

Spectral Estimation

87203 – Multimedial Signal Processing 1st Module

Politecnico di Milano – Polo regionale di Como



NON PARAMETRIC SPECTRAL ESTIMATION

Ex1: Periodogram

Ex2: Correlogram

Ex3: Blackman-Tukey

Ex4: Bartlett

Ex5: Welch

PARAMETRIC SPECTRAL ESTIMATION

Ex6: AutoRegressive model



Exercise 1 (1/4)

- Goal of spectral estimation: from one finite record of a stationary data sequence, estimate how the total power is distributed over frequency.
- Power Spectral Density

■ DTFT of the ACS:
$$\phi(\omega) = \sum_{k=-\infty}^{\infty} r(k) e^{-j\omega k}$$

$$\phi(\omega) = \lim_{N \to \infty} E \left\{ \frac{1}{N} \left| \sum_{n=1}^{N} y(n) e^{-j\omega n} \right|^{2} \right\}$$

Exercise 1 (2/4)

■ Hint: Remember that AutoCorrelation Sequence is: $r(k) = E[y(n) \ y*(n-k)]$

• Properties: r(k) = r*(-k)

$$r(0) \ge |r(k)|$$

Hint: Remember that:

$$\begin{array}{c|c} e(t) & y(t) = e(t) * h(t) \\ \hline \phi_e(\omega) & \phi_y(\omega) = |H(\omega)|^2 \phi_e(\omega) \end{array}$$

$$r_{y}(k) = \sum_{p=-\infty}^{\infty} \sum_{m=-\infty}^{\infty} h_{p} h_{m}^{*} r_{e} (m+k-p)$$



Exercise 1 (3/4)

- Goal: Estimate the power spectral density of the signal "flute2" by means of <u>periodogram</u>.
- Hints on periodogram: the spectrum estimation using periodogram is given by the following equation:

$$\hat{\phi}_P(\omega) = \frac{1}{N} \left| \sum_{n=1}^N y(n) e^{-j\omega n} \right|^2$$

Exercise 1 (4/4)

- Pseudocode:
 - load the file flute2.way
 - consider 50ms of the input signal (y)
 - estimate PSD using parallelogram:
 - Arr N = length(y);
 - M = 2^ceil(log2(N)+1);%number of frequency bins
 - phip = $(1/N)*abs(fft(y,M)).^2$;



Exercise 1b (1/3)

- Goal: quantify the bias and variance of the periodogram
- N.B.: Periodogram is asymptotically unbiased and has large variance, even for large N.

$$E\left[\hat{\phi}_{P}(\omega)\right] = \sum_{k=-\infty}^{\infty} w_{B}(k) r(k) e^{-j\omega k} = \frac{1}{2\pi} \int_{-\pi}^{\pi} \phi(\xi) W_{B}(\omega - \xi) d\xi$$

$$\lim_{N \to \infty} \operatorname{var} \left\{ \hat{\phi}_{P}(\omega) \right\} = E \left\{ \left[\hat{\phi}_{P}(\omega_{1}) - \phi(\omega_{1}) \right] \right] \left[\hat{\phi}_{P}(\omega_{2}) - \phi(\omega_{2}) \right]$$

$$= \begin{cases} \phi^{2}(\omega_{1}) & \omega_{1} = \omega_{2} \\ 0 & \omega_{1} = \omega_{2} \end{cases}$$



Exercise 1b (2/3)

- Goal: quantify the bias and variance of the periodogram
- Pseudocode:
 - compute R realizations of N samples white noise
 - \bullet e = randn(N,R);
 - for each realization:
 - filter white noise by means of a LTI filter

$$Y(z) = H(z)E(z)$$

- compute the periodogram spectral estimate phip(i,:) = (1/N)*abs(fft(y,N)).^2;
- end



Exercise 1b (3/3)

- compute the ensemble mean: phip(RxN)
 - phipmean = mean(phip);
- compute the ensemble variance
 - phipvar = var(phip);
- Plot the true spectrum $\Phi(\omega)$
- Plot the average of R spectral estimates and compare the averaged spectrum and the true PSD
- Plot $E[\hat{\phi}_P(\omega)] \pm \sqrt{\text{var}[\hat{\phi}_P(\omega)]}$ and compare with the true PSD



Exercise 1c (1/4)

- Goal: quantify the bias and variance of the periodogram
- Procedure: filter a white noise by means of the following LTI filter

$$Y(z) = [H_1(z) + H_2(z)]E(Z)$$

With the following poles and zeros

$$p_{1} = \rho_{P}e^{j\pm\pi\omega_{1}}$$
 $p_{2} = \rho_{P}e^{j\pm\pi\omega_{2}}$ $\rho_{Z} = 0.95$ $\rho_{P} = 0.99$ $z_{1} = \rho_{Z}e^{j\pm\pi\omega_{1}}$ $z_{2} = \rho_{Z}e^{j\pm\pi\omega_{2}}$ $\omega_{1} = 0.6\pi$ $\omega_{2} = 2(0.3 + \alpha)\pi$ $\omega_{2} = 0.05$ 10



Exercise 1c (2/4)

Hint

$$H_{i}(z) = \frac{B_{i}(z)}{A_{i}(z)} = \frac{(1 - \rho_{z}e^{+j\omega_{i}}z^{-1})(1 - \rho_{z}e^{-j\omega_{i}}z^{-1})}{(1 - \rho_{p}e^{+j\omega_{i}}z^{-1})(1 - \rho_{p}e^{-j\omega_{i}}z^{-1})} = \frac{1 - 2\rho_{z}\cos(\omega_{1})z^{-1} + \rho_{z}^{2}z^{-2}}{1 - 2\rho_{p}\cos(\omega_{1})z^{-1} + \rho_{p}^{2}z^{-2}}$$

$$H(z) = H_1(z) + H_2(z) = \frac{B_1(z)}{A_1(z)} + \frac{B_2(z)}{A_2(z)} = \frac{B_1(z)A_2(z) + B_2(z)A_1(z)}{A_1(z)A_2(z)}$$

$$C(z) = A(z)B(z) \Rightarrow \mathbf{c} = conv(\mathbf{a}, \mathbf{b})$$

$$C(z) = A(z) + B(z) \Rightarrow \mathbf{c} = \mathbf{a} + \mathbf{b}$$



Exercise 1c (3/4)

- Goal: quantify the bias and variance of the periodogram
- Pseudocode:
 - compute R realizations of N samples white noise
 - \bullet e = randn(N,R);
 - for each realization:
 - filter white noise by means of a LTI filter

$$Y(z) = H(z)E(z)$$

- compute the periodogram spectral estimate phip(i,:) = (1/N)*abs(fft(y,N)).^2;
- end



Exercise 1c (4/4)

- compute the ensemble mean: phip(RxN)
 - phipmean = mean(phip);
- compute the ensemble variance
 - phipvar = var(phip);
- Plot the true spectrum $\Phi(\omega)$
- Plot the average of R spectral estimates and compare the averaged spectrum and the true PSD
- Plot $E[\hat{\phi}_P(\omega)] \pm \sqrt{\text{var}[\hat{\phi}_P(\omega)]}$ and compare with the true PSD



Exercise 2 (1/3)

- Goal: Estimate the power spectral density of the signal "flute2" by means of <u>correlogram</u>.
- Hints on correlogram: the spectrum estimation using correlogram is given by the following equation:

$$\hat{\phi}_C(\omega) = \sum_{k=-(N-1)}^{N-1} \hat{r}(k) e^{-j\omega k}$$

$$\hat{r}(k) = \frac{1}{N} \sum_{n=k+1}^{N} y(n) \ y * (n-k) \quad k \ge 0$$

Exercise 2 (2/3)

- Pseudocode:
 - load the file flute2.wav
 - consider 50ms of the input signal (y)
 - estimate ACS
 - [r lags] = xcorr(y, 'biased');
 - r = circshift(r,N);
 - estimate PSD using correlogram:
 - \bullet N = length(y);
 - $M = 2^cil(log2(2*N-1)+1);$

%number of frequency bins

phic = fft(r,M);



Exercise 2 (3/3)

- Hint: Matlab provides the functions:
 - "[r lag]=xcorr(x,'biased')" that produces a biased estimate of the autocorrelation (2N-1 samples) of the stationary sequence "x". "lag" is the vector of lag indices [-N+1:1:N-1].
 - "r = circshift(r,N)" that circularly shifts the values in the array r by N elements. If N is positive, the values of r are shifted down (or to the right). If it is negative, the values of r are shifted up (or to the left).



Exercise 3 (1/2)

- Goal: Estimate the power spectral density of the signal "flute2" by means of <u>Blackman-</u> <u>Tukey method</u>.
- Hints on B-T method: the spectrum estimation using BT method is given by the following equation:

$$\hat{\phi}_{BT}(\omega) = \sum_{k=-(M-1)}^{M-1} w(k)\hat{r}(k) e^{-j\omega k}$$

$$W(\omega) \ge 0$$

Exercise 3 (2/2)

- Pseudocode:
 - load the file flute2.wav
 - consider 50ms of the input signal -->N = length(y);
 - estimate ACS
 - [r lags] = xcorr(y, 'biased');
 - window with a bartlett window of the same length
 - rw = r.*bartlett(2*N-1);
 - r = circshift(r,N);
 - estimate PSD using BT:
 - Nfft = $2^{\text{ceil}(\log 2(2*N-1)+1)}$;
 - phiBT = real(fft(r,Nfft));



Exercise 3b (1/3)

Goal: quantify the bias and variance of the BT method

Pseudocode:

- compute R realizations of N samples white noise
 - \bullet e = randn(N,R);
- for each realization:
 - filter white noise by means of a LTI filter

$$Y(z) = H(z)E(z)$$

- compute the BT spectral estimate
- end



Exercise 3b (2/3)

- compute the ensemble mean: phip(RxN)
 - phipmean = mean(phip);
- compute the ensemble variance
 - phipvar = var(phip);
- Plot the true spectrum $\Phi(\omega)$
- Plot the average of R spectral estimates and compare the averaged spectrum and the true PSD
- Plot $E[\hat{\phi}_P(\omega)] \pm \sqrt{\text{var}[\hat{\phi}_P(\omega)]}$ and compare with the true PSD

Exercise 3b (3/3)

• QUESTION: Can you see any difference between periodogram and BT method?



Exercise 4 (1/3)

- Goal: Estimate the power spectral density of the signal "flute2" by means of <u>Bartlett</u> <u>method</u>.
- Basic idea: split up the available sample of N observations into L = N/M subsamples of M observations each, then average the periodograms obtained from the subsamples for each value of ω.



Exercise 4 (2/3)

• Matematically: the spectrum estimation using Bartlett method is given by the following equations:

$$y_i(n) = y((i-1)M + n)$$
 $n = 1,..., M$
$$i = 1,..., L \equiv \frac{N}{M}$$

$$\hat{\phi}_i(\omega) = \frac{1}{M} \left| \sum_{n=1}^M y_i(n) e^{-j\omega n} \right|^2 \qquad \hat{\phi}_B(\omega) = \frac{1}{L} \sum_{i=1}^L \hat{\phi}_i(\omega)$$

Exercise 4 (3/3)

Pseudocode:

- load the file flute2.wav
- consider 50ms of the input signal -->N = length(y);
- define the number of subsequences L and the number of samples for each of them M=ceil(N/L)
- for each subsequence:
 - consider the right samples: yl = y(1+I*M : M+I*M);
 - estimate periodogram: (1/M)*abs(fft(yl)).^2
- mean periodograms of the subsequences:
 - phil = phil + (1/M)*abs(fft(yl)).^2;
 - phiB=phil/L;



Exercise 5 (1/4)

Goal: Estimate the power spectral density of the signal "flute2" by means of <u>Welch</u> <u>method</u>.

- Basic idea: similar to Bartlett method but:
- allow overlap of subsequences
- use data window for each periodogram



Exercise 5 (2/4)

Matematically: the spectrum estimation using Welch method is given by the following equations:

$$y_i(n) = y((i-1)K + n)$$
 $n = 1,..., M$
 $i = 1,..., S$

$$\hat{\phi}_i(\omega) = \frac{1}{MP} \left| \sum_{n=1}^{M} v(n) y_i(n) e^{-j\omega n} \right|^2$$

$$\hat{\phi}_{W}(\omega) = \frac{1}{S} \sum_{i=1}^{S} \hat{\phi}_{i}(\omega)$$

$$P = \frac{1}{M} \sum_{n=1}^{M} \left| v(n) \right|^2$$

Exercise 5 (3/4)

- Pseudocode:
 - load the file flute2.way
 - consider 50ms of the input signal -->N = length(y);
 - define:
 - the number of samples for each subsequence: M
 - the number of new samples for each subsequence: K=M/4
 - the number of subsequences: S= N/K (M-K)/K;
 - the window: v = hamming(M);
 - $P = (1/M)*sum(v.^2);$

-

Exercise 5 (4/4)

- Pseudocode (continued):
 - for each subsequence:
 - consider the right samples:
 - xs = x(1+s*K : M+s*K);
 - window the subsequence:
 - V.*XS
 - estimate periodogram:
 - (1/(M*P))*abs(fft(v.*xs)).^2
 - mean periodograms of the subsequences:
 - phis = phis + $(1/(M*P))*abs(fft(v.*xs)).^2$;
 - phiW = phis/S;



NON PARAMETRIC SPECTRAL ESTIMATION

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PARAMETRIC SPECTRAL ESTIMATION

Ex6: AutoRegressive model



Exercise 6 (1/11)

The parametric of model-based methods of spectral estimation assume that the signal satisfies a generating model with known functional form, and then proceed in estimating the parameters in the assumed model Power Spectral Density

$$H(z) = \frac{B(z)}{A(z)}$$

$$\phi(\omega) = \left| \frac{B(\omega)}{A(\omega)} \right|^2 \sigma_e^2$$

Exercise 6 (2/11)

Hint: We have y(n).

From the estimated $\{a_n\}$ and $\{b_m\}$, we can compute the correspondent $\hat{\phi}(\omega)$:

- Depending on the values assumed by m and n we can have the following cases:
 - if m = 0 and $n \ne 0$, autoregressive model (AR), A(z)Y(z) = E(z)
 - if $m \ne 0$ and n = 0, moving average model (MA), Y(z) = B(z)E(z)
 - if m ≠ 0 and n ≠ 0, ARMA model (autoregressive, moving average), A(z)Y(z) = B(z)E(z)



Exercise 6 (3/11)

For AR class we have:

$$\begin{cases} r(0) + \sum_{i=1}^{n} a_i r(-i) = \sigma_e^2 \sum_{j=1}^{m} b_j h_j^* = \sigma_e^2 \\ r(k) + \sum_{i=1}^{n} a_i r(k-i) = 0 \end{cases}$$
 $k > 0$



Exercise 6 (4/11)

For AR class we have:

$$\begin{bmatrix} r(0) & r(-1) & \dots & r(-n) \\ r(1) & r(0) & \dots & \cdot \\ & \cdot & \cdot & \cdot \\ r(n) & \cdot & \dots & r(0) \end{bmatrix} \begin{bmatrix} 1 \\ a_1 \\ \vdots \\ a_n \end{bmatrix} = \begin{bmatrix} \sigma^2 \\ 0 \\ 0 \\ 0 \end{bmatrix}$$



Exercise 6 (5/11)

For AR class we have:

$$\begin{bmatrix} r(1) \\ . \\ r(n) \end{bmatrix} + \begin{bmatrix} r(0) & . & r(-n+1) \\ . & . & . \\ r(n-1) & . & r(0) \end{bmatrix} \begin{bmatrix} a_1 \\ . \\ a_n \end{bmatrix} = \begin{bmatrix} 0 \\ 0 \\ 0 \end{bmatrix}$$

$$r_n + R_n \theta = 0 \qquad \theta = [a_1, a_2, ..., a_n]$$



Exercise 6 (6/11)

Solution:

$$\theta = -R_n^{-1} r_n \qquad \theta = [a_1, a_2, ..., a_n]$$

$$\sigma^2 = r(0) + \sum_{i=1}^n a_i r(-i)$$

• Hint: if the true ACS is unknown, we can replace r(k) with $\hat{r}(k)$

• The PSD estimate is:
$$\hat{\phi}(\omega) = \frac{\hat{\sigma}^2}{\left|\hat{A}(\omega)\right|^2}$$



Exercise 6 (7/11)

 Goal: Estimate the power spectral density of the signal "y" by means of AR model.

Pseudocode:

Consider the signal y defined by the differential equation:

$$y(n)=a_1 y(n-1) + a_2 y(n-2) + a_3 y(n-3) + z(n)$$

- Estimate $\{a_p\}$ and σ_z with an AR model (order p)
- Plot estimated PSD and compare with the true PSD

Exercise 6 (8/11)

- Hint: Matlab provides the functions:
 - "[r lag]=xcorr(x,'biased')" that produces a biased estimate of the autocorrelation (2N-1 samples) of the stationary sequence "x". "lag" is the vector of lag indices [-N+1:1:N-1].
 - "R=toeplitz(C,R)" that produces a nonsymmetric Toeplitz matrix having C as its first column and R as its first row.
 - "R=toeplitz(R)" is a symmetric (or Hermitian) Toeplitz matrix.

Exercise 6 (9/11)

Pseudocode:

Consider the signal y defined by the differential equation:

$$y(n)=a_1 y(n-1) + a_2 y(n-2) + a_3 y(n-3) + z(n)$$

- sigmae = 10;
- $a = poly([0.99 \ 0.99*exp(j*pi/4) \ 0.99*exp(-j*pi/4)])$
- b = 1;
- z = sigmae*randn(N,1);
- y = filter(b, a, z);

Exercise 6 (10/11)

Pseudocode (continued):

- Estimate $\{a_p\}$ and σ_z with an AR model (order n)
 - n=3;
 - r = xcorr(y , 'biased');
 - Rx = toeplitz(r(N:N+n-1), r(N:-1:N-n+1));
 - rz = r(N+1:N+n);
 - theta = $-Rx^{(-1)*rz}$;
 - varz = r(N) + sum(theta.*r(N-1:-1:N-n));



Exercise 6 (11/11)

- Pseudocode (continued):
 - Plot estimated PSD and compare with the true PSD
 - plot(w, 10*log10(sigmae^2*abs(H).^2))
 - hold on
 - [He, w] = freqz(1,[1; theta], Nfft);
 - plot(w, 10*log10(varz*abs(He).^2), 'r')
 - legend ('true', 'estimated');
 - hold off



Exercise 6b (1/1)

An all-pole LTI filter is defined by the following poles:

[0.99, 0.9e+j0.2, 0.9e-j0.2, 0.95e+j0.5, 0.95e-j0.5]

- Feed a white noise process e(n) (re(n) = δ (n)) into the system to produce one realization of N = 1000 samples.
- Plot the true PSD
- Estimate the PSD using the Bartlett method.
- Estimate the PSD using a parametric method.
 Verify the effect of the number of parameters.



 Goal: Estimate the PSD, using AR model, of the signal obtained filtering a white noise by means of the following LTI filter

$$Y(z) = [H_1(z) + H_2(z)]E(Z)$$

With the following poles and zeros

 $\rho_{z} = 0.95$

$$H(z) = H_1(z) + H_2(z) = \frac{B_1(z)}{A_1(z)} + \frac{B_2(z)}{A_2(z)} = \frac{B_1(z)A_2(z) + B_2(z)A_1(z)}{A_1(z)A_2(z)}$$

Exercise 6c (2/3)

Pseudocode:

- compute R realizations of N samples white noise
 - \bullet e = randn(N,R);
- for each realization:
 - filter white noise by means of a LTI filter

$$Y(z) = H(z)E(z)$$

- compute the AR spectral estimate
- [P, w] = freqz(1, [1 theta'], w);
- Shat(r,:)=var_s*abs(P).^2;
- end



Exercise 6c (3/3)

- compute the ensemble mean: Shat(RxN)
 - ExpShat=mean(Shat,1);
- Plot the true spectrum $\Phi(\omega)$
- Plot the average of R spectral estimates and compare the averaged spectrum and the true PSD
- Observe what happens varying the order p