Part 3. Spectrum Estimation3.1 Classic Methods for Spectrum Estimation

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<u>Logistics</u>

- Last Lecture: lattice predictor
 - correlation properties of error processes
 - joint process estimator in lattice
 - inverse lattice filter structure
- Today:
 - Spectrum estimation: background and classical methods

Homework set

Summary of Related Readings on Part-II

2.1 Stochastic Processes and modeling

Haykin (4th Ed) 1.1-1.8, 1.12-1.14 Hayes 3.3 – 3.7 (3.5); 4.7

2.2 Wiener filtering

Haykin (4th Ed) Chapter 2 Hayes 7.1, 7.2, 7.3.1

2.3-2.4 Linear prediction and Levinson-Durbin recursion

Haykin (4th Ed) 3.1 – 3.3 Hayes 7.2.2; 5.1; 5.2.1 – 5.2.2, 5.2.4 – 5.2.5

2.5 Lattice predictor

Haykin (4th Ed) 3.8 – 3.10 Hayes 6.2; 7.2.4; 6.4.1

Summary of Related Readings on Part-III

Overview Haykins 1.16, 1.10

3.1 Non-parametric method

Hayes 8.1; 8.2 (8.2.3, 8.2.5); 8.3

3.2 Parametric method

Hayes 8.5, 4.7; 8.4

3.3 Frequency estimation

Hayes 8.6

Review

- On DSP and Linear algebra: Hayes 2.2, 2.3
- On probability and parameter estimation: Hayes 3.1 3.2

Spectrum Estimation: Background

- Spectral estimation: determine the power distribution in frequency of a random process
 - E.g "Does most of the power of a signal reside at low or high frequencies?" "Are there resonances in the spectrum?"
- Applications:
 - Needs of spectral knowledge in spectrum domain non-causal
 Wiener filtering, signal detection and tracking, beamforming, etc.
 - Wide use in diverse fields: radar, sonar, speech, biomedicine, geophysics, economics, ...
- Estimating p.s.d. of a w.s.s. process is equivalent to estimate autocorrelation at all lags

Spectral Estimation: Challenges

- When a limited amount of observation data are available
 - Can't get r(k) for all k and/or may have inaccurate estimate of r(k)
 - Scenario-1: transient measurement (earthquake, volcano, ...)
 - Scenario-2: constrained to short period to ensure (approx.) stationarity in speech processing

$$\hat{r}(k) = \frac{1}{N-k} \sum_{n=k+1}^{N} u[n] u^*[n-k], \ k = 0, 1, \dots M$$

• Observed data may have been corrupted by noise

Spectral Estimation: Major Approaches

- No assumptions on the underlying model for the data
- Periodogram and its variations (averaging, smoothing, ...)
- Minimum variance method
- ARMA, AR, MA models
- Maximum entropy method
- For harmonic processes that consist of a sum of sinusoids or complex-exponentials in noise

Spectral Estimation: Major Approaches

Nonparametric methods

- No assumptions on the underlying model for the data
- Periodogram and its variations (averaging, smoothing, ...)
- Minimum variance method
- Parametric methods
 - ARMA, AR, MA models
 - Maximum entropy method
- Frequency estimation (noise subspace methods)
 - For harmonic processes that consist of a sum of sinusoids or complex-exponentials in noise
- High-order statistics

Example of Speech Spectrogram



Figure 3 of SPM May'98 Speech Survey



Section 3.1 Classical Nonparametric Methods

Recall: given a w.s.s. process {x[n]} with

$$\begin{cases} E[x[n]] = m_x \\ E[x^*[n]x[n+k]] = r(k) \end{cases}$$

The power spectral density (p.s.d.) is defined as

As we can take DTFT on a specific realization of a random process, What is the relation between the DTFT of a specific signal and the p.s.d. of the random process?

 $-\frac{1}{2} \le f \le \frac{1}{2}$

(or $\omega = 2\pi f : -\pi \le \omega \le \pi$)

Section 3.1 Classical Nonparametric Methods

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$$\begin{cases} E[x[n]] = m_x \\ E[x^*[n]x[n+k]] = r(k) \end{cases}$$

The power spectral density (p.s.d.) is defined as

$$P(f) = \sum_{k=-\infty}^{\infty} r(k)e^{-j2\pi fk} \qquad -\frac{1}{2} \le f \le \frac{1}{2}$$

(or $\omega = 2\pi f : -\pi \le \omega \le \pi$)

As we can take DTFT on a specific realization of a random process, What is the relation between the DTFT of a specific signal and the p.s.d. of the random process?

\bigwedge

Ensemble Average of Squared Fourier Magnitude

 p.s.d. can be related to the ensemble average of the squared Fourier magnitude |X(ω)|²

Consider
$$P_M(f) \stackrel{\Delta}{=} \frac{1}{2M+1} \left| \sum_{n=-M}^{M} x[n] e^{-j2\pi f n} \right|^2$$

= $\frac{1}{2M+1} \sum_{n=-M}^{M} \sum_{m=-M}^{M} x[n] x^*[m] e^{-j2\pi f (n-m)}$

Ensemble Average of Squared Fourier Magnitude

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= $\frac{1}{2M+1} \sum_{n=-M}^{M} \sum_{m=-M}^{M} x[n] x^*[m] e^{-j2\pi f (n-m)}$

i.e., take DTFT on (2M+1) samples and examine normalized magnitude

Note: for each frequency f, $P_M(f)$ is a random variable

Ensemble Average of P_M(f)

$$E[P_{M}(f)] = \frac{1}{2M+1} \sum_{n=-M}^{M} \sum_{m=-M}^{M} r(n-m)e^{-j2\pi f(n-m)}$$
$$= \frac{1}{2M+1} \sum_{k=-2M}^{2M} (2M+1-|k|)r(k)e^{-j2\pi fk}$$

• Now, what if M goes to infinity?

Ensemble Average of P_M(f)

$$\begin{split} E[P_{M}(f)] &= \frac{1}{2M+1} \sum_{n=-M}^{M} \sum_{m=-M}^{M} r(n-m) e^{-j2\pi f(n-m)} \\ &= \frac{1}{2M+1} \sum_{k=-2M}^{2M} (2M+1-|k|) r(k) e^{-j2\pi fk} \\ &= \sum_{k=-2M}^{2M} \left(1 - \frac{|k|}{2M+1} \right) r(k) e^{-j2\pi fk} \\ &= \sum_{k=-2M}^{2M} r(k) e^{-j2\pi fk} - \frac{1}{2M+1} \sum_{k=-2M}^{2M} |k| r(k) e^{-j2\pi fk} \end{split}$$

Now, what if M goes to infinity?

P.S.D. and Ensemble Fourier Magnitude

If the autocorrelation function decays fast enough s.t.



P.S.D. and Ensemble Fourier Magnitude

If the autocorrelation function decays fast enough s.t.

$$\sum_{k=-\infty}^{\infty} |k| r(k) < \infty \quad (i.e., r(k) \to 0 \text{ rapidly for } k \uparrow)$$

then
$$\lim_{M \to \infty} E[P_M(f)] = \sum_{k=-\infty}^{\infty} r(k) e^{-j2\pi f k} = P(f)$$

p.s.d.
Thus
$$P(f) = \lim_{M \to \infty} E\left[\frac{1}{2M+1} \left|\sum_{n=-M}^{M} x[n] e^{-j2\pi f n}\right|^2\right] \quad (**)$$

3.1.1 Periodogram Spectral Estimator

(1) This estimator is based on (**)

Given an observed data set {x[0], x[1], ..., x[N-1]}, the periodogram is defined as

$$\hat{P}_{\text{PER}}(f) = \frac{1}{N} \left| \sum_{n=0}^{N-1} x[n] e^{-j2\pi f n} \right|^2$$

3.1.1 Periodogram Spectral Estimator

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Given an observed data set {x[0], x[1], ..., x[N-1]}, the periodogram is defined as

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$$X[n] \xrightarrow{} X_N[n] \xrightarrow{} X_N[n] \xrightarrow{} X_N(K) \xrightarrow{}$$

An Equivalent Expression of Periodogram

The periodogram estimator can be given in terms of r(k)

$$\hat{P}_{\text{PER}}(f) = \sum_{k=-(N-1)}^{N-1} \hat{r}(k) e^{-j2\pi fk}$$

where
$$r(k)$$

 \wedge

 The quality of the estimates for the higher lags of r(k) may be poorer since they involve fewer terms of lag products in the averaging operation

Exercise: to show this from the periodogram definition in last page

An Equivalent Expression of Periodogram

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$$\hat{P}_{\text{PER}}(f) = \sum_{k=-(N-1)}^{N-1} \hat{r}(k) e^{-j2\pi fk}$$

where
$$\hat{r}(k) = \frac{1}{N} \sum_{n=0}^{N-1-k} x^*[n] x[n+k]; \hat{r}(-k) = \hat{r}^*(k) \text{ for } k \ge 0$$

 The quality of the estimates for the higher lags of r(k) may be poorer since they involve fewer terms of lag products in the averaging operation

Exercise: to show this from the periodogram definition in last page

Nonparametric spectral estimation [22]

(2) Filter Bank Interpretation of Periodogram

For a particular frequency of f_{0} .

$$\hat{P}_{\text{PER}}(f_{0}) = \frac{1}{N} \left| \sum_{k=0}^{N-1} e^{-j2\pi f_{0}k} x[k] \right|^{2}$$
$$= \left[N \cdot \left| \sum_{k=0}^{N-1} h[n-k] x[k] \right|^{2} \right]_{n=0}$$
where
$$h[n] =$$

Impulse response of the filter h[n]: a windowed version of a complex exponential

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$$\hat{P}_{\text{PER}}(f_0) = \frac{1}{N} \left| \sum_{k=0}^{N-1} e^{-j2\pi f_0 k} x[k] \right|^2$$
$$= \left[N \cdot \left| \sum_{k=0}^{N-1} h[n-k] x[k] \right|^2 \right]_{n=0}$$

, $h[n] = \begin{cases} \frac{1}{N} \exp(j2\pi f_0 n) & \text{for } n = -(N-1), ..., -1, 0; \\ 0 & \text{otherwise} \end{cases}$

- Impulse response of the filter h[n]: a windowed version of a complex exponential

Frequency Response of h[n]

$$H(f) = \frac{\sin N\pi (f - f_0)}{N \sin \pi (f - f_0)} \exp[j(N - 1)\pi (f - f_0)]$$

sinc-like function centered at f_{0:}



Frequency Response of h[n]

$$H(f) = \frac{\sin N\pi (f - f_0)}{N\sin \pi (f - f_0)} \exp[j(N - 1)\pi (f - f_0)]$$

sinc-like function centered at $f_{0:}$

- H(f) is a bandpass filter
 - Center frequency is f_0
 - 3dB bandwidth $\approx 1/N$



Periodogram: Filter Bank Perspective

- Can view the periodogram as an estimator of power spetrum that has a built-in filterbank
 - The filter bank ~ a set of bandpass filters

$$\hat{P}_{\text{PER}}(f_0) = \left[N \cdot \left| \sum_{k=0}^{N-1} h[n-k] x[k] \right|^2 \right]_{n=0}$$

Periodogram: Filter Bank Perspective

- Can view the periodogram as an estimator of power spetrum that has a built-in filterbank
 - The filter bank ~ a set of bandpass filters
 - The estimated p.s.d. for each frequency f_0 is the power of one output sample of the bandpass filter centering at f_0

$$\hat{P}_{\text{PER}}(f_0) = \left[N \cdot \left| \sum_{k=0}^{N-1} h[n-k] x[k] \right|^2 \right]_{n=0}$$

E.g. White Gaussian Process

[Lim/Oppenheim Fig.2.4] Periodogram of zero-mean white Gaussian noise using N-point data record: N=128, 256, 512, 1024



The random fluctuation (measured by variance) of the periodogram does not decrease with increasing N
 periodogram is not a consistent estimator

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(3) How Good is Periodogram for Spectral Estimation?

If
$$N \to \infty$$
, will $\stackrel{\wedge}{P}_{\text{PER}} \to \text{p.s.d.} P(f)$?

Estimation: Tradeoff between bias and variance

$$E(\hat{\theta}) \neq \theta$$
$$E[|\hat{\theta} - E(\hat{\theta})|^{2}] = ?$$

• For white Gaussian process, we can show that at $f_k = k/N$

$$\Rightarrow E[\hat{P}_{PER}(f\kappa)] = P(f\kappa), \ \kappa = 0, 1, \dots, \frac{N}{2}$$

$$Var[\hat{P}_{PER}(f\kappa)] = \begin{cases} P^{2}(f\kappa), \ \kappa = 1, \dots, \frac{N}{2} - 1 \\ 2P^{2}(f\kappa), \ \kappa = 0, \frac{N}{2} \end{cases} \propto P^{2}(f\kappa)$$

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Performance of Periodogram: Summary

- The periodogram for white Gaussian process is an <u>unbiased</u> estimator but not <u>consistent</u>
 - The variance does not decrease with increasing data length
 - Its standard deviation is as large as the mean (equal to the quantity to be estimated)
- Reasons for the poor estimation performance
 - Given N real data points, the # of unknown parameters {P(f_0), ... P($f_{N/2}$)} we try to estimate is N/2, i.e. proportional to N
- Similar conclusions can be drawn for processes with arbitrary p.s.d. and arbitrary frequencies
 - Asymptotically unbiased (as N goes to infinity) but inconsistent

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3.1.2 Averaged Periodogram

- As one solution to the variance problem of periodogram
 - Average K periodograms computed from K sets of data records

$$\hat{P}_{AV PER}(f) = \frac{1}{K} \sum_{m=0}^{K-1} \hat{P}_{PER}^{(m)}(f)$$
where $\hat{P}_{PER}^{(m)}(f) = \frac{1}{L} \left| \sum_{n=0}^{L-1} x_m[n] e^{-j2\pi f n} \right|^2$

And the K sets of data records are $\{x_0[0], ..., x_0[L-1]; x_1[n], 0 \le n \le L-1; ... \ \{x_{K-1}[n-1], 0 \le n \le L-1\}$

Performance of Averaged Periodogram

- If K sets of data records are uncorrelated with each other, we have: $(f_i = i/L)$

 $P_{PER}^{(M)}(f)$ i.i.d. (m=0,1, ... L-1) for white Gaussian process

$$\propto \frac{1}{K} P^2(f_i)$$

Performance of Averaged Periodogram

- If K sets of data records are uncorrelated with each other, we have : $(f_i = i/L)$

 $\widehat{P}_{PER}^{(M)}(f) \text{ i.i.d. (m=0,1, ... L-1) for white Gaussian process}$ $\Longrightarrow \quad \forall \text{ar-} [\widehat{P}_{AVPER}^{}(f)] = \qquad \qquad \propto \frac{1}{K} P^{2}(f_{i})$ $\begin{cases} \frac{1}{K} P^{2}(f_{i}) & i = 1, 2, \dots, \frac{L}{2} - 1 \\ \frac{2}{K} P^{2}(f_{i}) & i = 0, \frac{L}{2} \end{cases}$ i.e., $K \uparrow \rightarrow \text{Var} \downarrow$, and $\text{Var} \rightarrow 0$ for $K \rightarrow \infty$

i.e., consistent estimate

Practical Averaged Periodogram

- Usually we partition an available data sequence of length N – into K non-overlapping blocks, each block has length L (i.e. N=KL) i.e. r [n] = r[n + mL]
 - $x_m[n] = x[n+mL], \qquad n = 0, 1, ..., L-1$ m = 0, 1, ..., K-1
- Since the blocks are contiguous, the K sets of data records may not be completely uncorrelated
 - Thus the variance reduction factor is in general less than K
- Periodogram averaging is also known as the Bartlett's method

Averaged Periodogram for Fixed Data Size

 Given a data record of fixed size N, will the result be better if we segment the data into more and more subrecords?

We examine for a <u>real-valued</u> stationary process:

 $\hat{P}_{\text{PER}}^{(0)}(f) = \sum_{l=1}^{L-1} \hat{r}^{(0)}(l) e^{-j2\pi f l}$

l = -(L - 1)

 $\hat{r}^{(0)}(l) = \frac{1}{L} \sum_{n=1}^{L-1-|l|} x[n]x[n+|l|]$

$$E\begin{bmatrix} \stackrel{\wedge}{P}_{\text{AV PER}}(f) \end{bmatrix} = E\begin{bmatrix} \frac{1}{K} \sum_{m=0}^{K-1} \stackrel{\wedge}{P}_{\text{PER}}(f) \end{bmatrix} = E\begin{bmatrix} \hat{P}_{\text{PER}}^{(0)}(f) \end{bmatrix}$$

identical distribution for all m

Note

where

an equivalent expression to definition in terms of x[n]

Mean of Averaged Periodogram



Mean of Averaged Periodogram

$$\Rightarrow E[\hat{\Gamma}^{(0)}(l)] = (I - \frac{|l|}{L}) M(l) \text{ for } |l| \leq L - I$$

$$\triangleq W(l)$$

$$\stackrel{\cdot}{=} E[\hat{P}_{AVPER}(f)] = \sum_{l=-(L-I)}^{L-1} W(l) \Gamma(l) e^{-j2\pi f l}$$

$$W[K] = \begin{cases} 1 - \frac{|K|}{L} \text{ for } |K| \leq L-1 & W(f) \\ 0 & 0 \cdot W. & \text{triangular} & \text{3dB b.W.} \\ (Barlett) & \rightarrow & K \approx \frac{|K|}{L} \\ \Rightarrow W(f) = \frac{1}{L} \left(\frac{Sin TI f L}{Sin TI f} \right)^{\frac{1}{2}} & \text{window}^{\frac{1}{2}} \\ & \text{window}^{\frac{1}{2}} & 0 & f \end{cases}$$

Mean of Averaged Periodogram (cont'd)

$$\begin{split} E[\hat{P}_{\text{AV PER}}(f)] &= \text{DTFT}[\{w[k]r(k)\}]_{f} & \text{multiplication in time} \\ &= \int_{-\frac{1}{2}}^{\frac{1}{2}} W(f - \eta) P(\eta) d\eta & \text{convolution in frequency} \\ &\neq P(f) \end{split}$$

- Biased estimate (both averaged and regular periodogram)
 - The convolution with the window function w[k] lead to the mean of the averaged periodogram being smeared from the true p.s.d

Mean of Averaged Periodogram (cont'd)

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- Biased estimate (both averaged and regular periodogram)
 - The convolution with the window function w[k] lead to the mean of the averaged periodogram being smeared from the true p.s.d
- Asymptotic unbiased as $L \rightarrow \infty$
 - To avoid the smearing, the window length L must be large enough so that the narrowest peak in P(f) can be resolved
- This gives a tradeoff between bias and variance Small K => better resolution (smaller smearing/bias) but larger variance

Non-parametric Spectrum Estimation: Recap

• Periodogram

- Motivated by relation between p.s.d. and squared magnitude of DTFT of a finite-size data record
- Variance: won't vanish as data length N goes infinity ~ "inconsistent"
- Mean: asymptotically unbiased w.r.t. data length N in general
 - equivalent to apply triangular window to autocorrelation function (windowing in time gives smearing/smoothing in freq domain)
 - unbiased for white Gaussian
- Averaged periodogram
 - Reduce variance by averaging K sets of data record of length L each
 - Small L increases smearing/smoothing in p.s.d. estimate thus higher bias → equiv. to triangular windowing
- Windowed periodogram: generalize to other symmetric windows

Case Study on Non-parametric Methods

• Test case: a process consists of narrowband components (sinusoids) and a broadband component (AR)

 $- x[n] = 2 \cos(\omega_1 n) + 2 \cos(\omega_2 n) + 2 \cos(\omega_3 n) + z[n]$ where $z[n] = -a_1 z[n-1] + v[n], a_1 = -0.85, \sigma^2 = 0.1$ $\omega_1/2\pi = 0.05, \omega_2/2\pi = 0.40, \omega_3/2\pi = 0.42$

- N=32 data points are available \rightarrow periodogram resolution f = 1/32
- Examine typical characteristics of various non-parametric spectral estimators

(Fig.2.17 from Lim/Oppenheim book)





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Nonparametric spectral estimation [43]

3.1.3 Periodogram with Windowing

Review and Motivation

The periodogram estimator can be given in terms of $\stackrel{\wedge}{r(k)}$ $\stackrel{\wedge}{P}_{\text{PER}}(f) = \sum_{k=-(N-1)}^{N-1} \stackrel{\wedge}{r(k)} e^{-j2\pi f k}$ where $\stackrel{\wedge}{r(k)} = \frac{1}{N} \sum_{n=0}^{N-1-k} x^*[n]x[n+k]; \stackrel{\wedge}{r(-k)} = \stackrel{\wedge}{r(k)} \stackrel{\wedge}{r(k)} for k \ge 0$

- The higher lags of r(k), the poorer estimates since the estimates involve fewer terms of lag products in the averaging operation
- Solution: weigh the higher lags less
 - Trade variance with bias

Windowing

• Use a window function to weigh the higher lags less

i.e.
$$\hat{P}_{Win}(f) = \sum_{K=-(N-1)}^{N-1} W(K) \hat{\Gamma}(K) e^{-j2\pi fK}$$

where $W(K)$ is a "lag window" with properties of:
 $\bigcirc 0 \le W(K) \le W(0] = 1$ $w(0)=1$ preserves variance $r(0)$
 $\bigcirc W(-K) = W(K)$ symmetric
 $\bigcirc W(K) = 0$ for $|K| > M$ where $M \le N-1$
 $\bigoplus W(f)$ must be chosen to ensure $\hat{P}_{Win}(f) \ge 0$

- Effect: periodogram smoothing
 - Windowing in time \(Convolution/filtering the periodogram)
 - Also known as the Blackman-Tukey method

Common Lag Windows

• Much of the art in non-parametric spectral estimation is in choosing an appropriate window (both in type and length)

Name	Definition	Fourier Transform
Rectangular	$w(k) = \begin{cases} 1, & k \le M \\ 0, & k > M \end{cases}$	$W(\omega) = W_{\mathcal{R}}(\omega)$ $= \frac{\sin \frac{\omega}{2}(2M + 1)}{\sin \omega/2}$
Bartlett	$w(k) = \begin{cases} 1 - \frac{ k }{M}, & k \le M \\ 0, & k > M \end{cases}$	$W(\omega) = W_{B}(\omega)$ $= \frac{1}{M} \left(\frac{\sin M\omega/2}{\sin \omega/2} \right)^{2}$
Hanning	$w(k) = \begin{cases} \frac{1}{2} + \frac{1}{2}\cos\frac{\pi k}{M}, & k \le M\\ 0, & k > M \end{cases}$	$W(\omega) = \frac{1}{4} W_R(\omega - \pi/M) + \frac{1}{2} W_R(\omega) + \frac{1}{4} W_R(\omega + \pi/M)$
Hamming	$w(k) = \begin{cases} 0.54 + 0.46 \cos \frac{\pi k}{M}, & k \le M\\ 0, & k > M \end{cases}$	$W(\omega) = 0.23 W_{R}(\omega - \pi/M) + 0.54 W_{R}(\omega) + 0.23 W_{R}(\omega + \pi/M)$
Parzen	$w(k) = \begin{cases} 2\left(1 - \frac{ k }{M}\right)^3 - \left(1 - 2\frac{ k }{M}\right)^3, & k \le M/2\\ 2\left(1 - \frac{ k }{M}\right)^3, & \frac{M}{2} \le k \le N \end{cases}$	2 $W(\omega) = \frac{8}{M^3} \left(\frac{3}{2} \frac{\sin^4 M \omega / 4}{\sin^4 \omega / 2} \right)$ Table $M = -\frac{\sin^4 M \omega / 4}{\sin^2 \omega / 2}$

Table 2.1 common lag window (from Lim-Oppenheim book)

Nonparametric spectral estimation [46]

Discussion: Estimate r(k) via Time Average

• Normalizing the sum of (N-k) pairs

by a factor of 1/N? v.s. by a factor of 1/(N-k)? <u>Biased</u> (low variance) <u>Unbiased</u> (may not non-neg. definite) $\hat{P}_{(k)} = \frac{1}{N} \sum_{n=0}^{N-1-k} X(n+K) X(n), \quad \hat{P}_{(k)} = \frac{1}{N-k} \sum_{n=0}^{N-1-k} X(n+K) X(n)$ $E(\hat{\Gamma}_{1}(k)) =$ $E(\Gamma(K)) =$ Hints on showing \bullet the non-negative definiteness: using $r_1(k)$ to construct correlation matrix • For $r_2(k)$ HW#8

Discussion: Estimate r(k) via Time Average

• Normalizing the sum of (N-k) pairs

by a factor of 1/N? v.s. by a factor of 1/(N-k)? Biased (low variance) <u>Unbiased</u> (may not non-neg. definite) $\hat{P}_{(k)} = \frac{1}{N} \sum_{n=0}^{N-1-k} X(n+K) X(n), \quad \hat{P}_{(k)} = \frac{1}{N-K} \sum_{n=0}^{N-1-k} X(n+K) X(n)$ $E(\hat{\Gamma}_{N}(K)) = \frac{N-K}{N}\Gamma(K) \qquad E(\hat{\Gamma}_{N}(K)) = \Gamma(K)$ Hints on showing $\hat{R}_{N} = X^{H}X$, where Hints on showing the non-negative definiteness: using $r_1(k)$ to construct correlation matrix $X = \sqrt{N} \begin{bmatrix} X(0) & 0 & 0 \\ X(1) & X(0) & 0 \\ X(1) & X(1) & 0 \\ X(1) & X(1) & X(1) \\ X(1) & X(1) & X$ \bullet • For $r_2(k)$ HW#8

3.1.4 Minimum Variance Spectral Estimation (MVSE)

- Recall: filter bank perspective of periodogram
 - The periodogram can be viewed as estimating the p.s.d. by forming a bank of narrowband filters with sinc-like response
 - The high sidelobe can lead to "leakage" problem:
 - large output power due to p.s.d outside the band of interest
- MVSE designs filters to minimize the leakage from out-ofband spectral components
 - Thus the shape of filter is dependent on the frequency of interest and data adaptive

(unlike the identical filter shape for periodogram)

– MVSE is also referred to as the Capon spectral estimator

Main Steps of MVSE Method

- Design a bank of bandpass filters H_i(f) with center frequency f_i so that
 - Each filter rejects the maximum amount of out-of-band power
 - And passes the component at frequency f_i without distortion
- Filter the input process { x[n] } with each filter in the filter bank and estimate the power of each output process
- Set the power spectrum estimate at frequency *f_i* to be the power estimated above divided by the filter bandwidth

Formulation of MVSE

The MVSE designs a filter H(f) for each frequency of interest f_0

minimize the output power

(i.e., to pass the components at f₀ w/o distortion)

Formulation of MVSE

The MVSE designs a filter H(f) for each frequency of interest f_0

minimize the output power

$$\rho = \int_{-\frac{1}{2}}^{+\frac{1}{2}} \left| H(f) \right|^2 P(f) df$$

subject to
$$H(f_0) = 1$$

(i.e., to pass the components at f₀ w/o distortion)



Deriving MVSE Solutions

Output Power From H(f) filter

From the filter bank perspective of periodogram:

$$H(f) = \sum_{n=-(N-1)}^{0} h[n] e^{-j2\pi fn}$$

$$\rho = \int_{-\frac{1}{2}}^{\frac{1}{2}} \sum_{k=-(N-1)}^{0} h[k] e^{-j2\pi fk} \sum_{l=-(N-1)}^{0} h^*[l] e^{j2\pi fl} P(f) df$$

Equiv. to filter r(k)with { $h(k) \otimes h^*(-k)$ } and evaluate at output time k=0

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Nonparametric spectral estimation [54]

Output Power From H(f) filter

From the filter bank perspective of periodogram:

$$H(f) = \sum_{n=-(N-1)}^{0} h[n] e^{-j2\pi fn}$$



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Matrix-Vector Form of MVSE Formulation



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Nonparametric spectral estimation [56]

Matrix-Vector Form of MVSE Formulation

Define
$$\begin{bmatrix} h(0) \\ h(-1) \\ \vdots \\ h(-(N-1)) \end{bmatrix} \Rightarrow f = h^{H} R^{T} h$$
$$[h(0), h(-1), \dots, h(1-N)] \begin{bmatrix} h(0) h(-1) & \cdots \\ h(1) h(0) \\ \vdots \\ \vdots \\ h(-(N-1)) \end{bmatrix}$$
$$[h(0), h(-1), \dots, h(1-N)] \begin{bmatrix} h(0) h(-1) & \cdots \\ h(0) h(-1) \\ \vdots \\ \vdots \\ h(-1) h(0) \end{bmatrix}$$
$$\Rightarrow The constraint can be written in vector form as $\underline{h}^{H} \underline{e} = 1$
$$\underbrace{e^{j2\pi f_{0}}}_{H(f_{0})}$$$$

Thus the problem becomes

$$\min_{\underline{h}} \underline{h}^{H} R^{T} \underline{h} \qquad \text{subject to} \quad \underline{h}^{H} \underline{e} = 1$$

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Nonparametric spectral estimation [57]

Solution of MVSE

$$J \stackrel{def}{=} \underline{h}^{H} R^{T} \underline{h} + \operatorname{Re} \left[2\lambda (1 - \underline{h}^{H} \underline{e}) \right]$$

- Use Lagrange multiplier approach for solving the constrained optimization problem
 - Define real-valued objective function s.t. the stationary condition can be derived in a simple and elegant way based on the theorem for complex derivative/gradient operators

$$\begin{split} \min_{\underline{h},\lambda} J &= \underline{h}^{H} R^{T} \underline{h} + \lambda (1 - \underline{h}^{H} \underline{e}) + \left[\lambda (1 - \underline{h}^{H} \underline{e}) \right]^{*} \implies \underline{h} = \lambda \left(R^{T} \right)^{-1} \underline{e} \text{ and } \underline{h}^{H} \underline{e} = 1 \\ &= \underline{h}^{H} R^{T} \underline{h} + \lambda (1 - \underline{h}^{H} \underline{e}) + \lambda^{*} (1 - \underline{e}^{H} \underline{h}) \\ \text{either } \nabla_{\underline{h}^{*}} J &= 0 \implies R^{T} \underline{h} - \lambda \underline{e} = 0 \\ \text{or } \nabla_{\underline{h}} J = 0 \implies \left(\underline{h}^{H} R^{T} \right)^{T} - \lambda^{*} \underline{e}^{*} = 0 \\ \implies \left(R^{T} \right)^{H} \underline{h} - \lambda \underline{e} = 0 \implies R^{T} \underline{h} - \lambda \underline{e} = 0 \end{split}$$

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Solution of MVSE (cont'd)

The optimal filter:

$$\underline{h} = \frac{\left(R^{T}\right)^{-1} \underline{e}}{\underline{e}^{H} \left(R^{T}\right)^{-1} \underline{e}}$$

It follows that

$$\rho = \underline{h}^{H} R^{T} \underline{h} = \underline{h}^{H} \lambda R^{T} (R^{T})^{-1} \underline{e}$$
$$= \lambda \underline{h}^{H} \underline{e} = \lambda = \frac{1}{\underline{e}^{H} (R^{T})^{-1} \underline{e}}$$

MVSE: Summary

If choosing the bandpass filters to be FIR of length p, its 3dB-b.w. is approximately 1/p

Thus the MVSE is

$$\hat{P}_{\rm MV}(f) = \frac{p}{\underline{e}^{H}(\hat{R}^{T})^{-1}\underline{e}}$$

(i.e. normalize by filter b.w.)

 $\hat{R} \text{ is } p \times p$ correlation matrix $e = \begin{bmatrix} 1 \\ exp(j2\pi f) \\ \vdots \\ exp(j2\pi f(p-1)) \end{bmatrix}$

- MVSE is a data adaptive estimator and provides improved resolution over periodogram
 - Also referred to as "High-Resolution Spectral Estimator"
 - Does not assume a particular underlying model for the data

Recall: Case Study on Non-parametric Methods

• Test case: a process consists of narrowband components (sinusoids) and a broadband component (AR)

 $- x[n] = 2 \cos(\omega_1 n) + 2 \cos(\omega_2 n) + 2 \cos(\omega_3 n) + z[n]$ where $z[n] = -a_1 z[n-1] + v[n], a_1 = -0.85, \sigma^2 = 0.1$ $\omega_1/2\pi = 0.05, \omega_2/2\pi = 0.40, \omega_3/2\pi = 0.42$

- N=32 data points are available \rightarrow periodogram resolution f = 1/32
- Examine typical characteristics of various non-parametric spectral estimators

(Fig.2.17 from Lim/Oppenheim book)





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Nonparametric spectral estimation [62]

<u>Reference</u>

Recall: Filtering a Random Process



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Nonparametric spectral estimation [64]

Chi-Squared Distribution

If
$$x cn] \sim iid N(o,1)$$
 for $n=0,1,...,N-1$, and
 $y = \sum_{n=0}^{N-1} x^{*} cn]$,
then y follows chi-squared distribution of
degree N, i.e. $y \sim Xn^{*}$
and $E[y] = N$, $Var(y) = 2N$

Chi-Squared Distribution (cont'd)

$$p.d.f. of \forall \sim \chi_{N}^{2}:$$

$$p(\forall) = \begin{cases} \frac{1}{2^{N/2} \prod (N/2)} & \forall 2^{N/2} - 1 - \frac{\forall}{2} \\ 0 & \text{if } \forall \geq 0 \end{cases}$$

$$p(\forall) = \begin{cases} \frac{1}{2^{N/2} \prod (N/2)} & \text{if } \forall \geq 0 \\ 0 & \text{if } \forall \leq 0 \end{cases}$$

$$\text{Where } \prod (\cdot) \text{ is the gamma integral}$$

$$\prod (\chi+1) = \int_{0}^{\infty} \forall^{\chi} e^{-\forall} d\forall \text{ for } \chi > -1.$$

$$Note \text{ if } \chi \text{ is an integer}, \prod (n+1) = n \prod (n) = n!$$

Periodogram of White Gaussian Process

For
$$f_{K} = K/N$$
, it can be shown that

$$\begin{cases} \frac{2\hat{P}_{PER}(f_{K})}{P(f_{N})} \sim \chi_{12}^{2} \quad \text{for } K=1,2,\dots,\frac{N-1}{2}, \\ P(f_{N}) \\ \frac{\hat{P}_{PER}(f_{K})}{P(f_{N})} \sim \chi_{11}^{2} \quad \text{for } K=0, \frac{N}{2} \end{cases}$$

$$\Rightarrow E[\hat{P}_{PER}(f\kappa)] = P(f\kappa), \kappa = 0, 1, \dots, N/2$$

$$Var[\hat{P}_{PER}(f\kappa)] = \begin{cases} P^2(f\kappa), \kappa = 1, \dots, \frac{N}{2} - 1\\ 2P^2(f\kappa), \kappa = 0, \frac{N}{2} \end{cases}$$

See proof in Appendix 2.1 in Lim-Oppenheim Book: - Basic idea is to examine the distribution of real and imaginary part of the DFT, and take the magnitude

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Stamp image from USPS web



Happy Thanksgivings!

Nonparametric spectral estimation [68]