Single speech source: S(k, l)
- Multi-path ATF: A(k, l)
- Unknown noise: N(k, l)
- System model: $\boldsymbol{Z}(k,l) = S(k,l)\boldsymbol{A}(k,l) + \boldsymbol{N}(k,l)$
- PSDs: $\Phi_{ZZ} = \Phi_{SS} + \Phi_{NN} = AA^H \Phi_S + \Phi_{NN}$



Super-directive Beamforming: MVDR, GSC, GEV

MVDR

Optimal MWF = MVDR + Wiener Postfilter: [Vary and Martin, 2006]

$$\mathbf{W}_{OPT} = \mathbf{\Phi}_{ZZ}^{-1} \mathbf{A} \mathbf{\Phi}_{S} = \underbrace{\frac{\mathbf{\Phi}_{NN}^{-1} \mathbf{A}}{\mathbf{A}^{H} \mathbf{\Phi}_{NN}^{-1} \mathbf{A}}}_{\mathbf{W}_{MVDR}} \cdot \underbrace{\frac{\mathbf{\Phi}_{S}}{\mathbf{\Phi}_{S} + \left[\mathbf{A}^{H} \mathbf{\Phi}_{NN}^{-1} \mathbf{A}\right]^{-1}}}_{G = \frac{\xi}{1+\xi}}$$

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- Substitute ATF A by steering vector F:
 - use dominant Eigenvector: $F
 ightarrow v_{S_1}$ [Pfeifenberger et al., 2016]
 - EVD of the speech PSD: $oldsymbol{\Phi}_{SS} = \sum_{m=1}^{M} oldsymbol{v}_{S_m} oldsymbol{v}_{S_m}^H \lambda_{S_m}$
 - includes multi-path propagation: $F = A \left[\frac{\phi_S}{\lambda_{S_1} A^H v_{S_1}} \right]$

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- Postfilter G:
 - uses multi-channel SNR: $\xi = \text{Tr} \{ \Phi_{NN}^{-1} \Phi_{SS} \}$

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Required: $\Phi_{SS}(k, l)$ and $\Phi_{NN}(k, l)$



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 - $W_{MVDR} pprox W_{GSC} = F BH_{AIC}$ [Hoshuyama et al., 1999]



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Required: $\Phi_{SS}(k, l)$



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GEV

- Maximizes the SNR ξ : [Warsitz and Haeb-Umbach, 2007]
 - $W_{SNR} = \arg \max_{W} \xi$
 - eigenvalue problem (rank = 1): $\Phi_{NN}^{-1}\Phi_{SS}W_{SNR} = \xi W_{SNR}$



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- Modification for reduced distortion: [Pfeifenberger et al., 2016]
 - $W_{GEV} = PF$
 - reduced distortions: $\boldsymbol{W}_{GEV}^{H} \boldsymbol{A} \approx 1$
 - projection matrix: $P = \frac{\Phi_{NN} W_{SNR} W_{SNR}^H}{W_{SNR}^H \Phi_{NN} W_{SNR}}$



¹⁰ GEV

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Speech Mask Estimation



² Speech Mask Estimation

Speech and noise PSD estimates: [Higuchi et al., 2016]

$$\hat{\Phi}_{SS}(k,l) = \frac{\sum_{t=l}^{l+T} Z(k,t) Z^{H}(k,t) p_{\text{SPP}}(k,t)}{\sum_{t=l}^{l+T} p_{\text{SPP}}(k,t)}$$

$$\hat{\Phi}_{NN}(k,l) = \frac{\sum_{t=l}^{l+T} Z(k,t) Z^{H}(k,t) (1-p_{\text{SPP}}(k,t))}{\sum_{t=l}^{l+T} (1-p_{\text{SPP}}(k,t))}$$



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- Speech presence probability p_{SPP}:
 - ground truth: $p_{\text{SPP,opt}} = \frac{\xi}{1+\xi}$
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- Speech presence probability p_{SPP}:
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Observation: $p_{\text{SPP,opt}}$ is related to the dominant Eigenvector v_{Z_1}

EVD of the noisy speech PSD: $m{\Phi}_{ZZ} = \sum_{m=1}^M m{v}_{Z_m} m{v}_{Z_m}^H \lambda_{Z_m}$



Distribution of $v_{Z_1}(k, l)$ colored with $p_{\text{SPP,opt}}(k, l)$, for $k \approx 2650 Hz$:





Distribution of $v_{Z_1}(k, l)$ colored with $p_{\text{SPP,opt}}(k, l)$, for $k \approx 2650 Hz$:



Speech Mask Estimation





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Experiments

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Experiments

- Feature vector variants:
 - Eigenvectors: $\boldsymbol{x}_{\text{ev}}(k,l) = \left[\text{Re}\{\boldsymbol{v}_{Z_1}(k,l)\}^T, \text{Im}\{\boldsymbol{v}_{Z_1}(k,l)\}^T \right]^T$
 - Eigenvector-deltas: $\boldsymbol{x}_{\text{evd}}(k,l) = |\boldsymbol{v}_{Z_1}(k,l)^H \boldsymbol{v}_{Z_1}(k,l+\Delta)|$
 - Energy per channel: $\boldsymbol{x}_{psd}(k,l,m) = 20log_{10}|Z_m(k,l)|$

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 - Energy per channel: $\boldsymbol{x}_{psd}(k,l,m) = 20log_{10}|Z_m(k,l)|$
- NN variants:
 - ev_lstm : LSTM cells + $x_{ev}(k, l)$ features
 - evd_lstm: LSTM cells + x_{evd}(k, l) features
 - evd_mlp: FF layers + x_{evd}(k, l) features
 - psd_lstm: LSTM cells + $x_{psd}(k, l)$ features



Training data: CHiME4 corpus [Barker et al., 2015]

- 2 and 6 channel data
- 14658 utterances
- 4 background noise types: BUS, STR, PED, CAF
- 12 speakers
- provides ground truth $\xi(k, l)$ for training the NN





Speech mask prediction error

architactura	n_{Δ}		predi	iction erro	# of woights	
architecture		n _h	train	valid	test	# OI WEIGINS
ev_lstm	-	-	3.375	4.568	5.166	557176
ev_lstm	-	10	2.176	3.119	3.347	799784
ev_lstm	-	20,10	1.889	2.685	3.003	1457704
evd_lstm	3	10	2.308	2.299	2.823	614744
evd_lstm	5	10	2.251	2.244	2.689	655864
evd_lstm	7	-	2.750	2.761	3.730	546896
evd_lstm	7	10	2.281	2.267	2.690	696984
evd_lstm	7	20,10	2.184	2.183	2.520	1252104
evd_mlp	3	10	2.452	2.424	3.212	76843
evd_mlp	5	10	2.405	2.372	3.069	81983
evd_mlp	7	-	2.752	2.762	3.975	68362
evd_mlp	7	10	2.384	2.376	3.156	87123
evd_mlp	7	20,10	2.349	2.285	2.825	156513
psd_lstm	-	-	3.489	4.391	4.603	544840
psd₋lstm	-	10	2.897	3.722	3.741	676424
psd_lstm	-	20,10	2.711	3.415	3.489	1210984

$$\mathrm{error} = \frac{100}{KL} \sum_{k=1}^{K} \sum_{l=1}^{L} \left| p_{\mathrm{SPP,est,2}}(k,l) - p_{\mathrm{SPP,opt}}(k,l) \right|$$



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PESQ and PEASS/OPS scores [Emiya et al., 2011]

architecture	n_{Δ}	n_h	PL	ESQ [MC	OPS [%]			
architecture			train	valid	test	train	valid	test
ev_lstm, MVDR, 6ch	-	20,10	2.204	1.850	1.788	62	46	39
evd_lstm, MVDR, 6ch	7	20,10	1.948	1.773	1.748	53	45	39
evd_mlp, MVDR, 6ch	3	10	1.866	1.713	1.630	50	45	40
psd_lstm, MVDR, 6ch	-	20,10	1.826	1.663	1.636	54	47	45
ev_lstm, GSC, 6ch	-	20,10	2.045	1.760	1.742	51	41	37
evd_lstm, GSC, 6ch	7	20,10	1.889	1.714	1.706	46	39	37
evd₋mlp, GSC, 6ch	3	10	1.822	1.667	1.602	43	38	37
psd_lstm, GSC, 6ch	-	20,10	1.783	1.620	1.622	49	43	43
ev_lstm, GEV, 6ch	-	20,10	2.443	2.007	1.891	72	58	51
evd_lstm, GEV, 6ch	7	20,10	2.226	1.969	1.874	67	59	52
evd₋mlp, GEV, 6ch	3	10	2.131	1.900	1.758	65	58	51
*psd_lstm, GEV, 6ch	-	20,10	1.977	1.758	1.724	63	54	48
ev_lstm, GEV, 2ch	-	10,5	1.965	1.706	1.725	51	44	45
evd₋mlp, GEV, 2ch	3	5	1.980	1.778	1.774	44	40	40
BeamformIt!, 5ch	-	-	1.350	1.292	1.326	31	36	35
**CGMM-EM, 6ch	-	-	1.635	1.483	1.468	48	42	38

*similar to CHiME4-contributions: [Erdogan et al., 2016] and [Heymann et al., 2016] **CHiME3 winner: [Higuchi et al., 2016]



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• Example 1: мо4_422C0205_CAF







Results



Example 2: F01_22HC010W_BUS





complementary output: $Y_{n,GEV}(k, l)$



² Conclusion

- Take-home message:
 - the MVDR, GSC and GEV Beamformers and the Postfilter solely depend on the speech mask
 - speech mask estimate can be learned from eigenvector features
- Future work:
 - reduce NN size and complexity
 - multiple speakers



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Thank you for your attention!



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