

EE4C03 STATISTICAL DIGITAL SIGNAL PROCESSING AND MODELING

4 November 2020, 13:30–16:30

Block 1 (13:30-15:00)

Open book, strictly timed take-home exam. (Electronic) copies of the book and the course slides allowed. No other tools except a basic pocket calculator permitted.

Upload answers during 14:55–15:05

This block consists of three questions (25 points); more than usual, and this will be taken into account during grading. Answer in Dutch or English. Make clear in your answer how you reach the final result; the road to the answer is very important. Write your name and student number on each sheet.

Hint: Avoid losing too much time on detailed calculations, write down the general approach first.

Question 1 (8 points)

- (a) Let $x(n) = 4 + v(n)$, where $v(n)$ is real-valued zero mean i.i.d. noise with variance $\sigma_v^2 = 1$. The signal is filtered by an FIR filter with impulse response

$$h(n) = \begin{cases} 1 & n = 0, \\ -\frac{1}{2} & n = 1, \\ 0 & \text{otherwise.} \end{cases}$$

The output of the filter is $y(n) = h(n) * x(n)$.

Compute the mean value $m_y(n) = E\{y(n)\}$ and the autocovariance sequence $c_y(k)$.

- (b) For a zero mean WSS process, let the autocorrelation sequence be given by

$$r(k) = A \cos(\omega_0 k),$$

and construct the $N \times N$ correlation matrix \mathbf{R} .

Derive that the rank of \mathbf{R} is equal to 2.

Hint: first write $\cos(\omega_0 k)$ as the sum of two exponentials.

- (c) *True or False:* is this a valid autocorrelation matrix?

$$\mathbf{R} = \begin{bmatrix} 2 & 2 & 0 & 0 \\ 2 & 2 & 2 & 0 \\ 0 & 2 & 2 & 2 \\ 0 & 0 & 2 & 2 \end{bmatrix}$$

(Motivate your answer.)

- (d) Suppose we have N samples of a zero mean WSS random process $x(n)$, but with one sample missing:

$$\mathbf{x} = [x(0), x(1), \dots, x(n_0 - 1), x(n_0 + 1), \dots, x(N - 1)]^T$$

As usual, denote the correlation sequence by $r(k) = E\{x(n+k)x^*(n)\}$, and denote the correlation matrix $\mathbf{R} = E\{\mathbf{x}\mathbf{x}^H\}$.

Write down \mathbf{R} in terms of $r(k)$. Is this a Toeplitz matrix? And can the usual correlation matrix corresponding to a sequence with no missing samples be reconstructed?

- (e) *True or False:* if $y(n) = x_1(n) + x_2(n)$, and $x_1(n)$ is an AR(1) process and $x_2(n)$ is an AR(2) process, then $y(n)$ is an AR(3) process. (Motivate your answer.)

Question 2 (8 points)

Let

$$x(n) = \sin(\omega_0 n + \phi_0),$$

where ϕ_0 is a random variable, uniformly distributed in the range $[-\pi, \pi]$, and ω_0 is fixed.

- (a) Show that $x(n)$ is zero mean, and derive that the autocovariance sequence of $x(n)$ is

$$r_x(k) = \frac{1}{2} \cos(\omega_0 k).$$

- (b) Prove that $x(n)$ satisfies the identity

$$x(n) - 2\cos(\omega_0)x(n-1) + x(n-2) = 0$$

Hint: $\sin(A+B) = \sin(A)\cos(B) + \cos(A)\sin(B)$.

- (c) Explain that this implies that $x(n)$ is “*deterministic*” (predictable).
 (d) For a specific ω_0 , the first few samples of $r_x(k)$ are

$$r_x(0) = 0.5, \quad r_x(1) = 0.25, \quad r_x(2) = -0.25, \quad r_x(3) = -0.5 \quad \dots$$

Using the Schur algorithm, compute, step by step, the reflection coefficients Γ_1 and Γ_2 .

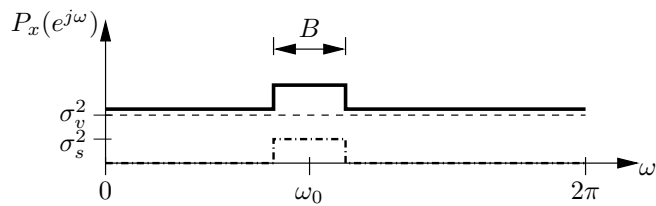
- (e) Looking at $|\Gamma_2|$, what can you conclude?
 (f) Suppose now that $x(n)$ is the sum of 2 sinusoids. What can you say about the reflection coefficients?

Question 3 (9 points)

We are given 2000 samples of a single signal in noise,

$$x(n) = s(n) + v(n)$$

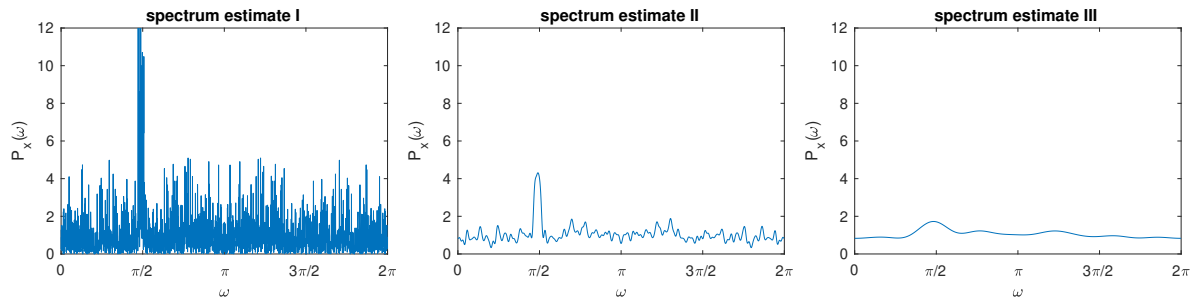
where $v(n)$ is zero mean white Gaussian noise with variance $\sigma_v^2 = 1$, while $s(n)$ is a weak astronomical signal, modeled as zero mean Gaussian noise with power spectrum $P_s(e^{j\omega})$, the dash-dot line in the figure. Based on physics, we expect the signal at $\omega_0 = 1.5$ rad and with a bandwidth $B = 0.16$ rad. We do not know its power in the passband.



We wish to detect the presence of the astronomical signal from a Bartlett spectrum estimate $\hat{P}_B(e^{j\omega})$. The $N = 2000$ samples are split into K blocks of L samples, $KL = 2000$. Three different options for (L, K) are considered:

- (i) $L = 10, K = 200$; (ii) $L = 100, K = 20$; (iii) $L = 2000, K = 1$.

Assume for the moment $\sigma_s^2 = 4$. The resulting spectrum estimates for $P_x(e^{j\omega})$, in random order, are:



- Which resolution is obtained for each choice of (L, K) ? Also indicate how this relates to B .
- What is the variance on $\hat{P}_B(e^{j\omega})$ in each case? (Make a distinction between outside/inside the passband of $P_s(e^{j\omega})$.)
- Which spectrum estimate corresponds to which parameter set (L, K) ?
- Explain the observed differences between spectrum *I* and *II*. In particular, explain why the height of the observed peak in spectrum *II* is much lower than that in spectrum *I*.
- Explain the observed differences between spectrum *II* and *III*. In particular, explain why the height of the observed peak in spectrum *III* is much lower than that in spectrum *II*.
- Which spectrum estimate would you use to estimate σ_s^2 , and how would you estimate it? What is the variance of that estimate?

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Block 2 (15:20-16:50)

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Upload answers during 16:45–16:55

This block consists of three questions (25 points); more than usual, and this will be taken into account during grading. Answer in Dutch or English. Make clear in your answer how you reach the final result; the road to the answer is very important. Write your name and student number on each sheet.

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Question 4 (9 points)

Consider a stochastic process consisting of a complex exponential in complex noise. Mathematically, this can be written as

$$x(n) = \alpha e^{j\omega_0 n} + \beta v(n), \quad (1)$$

where α and β are some weighting parameters, ω_0 is the frequency of the complex exponential and $v(n)$ is complex white Gaussian noise with mean 0 and variance 1.

We will fit an all-pole model of order two to this process, which can be written as

$$H(z) = \frac{b(0)}{1 + a(1)z^{-1} + a(2)z^{-2}},$$

with filter coefficients $b(0)$, $a(1)$ and $a(2)$.

Let us first assume that $\alpha = 1$ and $\beta = 0$, i.e., $x(n) = e^{j\omega_0 n}$, and that $N + 1$ samples from $x(n)$ are given, i.e., $x(0), x(1), \dots, x(N)$.

- (a) Use the autocorrelation method based on these $N + 1$ samples to compute $a(1)$ and $a(2)$. Provide all the steps of the derivation.

Hint: The inverse of the matrix $\begin{bmatrix} a & b \\ c & d \end{bmatrix}$ is given by $\frac{1}{ad-bc} \begin{bmatrix} d & -b \\ -c & a \end{bmatrix}$.

- (b) Use the covariance method based on these $N + 1$ samples to compute $a(1)$ and $a(2)$. Provide all the steps of the derivation.

Let us now assume the general case where both α and β are non-zero.

- (c) Compute the correlation sequence of $x(n)$ based on the model (1). Is the process $x(n)$ wide-sense stationary? Why or why not?

- (d) Use the Yule Walker model to compute $a(1)$ and $a(2)$. Provide all steps.
- (e) If you set $\alpha = 1$ and $\beta = 0$, how does the solution of (e) look like? Does it correspond to the solution of (a) and/or (b) above?
- (f) If you now set $\alpha = 0$ and $\beta = 1$, how does the solution of (e) look like? Can you interpret that solution? Or in other words, is this the solution you would expect?

Question 5 (8 points)

In this question, we will develop the concept of optimal filtering but applied to compressed signals. Assume all signals are real-valued. Let us consider a stationary stochastic process $x(n)$ which depends on some stationary desired signal $d(n)$. The auto-correlation sequence of $x(n)$ is given by $r_x(k) = E\{x(n)x(n-k)\}$ and the cross-correlation sequence between $d(n)$ and $x(n)$ by $r_{dx}(k) = E\{d(n)x(n-k)\}$.

The signal $x(n)$ is now compressed using the following steps: i) $x(n)$ is split into non-overlapping windows of size N , denoted as $\mathbf{x}_i = [x(iN), x(iN+1), \dots, x(iN+N-1)]^T$; ii) every window \mathbf{x}_i is compressed by applying an $M \times N$ matrix \mathbf{C} to it ($M < N$), leading to a smaller window $\mathbf{y}_i = \mathbf{C}\mathbf{x}_i$; iii) the smaller windows \mathbf{y}_i are concatenated leading to the sequence $y(m)$. In short, every N samples of $x(n)$ are transformed into $M < N$ samples of $y(m)$ using the matrix \mathbf{C} .

- (a) Express the auto-correlation matrix $\mathbf{R}_x = E\{\mathbf{x}_i\mathbf{x}_i^T\}$ as a function of $r_x(k)$.
Hint: Focus on a single entry of the matrix and compute the related correlation.
- (b) If we define the windowed version of $d(n)$ as $\mathbf{d}_i = [d(iN), d(iN+1), \dots, d(iN+N-1)]^T$, express the cross-correlation matrix $\mathbf{R}_{dx} = E\{\mathbf{d}_i\mathbf{x}_i^T\}$ as a function of $r_{dx}(k)$.
Hint: Use the same hint as in (a).
- (c) Express the auto-correlation matrix $\mathbf{R}_y = E\{\mathbf{y}_i\mathbf{y}_i^T\}$ as a function of \mathbf{R}_x from (a). Similarly, express the cross-correlation matrix $\mathbf{R}_{dy} = E\{\mathbf{d}_i\mathbf{y}_i^T\}$ as a function of \mathbf{R}_{dx} from (b).
- (d) Suppose we want to estimate \mathbf{d}_i from \mathbf{y}_i using an $M \times N$ filter \mathbf{W} such that $\hat{\mathbf{d}}_i = \mathbf{W}^T\mathbf{x}_i$. Find the optimal filter \mathbf{W} by solving

$$\min_{\mathbf{W}} E\{\|\mathbf{W}^T\mathbf{y}_i - \mathbf{d}_i\|^2\}.$$

Give all the steps of the derivation and express the solution as a function of \mathbf{R}_x and \mathbf{R}_{dx} using the results derived in (c).

Hint: Solve the problem for the k th filter \mathbf{w}_k (k th column of \mathbf{W}) which is used to estimate $d(iN+k-1)$ (k th entry of \mathbf{d}_i). Then stack all the solutions.

- (e) Suppose now that $x(n) = d(n)$ and both are white Gaussian noise with mean 0 and variance 1. How can then $\hat{\mathbf{d}}_i$ be expressed as a function of \mathbf{d}_i ? Is $\hat{\mathbf{d}}_i = \mathbf{d}_i$? Why or why not?
- (f) Based on your derivation in (d), express the steepest gradient descent recursion to compute \mathbf{W} . From that recursion, derive the LMS algorithm to adaptively compute \mathbf{W} using knowledge of \mathbf{y}_i and \mathbf{d}_i .

Question 6 (8 points)

Suppose we want to use spectral estimation techniques for localizing an acoustic source that is picked up by a number of microphones. The source is at position ϕ_0 (we simply use a scalar index here to indicate the position) and transmits a white Gaussian signal $s(n)$ with mean 0 and variance 1. Microphone k , with $k = 1, 2, \dots, K$, is at position ϕ_k and picks up the signal

$$x_k(n) = a_k(\phi_0)s(n) + v_k(n),$$

where $a_k(\phi)$ is the attenuation from position ϕ to position ϕ_k , and $v_k(n)$ is spatially and temporally white Gaussian noise with mean 0 and variance σ^2 . We also define the column vectors $\mathbf{x}(n) = [x_1(n), \dots, x_K(n)]^T$, $\mathbf{v}(n) = [v_1(n), \dots, v_K(n)]^T$, and $\mathbf{a}(\phi) = [a_1(\phi), \dots, a_K(\phi)]^T$. Although not true in practice, we assume that $\mathbf{a}(\phi)$ is normalized for every position ϕ , i.e., $\|\mathbf{a}(\phi)\| = 1, \forall \phi$.

- (a) Express $\mathbf{x}(n)$ as a function of $\mathbf{v}(n)$, $\mathbf{a}(\phi_0)$, and $s(n)$. Based on this model, compute the correlation matrix of $\mathbf{x}(n)$, denoted by $\mathbf{R}_x = E\{\mathbf{x}(n)\mathbf{x}^T(n)\}$.
- (b) If the noise variance $\sigma^2 = 0$, then what is the rank of \mathbf{R}_x and why?

We now want to estimate the source position ϕ_0 using spectral estimation techniques. Therefore, we first consider an arbitrary position ϕ and we apply the *position-dependent* filter \mathbf{w} to $\mathbf{x}(n)$ leading to the output $y(n) = \mathbf{w}^T \mathbf{x}(n)$. The output power $P(\phi) = \mathbf{w}^T \mathbf{R}_x \mathbf{w}$, which can be interpreted as the “position” spectrum, can then be maximized to estimate the source position ϕ_0 .

- (c) Give the expression for the filter \mathbf{w} related to the periodogram. Using (a), give the expression for $P(\phi)$. What is the maximum of this spectrum and for which value of ϕ is it obtained?
- (d) Give the expression for the filter \mathbf{w} related to the minimum variance spectral estimation method. Using (a), give the expression for $P(\phi)$. What is the maximum of this spectrum and for which value of ϕ is it obtained?

Hint: You will have to use a specific form of the matrix inversion lemma, i.e., $(c + \mathbf{xx}^T)^{-1} = c^{-1}\mathbf{I} - c^{-1}(c + \mathbf{x}^T \mathbf{x})^{-1} \mathbf{xx}^T$.

For finding the filter related to the MUSIC method, we solve the following problem. For every source position ϕ we minimize the distance between \mathbf{w} and $\mathbf{a}(\phi)$ subject to the fact that \mathbf{w} is orthogonal to the noise subspace of \mathbf{R}_x described by the $K \times (K - r)$ matrix \mathbf{U}_n [note that r is the rank you have to derive in (b)]. This problem can be mathematically formulated as

$$\min_{\mathbf{w}} \|\mathbf{w} - \mathbf{a}(\phi)\|^2 \quad \text{subject to} \quad \mathbf{U}_n^T \mathbf{w} = \mathbf{0}. \quad (2)$$

- (e) Solve problem (2) using the method of Lagrange multipliers. Note that you have $K - r$ linear constraints so your Lagrange multiplier is a vector $\boldsymbol{\lambda}$ of size $K - r$. Prove that the solution is given by $\mathbf{w} = (\mathbf{I} - \mathbf{U}_n \mathbf{U}_n^T) \mathbf{a}(\phi)$.
- (f) The useful signal energy of this MUSIC filter for a source at position ϕ is given by $|\mathbf{w}^T \mathbf{a}(\phi)|^2$. Show that maximizing this useful signal energy is the same as maximizing the so-called MUSIC spectrum given by $[\mathbf{a}^T(\phi) \mathbf{U}_n \mathbf{U}_n^T \mathbf{a}(\phi)]^{-1}$.