

Speech Enhancement using Polynomial Eigenvalue Decomposition

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Introduction

- Single-channel subspace speech enhancement [Ephraim1995; Hu2002]
 - Use an EVD to decorrelate spectrally
- Multi-channel subspace speech enhancement [Asano2000]
 - Use an EVD to decorrelate spatially

⇒ **Limitation: Only decorrelates instantaneously**

- Other methods typically use STFT to process [Cohen2002; Ephraim1984; Gannot2001; Markovich2009]
 - Use DFT to divide broadband into multiple narrowband signals
 - Require a 4D tensor to model the space, time, spectral correlations

⇒ **Limitations: Lacks phase coherence across bands
: Ignores correlation between bands**

- Polynomial Matrices and Polynomial Eigenvalue Decomposition (PEVD)
 - Simultaneously captures correlation across space, time and frequency using a 3D tensor
 - Impose spatial decorrelation over a range of time shifts
 - No phase discontinuity
- PEVD-based broadband applications:
 - blind source separation [Redif2017]
 - adaptive beamforming [Weiss2015]
 - source identification [Weiss2017]

This Talk: PEVD for Speech Enhancement

Background

The received signal at the q -th sensor with time index n is

$$x_q(n) = \sum_{j=0}^J h_q(n-j)s(j) + v_q(n),$$

where

- $s(n)$ is the source signal,
- $h_q(n)$ is the channel modelled as an order J FIR filter,
- $v_q(n)$ is the noise signal at the q -th sensor.

The data vector collected from Q sensors is

$$\mathbf{x}(n) = [x_1(n), x_2(n), \dots, x_Q(n)]^T.$$

Assuming stationarity, space-time covariance matrix is

$$\mathbf{R}_{\mathbf{xx}}(\tau) = \mathbb{E}[\mathbf{x}(n)\mathbf{x}^H(n - \tau)],$$

where $(i, j)^{\text{th}}$ element is the correlation function $r_{ij}(\tau) = \mathbb{E}[x_i(n)x_j^*(n - \tau)]$ and τ is the time-shift.

Z-transform of $\mathbf{R}_{\mathbf{xx}}(\tau)$ is a para-Hermitian polynomial matrix

$$\mathcal{R}_{\mathbf{xx}}(z) = \sum_{\tau=-W}^W \mathbf{R}_{\mathbf{xx}}(\tau)z^{-\tau},$$

where $\mathbf{R}_{\mathbf{xx}}(\tau) \approx 0$ for $|\tau| > W$, calligraphy \mathcal{R} for tensor and regular \mathbf{R} for matrix.

The PEVD of $\mathcal{R}_{\mathbf{x}\mathbf{x}}(z)$ is defined as [McWhirter2007]

$$\mathcal{R}_{\mathbf{x}\mathbf{x}}(z) \approx \mathbf{U}^P(z)\mathbf{\Lambda}(z)\mathbf{U}(z) \Leftrightarrow \mathbf{\Lambda}(z) \approx \mathbf{U}(z)\mathcal{R}_{\mathbf{x}\mathbf{x}}(z)\mathbf{U}^P(z),$$

where $\mathbf{\Lambda}(z), \mathbf{U}(z)$ are the eigenvalue and eigenvector polynomial matrices and $\mathcal{R}_{\mathbf{x}\mathbf{x}}(z) = \mathcal{R}_{\mathbf{x}\mathbf{x}}^P(z) = \mathcal{R}_{\mathbf{x}\mathbf{x}}^H(z^{-1})$.

$\mathbf{U}(z)$ is a filterbank for $\mathbf{x}(z) \in \mathbb{C}^{Q \times 1 \times T}$ so that the outputs in

$$\mathbf{y}(z) = \mathbf{U}(z)\mathbf{x}(z) \implies \mathcal{R}_{\mathbf{y}\mathbf{y}}(z) \approx \mathbf{\Lambda}(z),$$

are strongly decorrelated.

PEVD algorithms include:

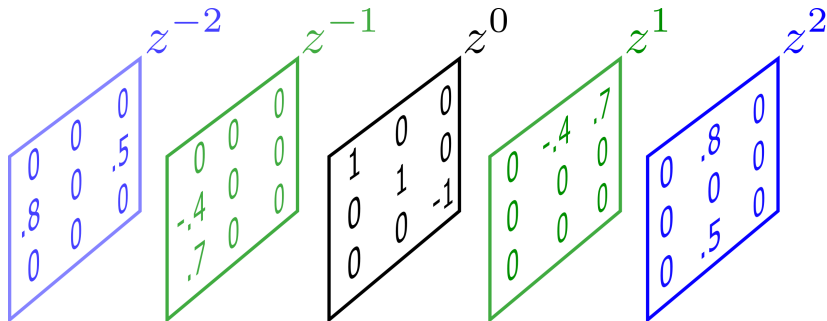
- Second-order Sequential Best Rotation (SBR2) [McWhirter2007]
- Sequential Matrix Diagonalization (SMD) [Redif2015]
- Householder-like PEVD [Redif2011]
- Tridiagonal PEVD [Neo2019]
- Multiple-shift SBR2/SMD [Wang2015; Corr2014]

Typically compute $\mathbf{R}_{\mathbf{x}\mathbf{x}}(0) = \mathbb{E}[\mathbf{x}(n)\mathbf{x}^H(n)]$:

$$\begin{array}{|c|} \hline z^0 \\ \hline \begin{array}{ccc} 1 & 0 & 0 \\ 0 & 1 & 0 \\ 0 & 0 & -1 \end{array} \\ \hline \end{array}$$

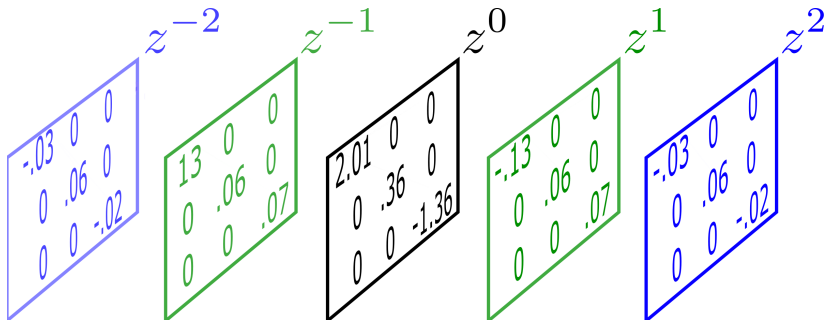
$\mathbf{R}_{\mathbf{x}\mathbf{x}}(0)$: instantaneous (spatial) covariance matrix / coefficient of z^0 .

Before diagonalization, $\mathcal{R}_{xx}(z)$:

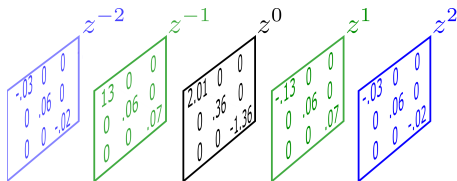


In this example, z^0 plane is diagonal but not at other planes.

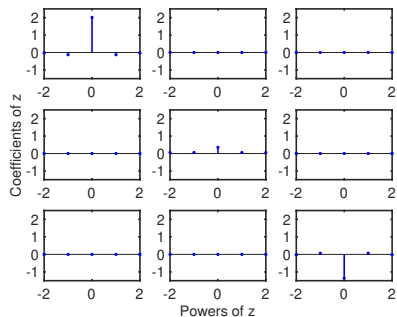
After diagonalization using PEVD, $\Lambda(z)$:



Equivalently, expressed as:

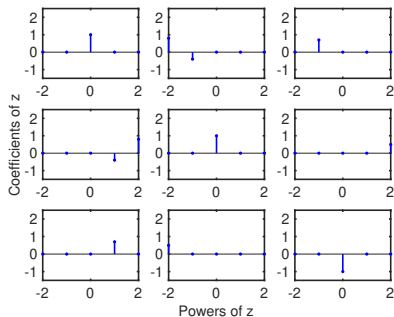


Polynomial with matrix coefficients.

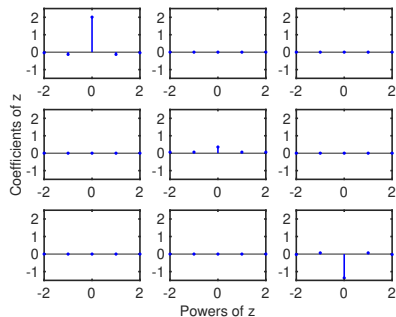


Matrix with polynomial elements.

The same example can be represented as:



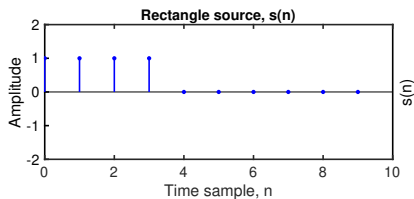
Original $\mathcal{R}_{xx}(z)$.



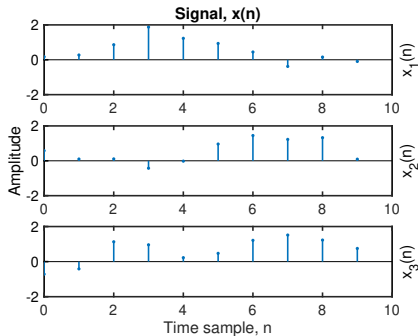
Diagonalized $\Lambda(z)$.

Application Examples

A rectangular pulse source signal arriving at the 3 sensors, corrupted by i.i.d. sensor noise: $\mathcal{N}(0, 0.1^2)$.

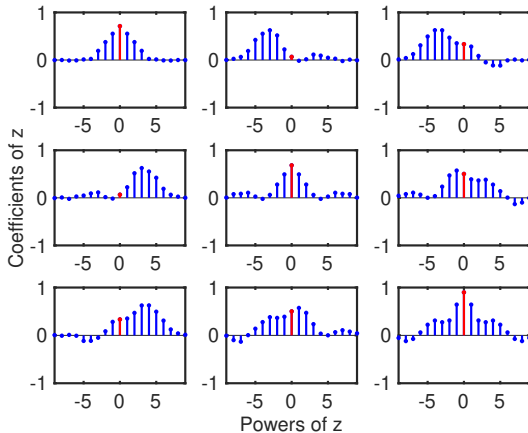


Source signal, $s(n)$.



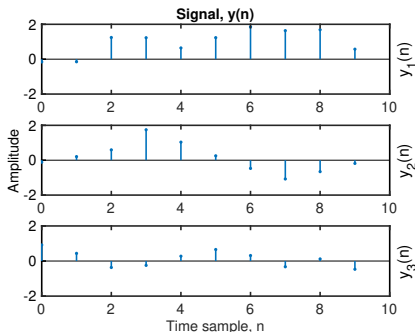
Received signals, $\mathbf{x}(n)$.

Corresponding space-time covariance matrix, $\mathcal{R}_{xx}(z)$

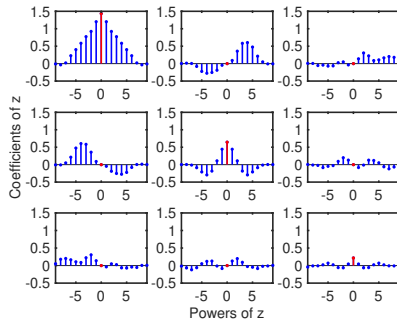


- instantaneous z covariance marked in red.

Using \mathbf{U} from EVD of $\mathbf{R}_{xx}(0)$ gives:



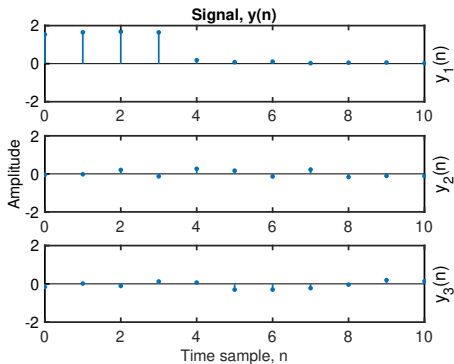
Weighted output, $\mathbf{y}(n)$.



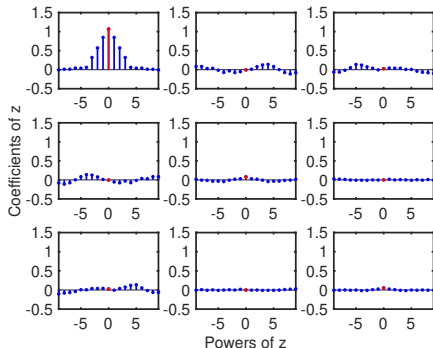
ST-covariance, $\mathcal{R}_{yy}(z)$.

Diagonalization using PEVD with $\delta = 0.0077$ gives:

Using $\mathbf{U}(z)$ from PEVD using $\delta = 0.0077$ gives:



Weighted output, $y(n)$, with arbitrary delays compensated.



ST-covariance, $\mathcal{R}_{yy}(z)$.

Proposed Methodology

If $s(n)$ is a speech signal, uncorrelated with noise

$$\mathbf{R}_{\mathbf{xx}}(z) = \left[\mathbf{u}_S^P(z) \mid \mathbf{u}_V^P(z) \right] \left[\begin{array}{c|c} \mathbf{\Lambda}_S(z) & \mathbf{0} \\ \hline \mathbf{0} & \mathbf{\Lambda}_V(z) \end{array} \right] \left[\begin{array}{c} \mathbf{u}_S(z) \\ \hline \mathbf{u}_V(z) \end{array} \right],$$

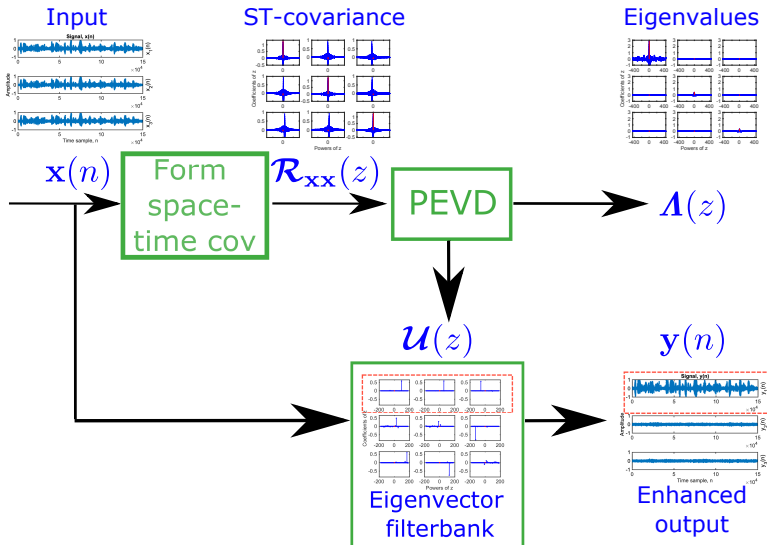
with orthogonal signal, $\{\cdot\}_S$ and noise subspaces, $\{\cdot\}_V$.

The output

$$\mathbf{y}(z) = \mathbf{U}(z)\mathbf{x}(z),$$

has the first element, $y_1(z) \in \mathbb{R}^{1 \times 1 \times T}$, as the denoised speech signal with space-time covariance matrix

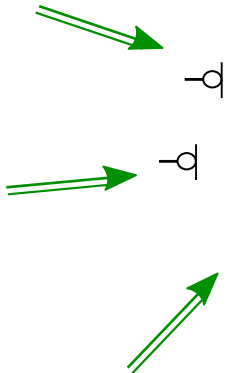
$$\mathbf{R}_{y_1 y_1} = \left[\mathbf{u}_S^P(z) \mid \mathbf{0} \right] \left[\begin{array}{c|c} \mathbf{\Lambda}_S(z) & \mathbf{0} \\ \hline \mathbf{0} & \mathbf{0} \end{array} \right] \left[\begin{array}{c} \mathbf{u}_S(z) \\ \hline \mathbf{0} \end{array} \right].$$



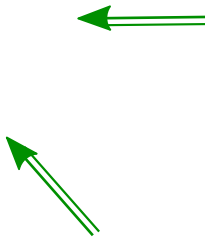
Experimental Results

Speech in Noise (Anechoic)

diffuse babble
5 dB SNR



TIMIT speech

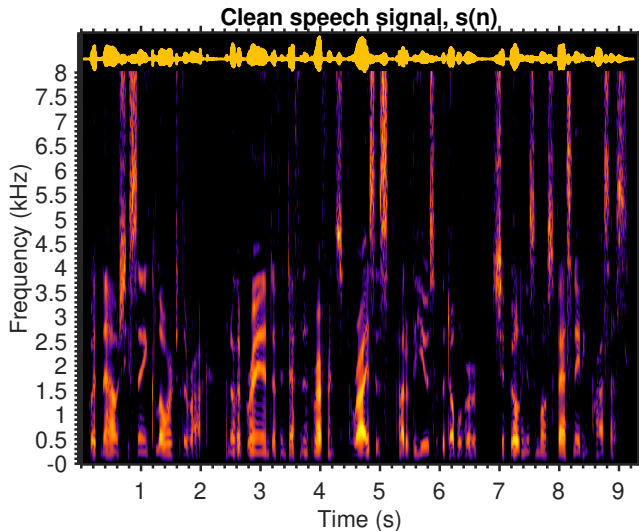


Comparative algorithms:

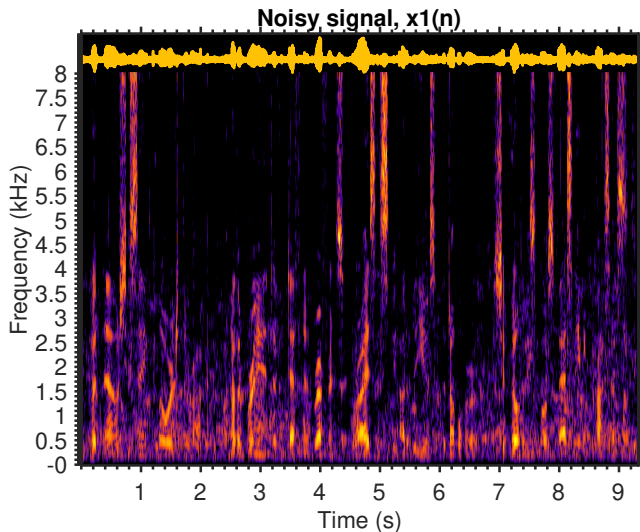
1. Log-Minimum Mean Square Error (Log-MMSE) [Ephraim 1984]
2. Multichannel Wiener Filter (MWF) - Relative Transfer Function (RTF) and noise estimator [Kuklasiński2016]
3. Oracle-MWF (O-MWF) - Given clean speech [Doclo2002]

Evaluation measures:

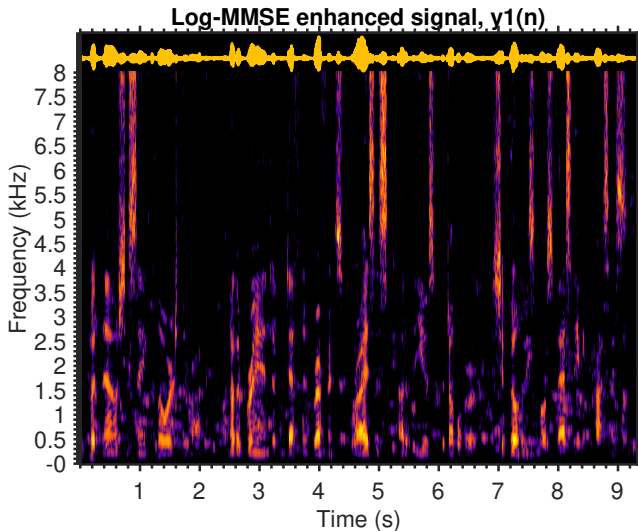
- Segmental SNR (SegSNR)
- Frequency weighted SegSNR (fwSegSNR) [Hu2008]
- STOI [Taal2011]
- PESQ [ITU-T P.862]



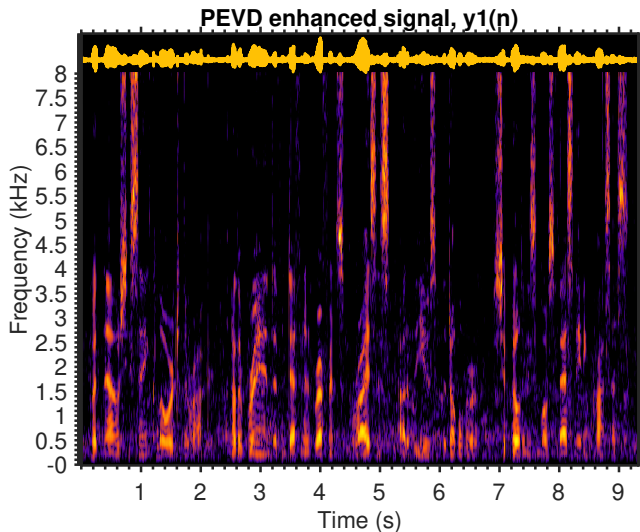
Clean | Noisy | Log-MMSE | PEVD



Clean | Noisy | Log-MMSE | PEVD



Clean | Noisy | Log-MMSE | PEVD



Clean | Noisy | Log-MMSE | PEVD

<i>Algorithm</i>	Δ SegSNR	Δ fwSegSNR	Δ STOI	Δ PESQ
log-MMSE	3.69 dB	2.46 dB	-0.007	0.08
MWF	1.07 dB	1.54 dB	0.002	0.15
O-MWF	4.67 dB	4.04 dB	0.084	0.31
PEVD	4.30 dB	4.00 dB	0.080	0.29

Clean



Noisy



log-MMSE



MWF



O-MWF



PEVD



Conclusion

- Polynomial matrices and PEVD as a tool for processing broadband multichannel signals
 - Polynomial matrices can simultaneously capture the correlation across space, time and frequency
 - PEVD can impose stronger decorrelation than the EVD
- Proposed a speech enhancement algorithm using PEVD
 - Performance approaches oracle MWF when $\text{SNR} \geq 5$ dB
 - No noticeable artifacts



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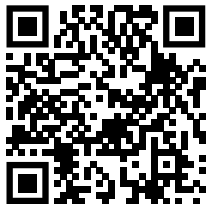


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Thank you



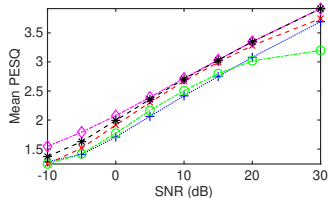
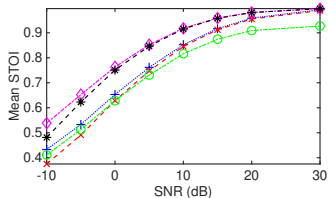
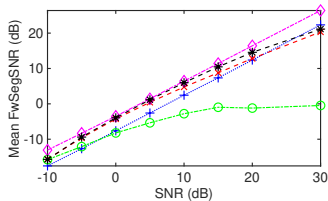
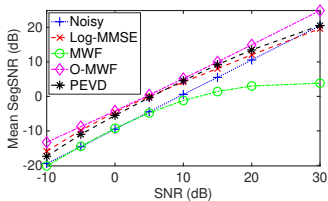


Figure Mean of the results for babble noise involving 150 trials.