

Matrix Analysis and Applications

Chapter 3: Applications of Eigendecomposition

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- 1 Karhunen-Loève Expansion
- 2 Direction-of-Arrival Estimation in Sensor Array Processing
- 3 Orthogonal Frequency Division Multiplexing in Digital Communications

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- 2 Direction-of-Arrival Estimation in Sensor Array Processing
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Preliminaries

Random processes: consider a scalar random process $\{x_n\}$, where $n \in \mathbb{Z}$ (\mathbb{Z} is the set of integers). Let

$$\mu_x(n) \triangleq \mathbb{E}\{x_n\}, \quad (1)$$

$$r_x(n, l) \triangleq \mathbb{E}\{x_n x_l^*\}, \quad (2)$$

$$c_x(n, l) \triangleq \mathbb{E}\{(x_n - \mathbb{E}\{x_n\})(x_l - \mathbb{E}\{x_l\})^*\}, \quad (3)$$

denote the mean, auto-correlation function and auto-covariance function of x_n , respectively, where $\mathbb{E}\{\cdot\}$ denotes the expectation.

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Examples:

- **Discrete-time signal, time series:** x_n is a signal sample or observation at time indicated by n . The functions $r_x(n, l)$ and $c_x(n, l)$ measure how two signal samples at time n, l are related.
- **Image:** x_n stores the value of a pixel, with the pixel's spatial location indicated by n . The functions $r_x(n, l)$ and $c_x(n, l)$ measure the spatial correlations.
- **Data analysis:** x_n is a data point (e.g., score of student n , preference of user n).

Preliminaries (cont'd)

Wide-sense stationary processes: a random process $\{x_n\}$ is called wide-sense stationary (WSS) if, for any n and l , we have

$$\mu_x(n) = \mu_x(l), \quad (4)$$

$$c_x(n, l) = c_x(n + i, l + i). \quad (5)$$

Preliminaries (cont'd)

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- A WSS process is a random process whose first-order and second-order statistics do not change with n (e.g., time in discrete-time signals).
- A zero-mean random process $\{x_n\}$, i.e., $\mu_x(n) = 0$ for all n , is WSS if

$$r_x(n, l) = r_x(n + i, l + i) \quad \text{for any } i.$$

- The same concept applies to a vector random process $\{\mathbf{x}_n\}$.

Preliminaries (cont'd)

Let $\{\mathbf{x}_j\}$, $\mathbf{x}_j \in \mathbb{C}^m$, be a zero-mean complex-valued WSS process.

Correlation matrix:

$$\mathbf{R}_x \triangleq \mathbb{E}\{\mathbf{x}_j \mathbf{x}_j^H\} = \begin{bmatrix} \mathbb{E}\{|x_{1j}|^2\} & \mathbb{E}\{x_{1j}x_{2j}^*\} & \cdots & \mathbb{E}\{x_{1j}x_{mj}^*\} \\ \mathbb{E}\{x_{2j}x_{1j}^*\} & \mathbb{E}\{|x_{2j}|^2\} & \cdots & \mathbb{E}\{x_{2j}x_{mj}^*\} \\ \vdots & \vdots & \ddots & \vdots \\ \mathbb{E}\{x_{mj}x_{1j}^*\} & \cdots & \cdots & \mathbb{E}\{|x_{mj}|^2\} \end{bmatrix}. \quad (6)$$

- **Properties:**

- \mathbf{R}_x is Hermitian (and symmetric for the real-valued case).
- \mathbf{R}_x is positive semidefinite (will be discussed in later lectures).

Representation by Orthonormal Expansion

For a WSS process $\{\mathbf{x}_j\}$, $\mathbf{x}_j \in \mathbb{C}^m$.

Aim: given a basis $\{\mathbf{q}_1, \dots, \mathbf{q}_m\} \subseteq \mathbb{C}^m$, represent each \mathbf{x}_j by

$$\mathbf{x}_n = \sum_{i=1}^m a_{ij} \mathbf{q}_i = \mathbf{Q} \mathbf{a}_j, \quad \text{for some coefficients } a_{ij}, \quad i = 1, \dots, m, \quad (7)$$

where $\mathbf{a}_j = [a_{1j}, \dots, a_{mj}]^T$ and $\mathbf{Q} = [\mathbf{q}_1, \dots, \mathbf{q}_m]$.

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where $\mathbf{a}_j = [a_{1j}, \dots, a_{mj}]^T$ and $\mathbf{Q} = [\mathbf{q}_1, \dots, \mathbf{q}_m]$.

Orthonormal expansion: perform (7) via an orthonormal basis, or unitary \mathbf{Q} .

- Examples: discrete Fourier transform (DFT), discrete cosine transform (DCT), Haar transform.
- For orthonormal expansion, each \mathbf{a}_j can be conveniently obtained via

$$\mathbf{a}_j = \mathbf{Q}^H \mathbf{x}_j. \quad (8)$$

Example: Discrete Fourier Transform (DFT)

The DFT basis matrix is $\mathbf{Q} = [\mathbf{q}_0, \mathbf{q}_1, \dots, \mathbf{q}_{m-1}]$, where

$$\mathbf{q}_i = \frac{1}{\sqrt{m}} \begin{bmatrix} 1 \\ e^{\frac{j2\pi i}{m}} \\ e^{\frac{j4\pi i}{m}} \\ \vdots \\ e^{\frac{j2\pi(m-1)i}{m}} \end{bmatrix}, \quad i = 0, 1, \dots, m-1 \quad (9)$$

and $j = \sqrt{-1}$.

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and $j = \sqrt{-1}$.

Exercise: verify that $\|\mathbf{q}_i\|_2^2 = 1$, and $\mathbf{q}_i^H \mathbf{q}_k = 0$ for all $i \neq k$. Also, if we write $\mathbf{Q}^H = [\mathbf{u}_0, \dots, \mathbf{u}_{m-1}]$, verify that

$$\mathbf{u}_i = \frac{1}{\sqrt{m}} \begin{bmatrix} 1 \\ e^{-j\frac{2\pi i}{m}} \\ e^{-j\frac{4\pi i}{m}} \\ \vdots \\ e^{-j\frac{2\pi(m-1)i}{m}} \end{bmatrix}. \quad (10)$$

Karhunen-Loève Expansion

Assume a zero-mean WSS $\{x_n\}$. From $\mathbf{a}_n = \mathbf{Q}^H \mathbf{x}_n$, one gets

$$\mathbf{R}_a \triangleq \mathbb{E}\{\mathbf{a}_n \mathbf{a}_n^H\} \quad (11)$$

$$= \mathbb{E}\{\mathbf{Q}^H \mathbf{x}_n \mathbf{x}_n^H \mathbf{Q}\} \quad (12)$$

$$= \mathbf{Q}^H \mathbb{E}\{\mathbf{x}_n \mathbf{x}_n^H\} \mathbf{Q} \quad (13)$$

$$= \mathbf{Q}^H \mathbf{R}_x \mathbf{Q}. \quad (14)$$

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$$= \mathbf{Q}^H \mathbf{R}_x \mathbf{Q}. \quad (14)$$

Problem: find a unitary basis matrix \mathbf{Q} such that \mathbf{R}_a is diagonal.

- Having a diagonal \mathbf{R}_a means that the elements of \mathbf{a}_n are **mutually uncorrelated**, which is good for applications such as compression.

Karhunen-Loève Expansion

Assume a zero-mean WSS $\{x_n\}$. From $a_n = Q^H x_n$, one gets

$$R_a \triangleq \mathbb{E}\{a_n a_n^H\} \quad (11)$$

$$= \mathbb{E}\{Q^H x_n x_n^H Q\} \quad (12)$$

$$= Q^H \mathbb{E}\{x_n x_n^H\} Q \quad (13)$$

$$= Q^H R_x Q. \quad (14)$$

Problem: find a unitary basis matrix Q such that R_a is diagonal.

- Having a diagonal R_a means that the elements of a_n are **mutually uncorrelated**, which is good for applications such as compression.

Karhunen-Loève (KL) expansion: set $Q = V$, where V is obtained from the eigendecomposition $R_x = V \Lambda V^H$.

- Under KL expansion, $R_a = \Lambda$ and R_a is diagonal.

More About KL Expansion

Since $\mathbf{R}_a = \mathbf{\Lambda}$, we have

$$\mathbf{R}_a = \begin{bmatrix} \mathbb{E}\{|a_{1n}|^2\} & & & \\ & \mathbb{E}\{|a_{2n}|^2\} & & \\ & & \ddots & \\ & & & \mathbb{E}\{|a_{mn}|^2\} \end{bmatrix} = \begin{bmatrix} \lambda_1 & & & \\ & \lambda_2 & & \\ & & \ddots & \\ & & & \lambda_m \end{bmatrix}, \quad (15)$$

or simply

$$\mathbb{E}\{|a_{in}|^2\} = \lambda_i, \quad i = 1, \dots, m. \quad (16)$$

More About KL Expansion

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or simply

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- The eigenvalues are also the average energies of the KL coefficients.
- There are many situations where the energy in the first few KL coefficients a_{in} dominates that in the remaining ones – which suggests the possibility of approximating $\{\mathbf{x}_n\}$ using fewer number of KL coefficients.

Application Example: Coding and Compression

Problem: encode $\{a_n\}$ with a (much) smaller data size than the original $\{x_n\}$; lossy coding is allowed.

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Transform coding (for generic bases): given a unitary basis matrix \mathbf{Q} ,

- 1) Transform each \mathbf{x}_n to $\mathbf{a}_n = \mathbf{Q}^H \mathbf{x}_n$;
- 2) Select a subset of coefficients in \mathbf{a}_n to encode, e.g., by selecting a number of r coefficients whose magnitudes are the largest;
 - The selected coefficients may also be quantized.
- 3) (Decoding) Reconstruct \mathbf{x}_n approximately by $\hat{\mathbf{x}}_n = \sum_{i \in \mathcal{I}_n} \hat{a}_{in} \mathbf{q}_i$, where $\{\hat{a}_{in}\}_{n \in \mathcal{I}_n}$ and $\mathcal{I}_n \subseteq \{1, \dots, m\}$ are the selected encoded coefficients and its index set, respectively.

KL transform (KLT): choose $\mathbf{Q} = \mathbf{V}$.

Remarks

- 1) To perform KL expansion, information of \mathbf{R}_x is required. One way is to estimate \mathbf{R}_x from data: given a number of N samples $\{\mathbf{x}_n\}_{n=1}^N$, compute the sampled correlation matrix

$$\hat{\mathbf{R}}_x = \frac{1}{N} \sum_{n=1}^N \mathbf{x}_n \mathbf{x}_n^H. \quad (17)$$

- 2) Consider the context of coding
 - While the decoder may not require information of \mathbf{R}_x , it still requires information of the eigen-basis vectors $\mathbf{v}_1, \dots, \mathbf{v}_n$.
 - We can encode $\mathbf{v}_1, \dots, \mathbf{v}_n$ too, but that may incur overhead in data size.
 - On the other hand, one may apply fixed models for \mathbf{R}_x . It has been found that for a certain class of correlation models, DCT is, or well approximates, KLT.
- 3) KLT is almost the same as **principal component analysis (PCA)**, which is extensively used for data analysis and dimensionality reduction.

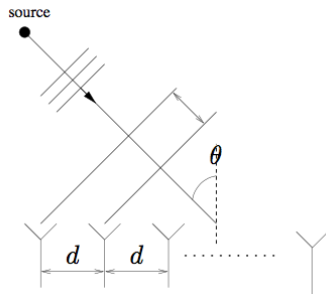
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Sensor Array Processing

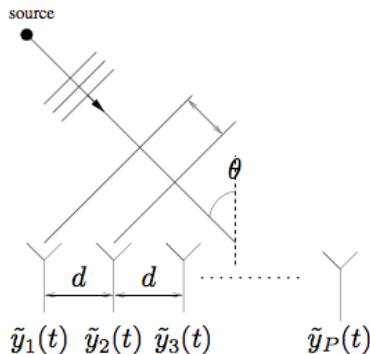
Sensor array processing deals with signal processing problems arising from the use of an array of sensors.

Applications: radar, sonar, wireless communications, seismology, radio astronomy, audio and speech,...



A key problem in this context is **direction-of-arrival (DoA) estimation** of sources.

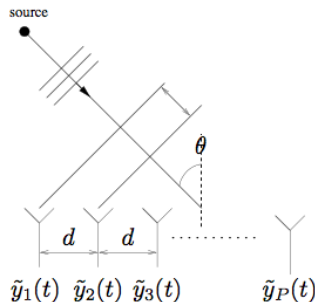
DoA Estimation: Preliminaries



Settings:

- Uniform linear array (ULA) with inter-sensor distance d and with P sensors,
- Far-field wave propagation so that the arriving source waves are planar.

DoA Estimation: Preliminaries (cont'd)



For one radiating source, the output of sensor p can be modeled as

$$\tilde{y}_p(t) = x \left(t - (p-1) \frac{d \sin \theta}{c} \right), \quad (18)$$

where $x(t)$ is the source signal impinging on sensor 1, $\theta \in [-\frac{\pi}{2}, \frac{\pi}{2}]$ denotes the source's DoA, and $c = 3 \times 10^8$ m/s refers to the wave propagation velocity.

DoA Estimation: Preliminaries (cont'd)

Assumption: narrowband carrier-modulated source waveform

$$x(t) = e^{j\omega_c t} s(t), \quad (19)$$

where ω_c is the center frequency, and $s(t)$ is narrowband in the sense that it satisfies

$$s(t - \tau) \simeq s(t), \quad \text{for any } \tau \in [-T_0, T_0] \quad (20)$$

and for some time duration T_0 .

DoA Estimation: Preliminaries (cont'd)

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and for some time duration T_0 .

Let $y_p(t) = e^{-j\omega_c t} \tilde{y}_p(t)$ be a demodulated signal for sensor p . Assuming $T_0 > Pd/c$, we get

$$y_p(t) = e^{-j\omega_c t} x \left(t - (p-1) \frac{d \sin \theta}{c} \right) \quad (21)$$

$$= e^{-j(p-1)\omega_c \frac{d \sin \theta}{c}} s \left(t - (p-1) \frac{d \sin \theta}{c} \right) \quad (22)$$

$$\simeq e^{-j(p-1)\omega_c \frac{d \sin \theta}{c}} s(t). \quad (23)$$

DoA Estimation: Preliminaries (cont'd)

Let $\mathbf{y}(t) = [y_1(t), \dots, y_P(t)]^T$, we know

$$\mathbf{y}(t) = \mathbf{a}(\theta)s(t), \quad (24)$$

where

$$\mathbf{a}(\theta) \triangleq \left[1, e^{-j\phi(\theta)}, e^{-j2\phi(\theta)}, \dots, e^{-j(P-1)\phi(\theta)} \right]^T, \quad (25)$$

is called the **steering vector**, in which

$$\phi(\theta) \triangleq \omega_c \frac{d \sin \theta}{c} = 2\pi \frac{d \sin \theta}{\lambda}, \quad (26)$$

and λ is the wavelength corresponding to the center frequency ω_c .

- To avoid **spatial aliasing**, i.e., $\mathbf{a}(\theta_1) = \mathbf{a}(\theta_2)$ for some $\theta_1 \neq \theta_2$, $\theta_1, \theta_2 \in \left[-\frac{\pi}{2}, \frac{\pi}{2}\right]$, we need

$$d < \frac{\lambda}{2}. \quad (27)$$

DoA Estimation: Problem Setup and Model

Scenario: K radiating sources, noisy observation.

Model: consider discrete-time sampling $\mathbf{y}[n] = \mathbf{y}(nT_s)$, where T_s is the sampling period. The signal $\mathbf{y}[n]$ is modeled as

$$\mathbf{y}[n] = \sum_{k=1}^K \mathbf{a}(\theta_k) s_k[n] + \boldsymbol{\nu}[n] \quad (28)$$

$$= \mathbf{A} \mathbf{s}[n] + \boldsymbol{\nu}[n], \quad n = 1, 2, \dots, \quad (29)$$

where

- $s_k[n]$ is the k^{th} source signal.
- θ_k is the DoA of the k^{th} source.
- $\boldsymbol{\nu}[n]$ is zero-mean spatially white noise, i.e., $\mathbb{E}\{\boldsymbol{\nu}[n]\boldsymbol{\nu}^H[n]\} = \sigma_{\boldsymbol{\nu}}^2 \mathbf{I}$.
- $\mathbf{A} = [\mathbf{a}(\theta_1), \dots, \mathbf{a}(\theta_K)]$.
- $\mathbf{s}[n] = [s_1[n], \dots, s_K[n]]^T$.

DoA Estimation: Problem Setup and Model (cont'd)

Assumptions:

- every $s_k[n]$ is zero-mean WSS.
- $\mathbf{s}[n]$ and $\boldsymbol{\nu}[n]$ are uncorrelated.

The correlation matrix of $\mathbf{y}[n]$ is given by

$$\mathbf{R}_y \triangleq \mathbb{E} \{ \mathbf{y}[n] \mathbf{y}^H[n] \} \quad (30)$$

$$= \mathbb{E} \{ \mathbf{A} \mathbf{s}[n] \mathbf{s}^H[n] \mathbf{A}^H \} + \mathbb{E} \{ \boldsymbol{\nu}[n] \boldsymbol{\nu}^H[n] \} \quad (31)$$

$$= \mathbf{A} \mathbf{R}_s \mathbf{A}^H + \sigma_\nu^2 \mathbf{I}, \quad (32)$$

where $\mathbf{R}_s = \mathbb{E} \{ \mathbf{s}[n] \mathbf{s}^H[n] \}$.

DoA Estimation: Problem Setup and Model (cont'd)

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where $\mathbf{R}_s = \mathbb{E} \{ \mathbf{s}[n] \mathbf{s}^H[n] \}$.

Problem: estimate $\theta_1, \dots, \theta_K$ from \mathbf{R}_y , assuming the model

$$\mathbf{R}_y = \mathbf{A} \mathbf{R}_s \mathbf{A}^H + \sigma_\nu^2 \mathbf{I}. \quad (33)$$

DoA Estimation: Subspace Approach

Idea: use eigendecomposition of \mathbf{R}_y and subspace properties to identify $\theta_1, \dots, \theta_K$.

Assumptions:

- $P > K$ (more sensors than sources).
- $\theta_1, \dots, \theta_K$ are distinct.
- \mathbf{R}_s has full rank (i.e., the sources may be correlated, but not coherent).

Property 1

The array response matrix \mathbf{A} has full rank.

Vandemonde Matrix

A matrix \mathbf{A} is said to be **Vandemonde** if it takes the form

$$\mathbf{A} = \begin{bmatrix} 1 & 1 & \cdots & 1 \\ z_1 & z_2 & \cdots & z_n \\ z_1^2 & z_2^2 & \cdots & z_n^2 \\ \vdots & \cdots & \ddots & \vdots \\ z_1^{m-1} & z_2^{m-1} & \cdots & z_n^{m-1} \end{bmatrix} \in \mathbb{C}^{m \times n}, \quad (34)$$

where $z_1, z_2, \dots, z_n \in \mathbb{C}$.

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where $z_1, z_2, \dots, z_n \in \mathbb{C}$.

Property 2

A Vandemonde matrix with $z_i \neq z_k$ for all $i \neq k$ has full rank.

- The array response matrix \mathbf{A} in DoA estimation is a special case of the Vandemonde matrix where $z_i = e^{-j\phi(\theta_i)}$.
- [Property 1](#) directly follows from [Property 2](#).

DoA Estimation: Subspace Approach

Denote the eigendecomposition of the signal correlation matrix $\mathbf{A}\mathbf{R}_s\mathbf{A}^H$ by

$$\mathbf{A}\mathbf{R}_s\mathbf{A}^H = \mathbf{V}\mathbf{\Lambda}\mathbf{V}^H. \quad (35)$$

Assume $\lambda_1 \geq \lambda_2 \geq \cdots \geq \lambda_P$.

Property 3

The eigenvalues $\lambda_1, \cdots, \lambda_P$ of $\mathbf{A}\mathbf{R}_s\mathbf{A}^H$ satisfy

$$\lambda_i > 0, \quad i = 1, \cdots, K \quad \text{while} \quad \lambda_{K+1} = \cdots = \lambda_P = 0. \quad (36)$$

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The eigenvalues $\lambda_1, \cdots, \lambda_P$ of $\mathbf{A}\mathbf{R}_s\mathbf{A}^H$ satisfy

$$\lambda_i > 0, \quad i = 1, \cdots, K \quad \text{while} \quad \lambda_{K+1} = \cdots = \lambda_P = 0. \quad (36)$$

Property 4

The eigendecomposition of \mathbf{R}_y is

$$\mathbf{R}_y = \mathbf{V}(\mathbf{\Lambda} + \sigma_v^2 \mathbf{I})\mathbf{V}^H, \quad (37)$$

where \mathbf{V} and $\mathbf{\Lambda}$ denote the eigenvector and eigenvalue matrices of $\mathbf{A}\mathbf{R}_s\mathbf{A}^H$, resp.

- **Property 4** means that the eigenvector matrix \mathbf{V} of the signal correlation matrix can be obtained from the eigendecomposition of \mathbf{R}_y .

DoA Estimation: Subspace Approach (cont'd)

Theorem 1

Partition $\mathbf{V} = [\mathbf{V}_1, \mathbf{V}_2]$, where

$$\mathbf{V}_1 = [\mathbf{v}_1, \dots, \mathbf{v}_K], \quad \mathbf{V}_2 = [\mathbf{v}_{K+1}, \dots, \mathbf{v}_P]. \quad (38)$$

The equation

$$\mathbf{V}_2^H \mathbf{a}(\theta) = \mathbf{0}, \quad \theta \in \left[-\frac{\pi}{2}, \frac{\pi}{2}\right] \quad (39)$$

is satisfied if and only if $\theta = \theta_i$ for any $i = 1, \dots, K$.

- The subspace approach amounts to using $\mathbf{V}_2^H \mathbf{a}(\theta) = \mathbf{0}$ (or its approximation) to identify the source DoAs.
- There are many ways to do so, leading to many algorithms.
- A classical and most popular one is multiple signal classification (MUSIC).

DoA Estimation: The MUSIC Algorithm

MUSIC: Multiple Signal Classification¹

- **Step1.** compute the sample correlation matrix $\hat{\mathbf{R}}_y = \frac{1}{N} \sum_{n=1}^N \mathbf{y}_n[n] \mathbf{y}_n^H[n]$.
- **Step2.** compute the eigenvector matrix of $\hat{\mathbf{R}}_y$, denoted by $\hat{\mathbf{V}}$.
- **Step3.** evaluate the 'pseudo-spectrum'

$$P_{\text{MUSIC}}(\theta) = \frac{1}{\|\hat{\mathbf{V}}_2^H \mathbf{a}(\theta)\|_2^2}, \quad \text{for } \theta \in \left[-\frac{\pi}{2}, \frac{\pi}{2}\right], \quad (40)$$

where $\hat{\mathbf{V}} = [\hat{\mathbf{v}}_{K+1}, \dots, \hat{\mathbf{v}}_P]$, and determine the DoAs by finding the peaks of $P_{\text{MUSIC}}(\theta)$.

¹R. O. Schmidt, "Multiple emitter location and signal parameter estimation," *IEEE Transactions on Antenna Propagation*, vol. AP-34, pp. 276–290, March 1986.

DoA Estimation: Numerical Demonstration

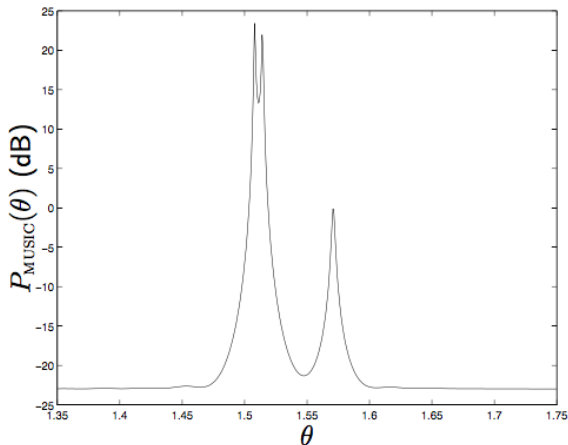


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Preliminary: Circulant Matrix

A matrix \mathbf{H} is said to be **circulant** if it takes the following structure

$$\mathbf{H} = \begin{bmatrix} h_0 & h_{N-1} & \cdots & h_2 & h_1 \\ h_1 & h_0 & \cdots & h_3 & h_2 \\ \vdots & \vdots & \ddots & \vdots & \vdots \\ h_{N-1} & h_{N-2} & \cdots & h_1 & h_0 \end{bmatrix} \quad (41)$$

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Eigenvectors: let $\mathbf{f}_k \triangleq \frac{1}{\sqrt{N}} \left[1, e^{j\frac{2\pi k}{N}}, \dots, e^{j\frac{2\pi(N-1)k}{N}} \right]^T \in \mathbb{C}^N$. It can be verified that

$$\mathbf{H}\mathbf{f}_k = d_k\mathbf{f}_k, \quad k = 0, \dots, N-1, \quad (42)$$

where

$$d_k = \frac{1}{\sqrt{N}} \sum_{n=0}^{N-1} h_n e^{-j\frac{2\pi nk}{N}}. \quad (43)$$

- $\{d_k\}$ is the normalized DFT of $\{h_n\}$.

Preliminary: Circulant Matrix (cont'd)

Let $\mathbf{F} \triangleq [\mathbf{f}_0, \dots, \mathbf{f}_{N-1}] \in \mathbb{C}^{N \times N}$.

- \mathbf{F} is unitary.
- \mathbf{F} is seen to be the inverse DFT (IDFT) matrix, while $\mathbf{F}^{-1} = \mathbf{F}^H$ the DFT matrix.
- DFT and IDFT can be efficiently computed (complexity order of $N \log(N)$).

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By the unitarity of \mathbf{F} , a circulant \mathbf{H} admits an eigendecomposition

$$\mathbf{H} = \mathbf{F} \mathbf{D} \mathbf{F}^H, \quad (44)$$

$$\mathbf{D} = \text{diag}(d_0, \dots, d_{N-1}). \quad (45)$$

- Application: **orthogonal frequency division multiplexing (OFDM)** in wireless communications.

OFDM: Background

Scenario: digital transmission over a linear time-invariant channel.

System Model:

$$y[n] = \sum_{l=0}^L h_l s[n-l] + \nu[n], \quad n \in \mathbb{Z}, \quad (46)$$

where

- $s[n]$: the transmitted data symbol stream,
- $y[n]$: the received signal (discrete-time sampled),
- h_l : the channel impulse response, with length $L + 1$,
- $\nu[n]$: AWGN noise.

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Problem formulation:

Originally, we wish to detect each $s[n]$ from $y[n]$ – which works if $h_0 \neq 0$, $h_l = 0$ for $l > 1$. However, in general, we have **intersymbol interference (ISI)**

$$y[n] = h_0 s[n] + \underbrace{\sum_{l \neq 0} h_l s[n-l]}_{\text{ISI}} + \nu[n]. \quad (47)$$

OFDM

Aim: precode $\{s[n]\}$ so that it can be received in a zero-ISI manner.

Consider a block transmission, where transmission is parsed into time blocks, and every block has N symbols, $s[0], s[1], \dots, s[N-1]$, to transmit.

Model for one time block:

$$y[n] = \sum_{l=0}^L h_l \tilde{x}[n-l] + \nu[n], \quad n = 0, 1, \dots, N+L-1, \quad (48)$$

where

$$\left. \begin{array}{l} \tilde{x}[0] = x[N-L] \\ \tilde{x}[1] = x[N-L+1] \\ \vdots \\ \tilde{x}[L-1] = x[N-1] \end{array} \right\} \text{cyclic prefix} \quad \begin{array}{l} \tilde{x}[L] = x[0] \\ \tilde{x}[L+1] = x[1] \\ \vdots \\ \tilde{x}[N+L-1] = x[N-1] \end{array}$$

and $x[n]$ is the precoded signal, which will be specified shortly.

OFDM (cont'd)

It can be verified that

$$\underbrace{\begin{bmatrix} y[L] \\ y[L+1] \\ \vdots \\ y[N+L-1] \end{bmatrix}}_{\triangleq \mathbf{y} \in \mathbb{C}^N} = \underbrace{\begin{bmatrix} h_0 & \cdots & \cdots & h_L & \cdots & h_1 \\ h_1 & h_0 & \cdots & \cdots & \ddots & \vdots \\ \vdots & \vdots & \ddots & \cdots & \cdots & h_L \\ h_L & & & \ddots & \ddots & \\ & \ddots & & h_L & \cdots & h_1 & h_0 \end{bmatrix}}_{\triangleq \mathbf{H} \in \mathbb{C}^{N \times N}} \underbrace{\begin{bmatrix} x[0] \\ x[1] \\ \vdots \\ x[N-1] \end{bmatrix}}_{\triangleq \mathbf{x} \in \mathbb{C}^N} + \underbrace{\begin{bmatrix} \nu[L] \\ \nu[L+1] \\ \vdots \\ \nu[N+L-1] \end{bmatrix}}_{\triangleq \boldsymbol{\nu} \in \mathbb{C}^N}, \quad (49)$$

where a linear system $\mathbf{y} = \mathbf{H}\mathbf{x} + \boldsymbol{\nu}$ with a circulant system response \mathbf{H} is obtained, in other words, we have

$$\mathbf{y} = \mathbf{F}\mathbf{D}\mathbf{F}^H \mathbf{x} + \boldsymbol{\nu}. \quad (50)$$

OFDM (cont'd)

- 1) At the transmitter side (IDFT), let $\mathbf{s} = [s[0], \dots, s[N-1]]^T$.

OFDM: choose $\mathbf{x} = \mathbf{F}\mathbf{s}$ (i.e., IDFT the symbol block).

The system model reduces to

$$\mathbf{y} = \mathbf{F}\mathbf{D}\mathbf{F}^H\mathbf{x} + \boldsymbol{\nu} = \mathbf{F}\mathbf{D}\mathbf{F}^H\mathbf{F}\mathbf{s} + \boldsymbol{\nu} = \mathbf{F}\mathbf{D}\mathbf{s} + \boldsymbol{\nu}. \quad (51)$$

- 2) At the receiver side (DFT), consider a transformation $\mathbf{z} = \mathbf{F}^H\mathbf{y}$ (i.e., DFT \mathbf{y} , done by the receiver). Since

$$\mathbf{z} = \mathbf{D}\mathbf{s} + \mathbf{F}^H\boldsymbol{\nu} \iff z_k = d_k s_k + [\mathbf{F}^H\boldsymbol{\nu}]_k, \quad k = 0, \dots, N-1, \quad (52)$$

zero ISI is achieved.

- The noise term $\mathbf{F}^H\boldsymbol{\nu}$ is only rotated (by the unitarity of \mathbf{F}), but not magnified.
- A related scheme is single-carrier modulation (SCM), which chooses $\mathbf{x} = \mathbf{s}$. The receiver can achieve zero ISI by $\mathbf{z} = \mathbf{F}\mathbf{D}^{-1}\mathbf{F}^H\mathbf{y}$ (can be efficiently computed via DFT/IDFT), assuming an invertible \mathbf{D} . However, noise may be magnified.

OFDM (cont'd)

Remark 1

- 1) *To capture the central idea, we have introduced OFDM using a simplified model, namely, no interference between time blocks is assumed, but actually there is.*
 - *The inter-block interference can be easily handled by the so-called guard time removal; find textbooks for details, see e.g., [R1].²*
- 2) *OFDM (also SCM) uses a block length of $N + L$ to transmit N symbols. For good efficiency, N should be much greater than L .*

²[R1] J. Li, X. Wu, and R. Laroia, *OFDMA Mobile Broadband Communications: A Systems Approach*, Cambridge University Press, 2013.

**Thank you
for your attention!**



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