

# 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 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  $\tau$  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:

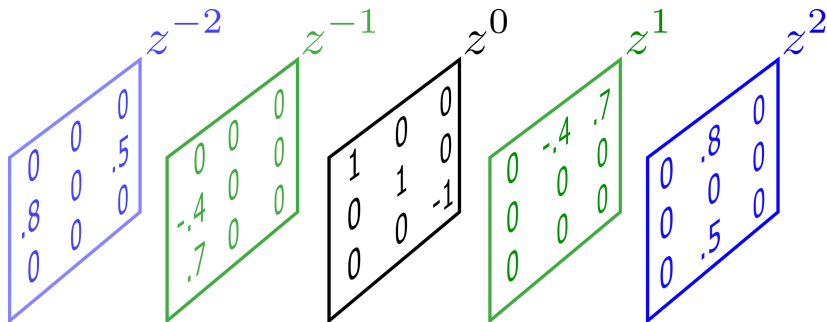
- Second-order Sequential Best Rotation (SBR2) [McWhirter2007]
- Sequential Matrix Diagonalization (SMD) [Redif2015]
- Householder-like PEVD [Redif2011]
- Tridiagonal PEVD [Neo2019]
- Multiple-shift SBR2/SMD [Wang2015; Corr2014]

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

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

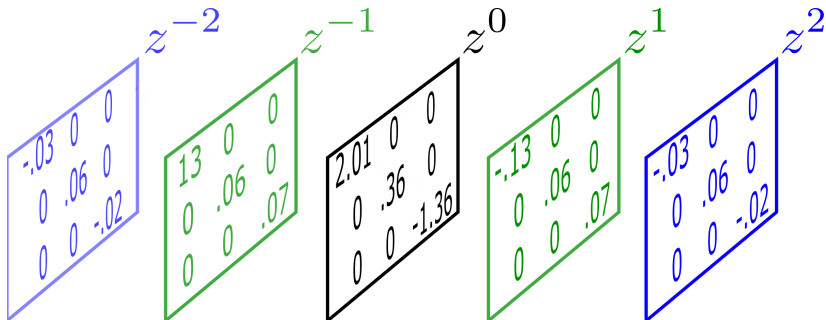
$\mathbf{R}_{\mathbf{x}\mathbf{x}}(0)$ : instantaneous spatial covariance matrix / coefficient of  $z^0$ .

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

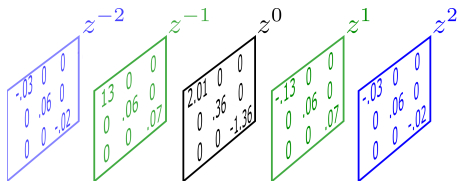


In this example,  $z^0$  plane is diagonal but not at other planes.

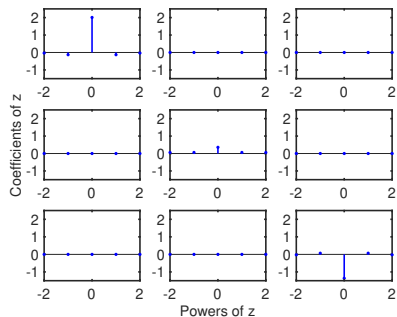
After diagonalization using PEVD,  $\Lambda(z)$ :



Equivalently, expressed as:

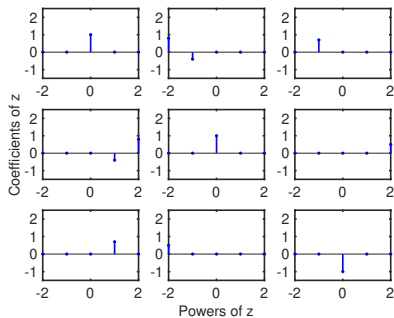


Polynomial with matrix coefficients.

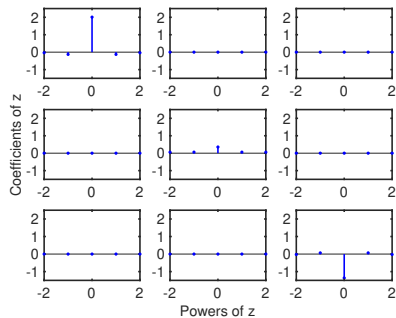


Matrix with polynomial elements.

The same example can be represented as:



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

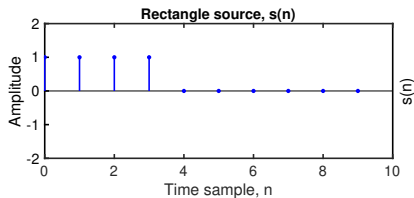


Diagonalized  $\Lambda(z)$ .

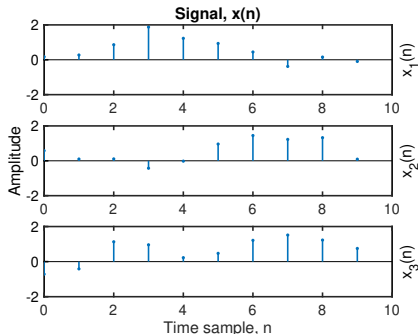
# Application Examples



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

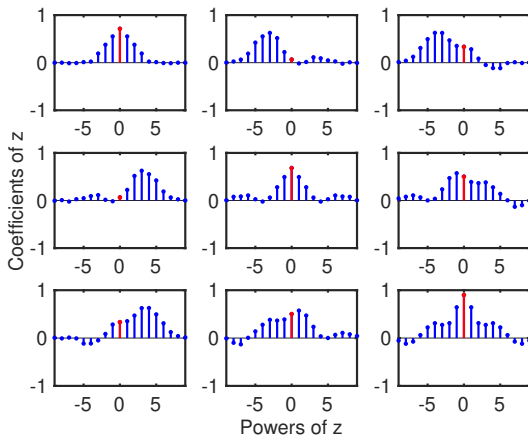


Source signal,  $s(n)$ .



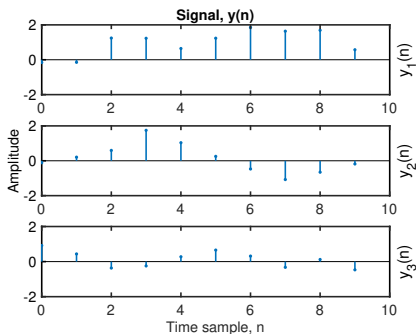
Received signals,  $\mathbf{x}(n)$ .

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

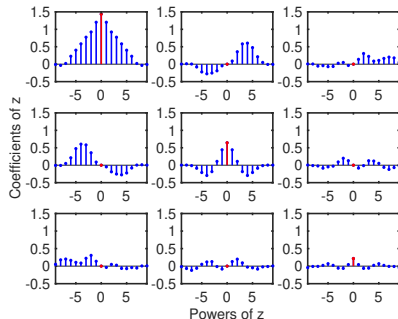


- instantaneous covariance,  $\mathbb{E}[\mathbf{x}(n)\mathbf{x}^H(n)]$ , marked in red.

Using  $\mathbf{U}$  from EVD gives:



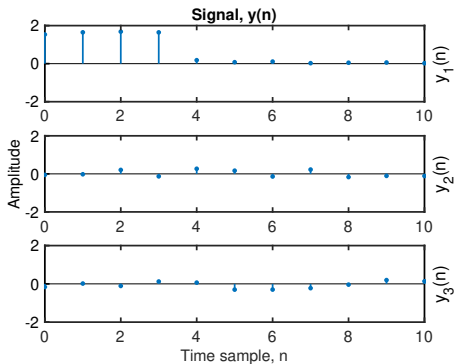
Weighted output,  $\mathbf{y}(n)$ .



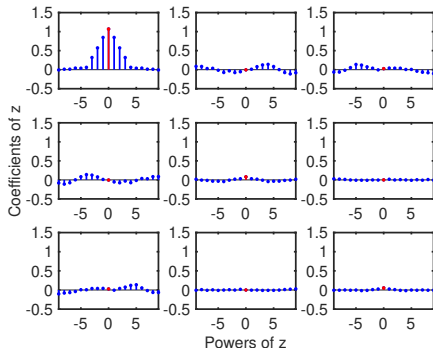
ST-covariance,  $\mathcal{R}_{yy}(z)$ .

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

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

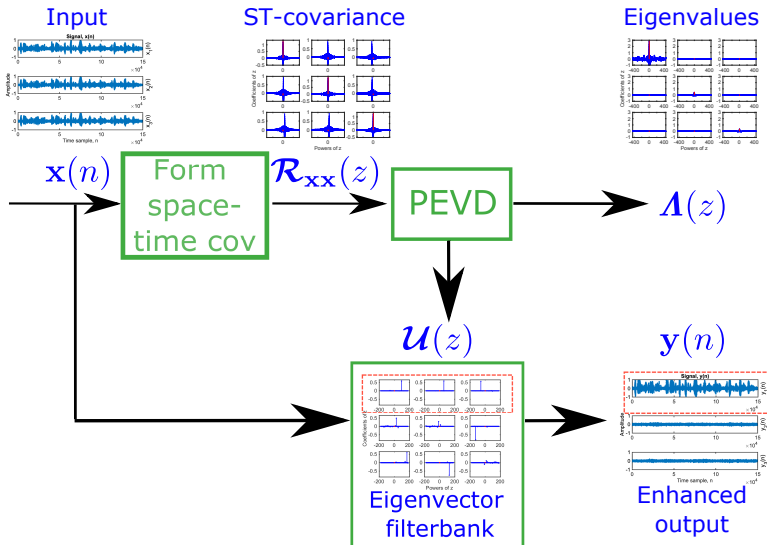
with orthogonal signal,  $\{\cdot\}_S$  and noise subspaces,  $\{\cdot\}_V$ .

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

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

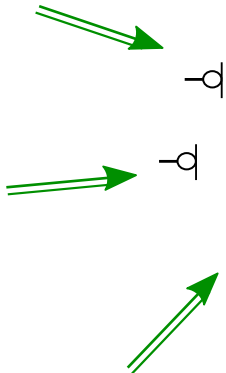




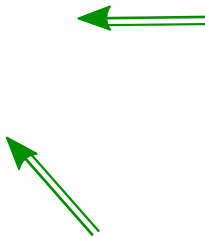
# Experimental Results

## Speech in Noise (Anechoic)

diffuse babble  
5 dB SNR



TIMIT speech

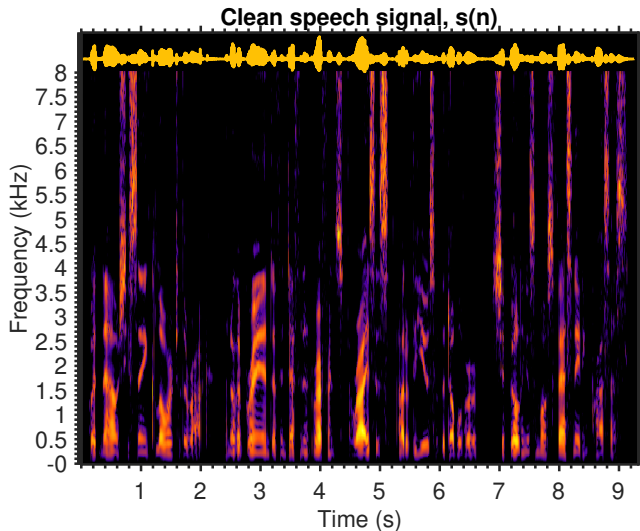


## Comparative algorithms:

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]

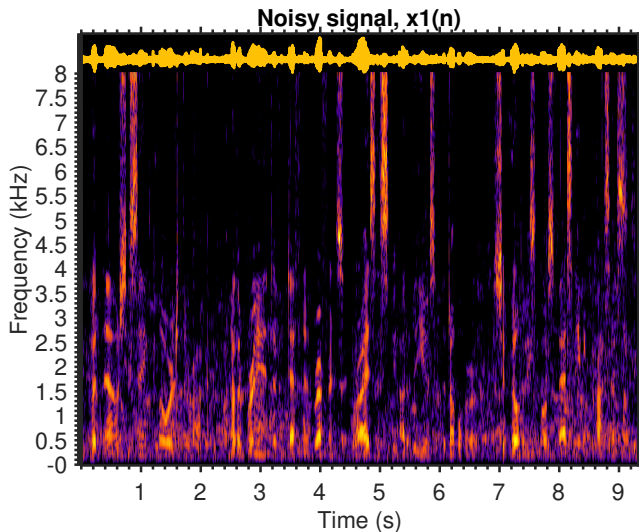
## Evaluation measures:

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

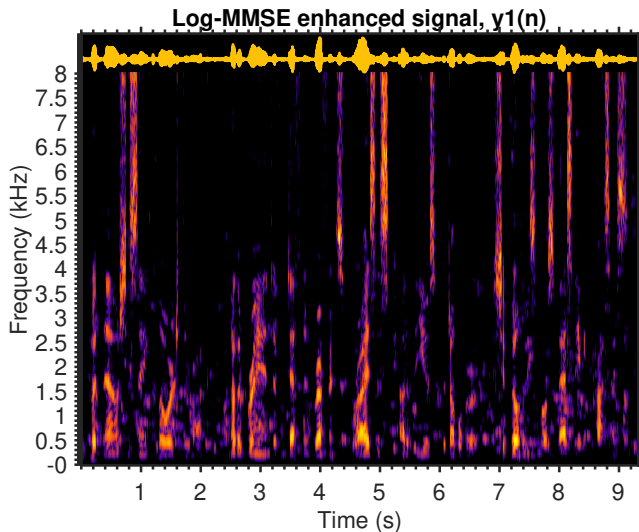


Clean | Noisy | Log-MMSE | PEVD

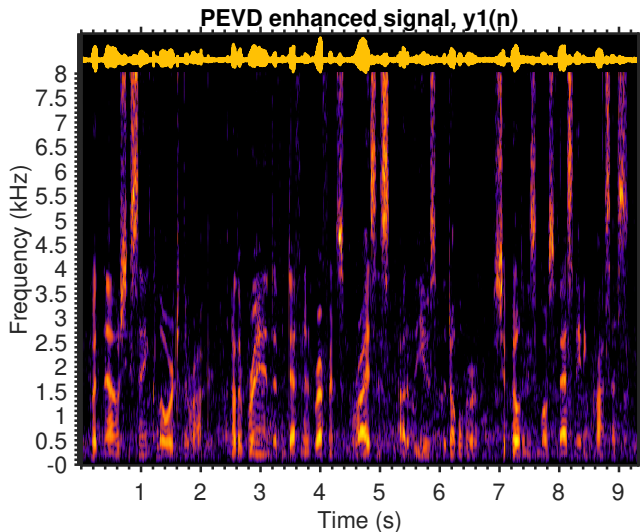
# Noisy Spectrogram (5 dB diffuse babble)



Clean | Noisy | Log-MMSE | PEVD



Clean | Noisy | Log-MMSE | PEVD



Clean | Noisy | Log-MMSE | PEVD

<i>Algorithm</i>	$\Delta$ SegSNR	$\Delta$ fwSegSNR	$\Delta$ STOI	$\Delta$ 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 covariance 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  $\text{SNR} \geq 5 \text{ dB}$
  - No noticeable artifacts



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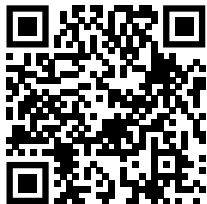


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# Thank you



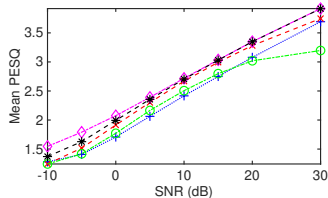
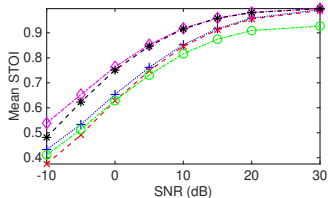
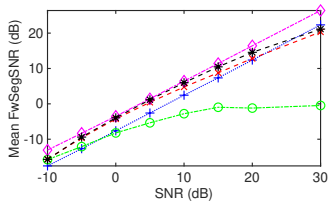
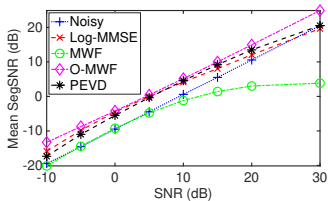


Figure Mean of the results for babble noise involving 150 trials.