



Digital Signal Processing 2 Les 6: Spectrale analyse

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Digital Signal Processing 2: Vakinhoud

- Les 1: Eindige woordlengte
- Les 2: Lineaire predictie
- Les 3: Optimale filtering
- Les 4: Adaptieve filtering
- Les 5: Detectieproblemen
- Les 6: Spectrale signaalanalyse
- Les 7: Schattingsproblemen 1
- Les 8: Schattingsproblemen 2
- Les 9: Sigma-Deltamodulatie
- Les 10: Transformatiecodering

Les 6: Spectrale analyse

Deterministic signals

short-time DFT, windowing, frequency measurement, spectrogram...

+ MATLAB exercise

Random signals

periodogram, periodogram averaging, periodogram smoothing + MATLAB exercise

Les 6: Literatuur

Deterministic signals

- B. Porat, A Course in Digital Signal Processing
- Ch. 6, "Practical Spectral Analysis"
 - Section 6.1, "The Effect of Rectangular Windowing"
 - Section 6.2, "Windowing"
 - Section 6.3, "Common Windows"
 - Section 6.4, "Frequency Measurement"

Random signals

- B. Porat, A Course in Digital Signal Processing
- Ch. 2, "Review of Frequency-Domain Analysis"
 - Section 2.9, "Discrete-Time Random Signals"
- Ch. 13, "Analysis and Modeling of Random Signals"
 - Section 13.1, "Spectral Analysis of Random Signals"

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• Section 13.2, "SA by Smoothed Periodogram"

Les 6: Spectrale analyse

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short-time DFT, windowing, frequency measurement, spectrogram...

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Deterministic signals: overview

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- Introduction: motivating example
- Rectangular windowing
- Windowing
- Frequency measurement
- Spectrogram
 - + MATLAB exercise

Introduction: motivating example (1)

- Example from musical signal processing
 - spectral analysis of Brahms' 4th symphony
 - motivation: automatic music transcription
 - some numbers:
 - 40 min of music
 - 44.1 kHz sampling rate
 - O(10⁸) samples
 - note:
 - [Porat, 1996] "such a task is still beyond our ability"
 - status in 2014: commercial software available, research ongoing



Introduction: motivating example (2)

- Naïve approach: calculate *N*-point DFT with $N = O(10^8)$
 - extremely high frequency resolution \sim 0.4 mHz
 - inefficient in terms of memory and processing resources
 - useless since result will be long-term spectrum average
- Meaningful approach: calculate sequence of short DFTs
 - naturally leads to time-frequency signal representation
 - = essence of (short-time) spectral analysis
 - example:
 - DFT length = 4096 (~ 92.9 ms signal segments)
 - frequency resolution = 11 Hz
 - 50 % overlap between successive signal segments
 - symphony = 52000 signal segments ⇒ 52000 length-4096 DFTs

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Introduction: motivating example (3)

Example of eight 92.9 ms segments from Brahms' symphony

time-domain waveforms

 magnitude spectra



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Rectangular windowing (1)

- Signal segmentation = rectangular windowing
 - consider picking short segment x[n] from long signal y[n]

$$x[n] = \begin{cases} y[n], & 0 \le n \le N-1 \\ 0, & \text{otherwise} \end{cases}$$

this operation can also be written as a multiplication of *y*[*n*] with a rectangular window

 $x[n] = y[n]w_r[n]$

where the window is defined as

$$w_r[n] = \begin{cases} 1, & 0 \le n \le N-1 \\ 0, & \text{otherwise} \end{cases}$$

Rectangular windowing (2)

- **Key question:** how is Fourier transform of rectangularwindowed signal related to that of original signal?
- Example:
 - exponential signal and its Fourier transform:

$$y[n] = \begin{cases} a^n, & n \ge 0\\ 0, & n < 0 \end{cases} \quad Y^{f}(\theta) = \frac{1}{1 - ae^{-j\theta}}$$

- Fourier transform of rectangular-windowed signal

$$X^{f}(\theta) = \sum_{n=0}^{N-1} a^{n} e^{-j\theta n} = \frac{1 - a^{N} e^{-j\theta N}}{1 - a e^{-j\theta}}$$

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(note: apply sum formula for geometric series)

Rectangular windowing (3)

• **Example**: *a* = 0.9, *N* = 16



- magnitude spectrum of x[n] approximates that of y[n]
- magnitude spectrum of y[n] exhibits smoother behavior

Rectangular windowing (4)

- General result:
 - time-domain multiplication = frequency-domain convolution

$$x[n] = y[n]w_r[n] \Rightarrow X^{\mathrm{f}}(\theta) = \frac{1}{2\pi} \left\{ Y^{\mathrm{f}} * W^{\mathrm{f}}_{\mathrm{r}} \right\}(\theta)$$

- Fourier transform of rectangular window:

$$W_{\rm r}^{\rm f}(\theta) = \sum_{n=0}^{N-1} e^{-j\theta n} = \frac{1 - e^{-j\theta N}}{1 - e^{-j\theta}}$$
$$= \underbrace{\frac{\sin(0.5\theta N)}{\sin(0.5\theta)}}_{\text{magnitude}} \underbrace{e^{-j0.5\theta(N-1)}}_{\text{phase}}$$

Rectangular windowing (5)

General result:

magnitude spectrum of rectangular window = Dirichlet kernel

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$$D(\theta, N) = \frac{\sin(0.5\theta N)}{\sin(0.5\theta)}$$



Rectangular windowing (6)

- Dirichlet kernel:
 - maximum value = *N*, occuring at $\theta = 0$
 - main lobe between zero crossings at $\theta = \pm \frac{2\pi}{N}$
 - side lobes between zero crossings at $\, \theta = \pm {2m\pi\over N}, \,\, m>1$
 - highest side lobe occurs at $\theta = \pm \frac{3\pi}{N}$ with amplitude $\approx \frac{2N}{3\pi}$

$$D(\theta, N) = \frac{\sin(0.5\theta N)}{\sin(0.5\theta)}$$

Rectangular windowing (7)

- Distortions due to rectangular windowing:
 - **smearing:** spectral lines become spectral lobes
 - bandwidth of main lobe is non-zero
 - loss of frequency resolution
 - neighboring spectral lines ($\Delta_{ heta} \leq rac{4\pi}{N}$) will become unseparable
 - side-lobe interference: leakage of energy into other bins
 - weak spectral lines can be masked by strong spectral lines in other frequency bins $\int_{N}^{D(\theta,N)}$
 - worst-case effect when weak/strong spectral lines are separated by $\Delta_{\theta} = \frac{(2m+1)\pi}{N}$

$$D(\theta, N) = \frac{\sin(0.5\theta N)}{\sin(0.5\theta)}$$

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Windowing (1)

- Rationale of window design:
 - time-domain multiplication = frequency-domain convolution

$$x[n] = y[n]w[n] \Rightarrow X^{\mathrm{f}}(\theta) = \frac{1}{2\pi} \left\{ Y^{\mathrm{f}} * W^{\mathrm{f}} \right\}(\theta)$$

- constraints on window sequence w[n]
 - finite duration
 - length *N* must agree with desired segmentation length
 - non-negative
- desirable properties of window sequence w[n]
 - main lobe as narrow as possible
 - side lobes as low as possible



Windowing (2)

- Rationale of window design:
 - Fourier transform of window sequence = kernel function
 - desired kernel shape = frequency-domain delta function (why don't we choose $W^{\rm f}(\theta)=2\pi\delta(\theta)$?)



window design = trade-off between main lobe width and side lobe levels

Windowing (3)

1. Rectangular window (revisited):

- main lobe width = $4\pi/N$

= narrowest of all

length-N windows!

- side lobe level = -13.5 dB
- example: N = 41



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Windowing (4)

2. Bartlett window (= triangular window):

- rationale: squaring kernel function halves side lobe level
- implementation:
 - convolve rectangular time-domain window sequence with itself
 - in case N is odd, start from length-(N+1)/2 rectangular window:

$$w_{t}[n] = \frac{2}{N+1} \{ w_{r} * w_{r} \} [n] = 1 - \frac{|2n - N + 1|}{N+1}, \quad 0 \le n \le N - 1$$

• corresponding kernel function:

$$W_{\rm t}^{\rm f}(\theta) = \frac{2}{N+1} D^2\left(\theta, 0.5(N+1)\right) e^{-j0.5\theta(N-1)} = \frac{2\sin^2\left\{0.25\theta(N+1)\right\}}{(N+1)\sin^2(0.5\theta)} e^{-j0.5\theta(N-1)}$$

• (note: similar expressions exist for case of even *N*)

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Windowing (5)

2. Bartlett window (= triangular window):

- main lobe width = $8\pi/(N+1)$
- side lobe level = -27 dB
- example: N = 41



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Windowing (6)

3. Hann window (= cosine window):

- rationale: superposition of 3 frequency-shifted Dirichlet kernels to reduce side lobe level
- implementation:
 - exploit modulation theorem of Fourier transform:

$$W_{\rm hn}^{\rm f}(\theta) = 0.5W_{\rm r}^{\rm f}(\theta) - 0.25W_{\rm r}^{\rm f}\left(\theta - \frac{2\pi}{N-1}\right) - 0.25W_{\rm r}^{\rm f}\left(\theta + \frac{2\pi}{N-1}\right)$$
$$w_{\rm hn}[n] = 0.5 - 0.25\exp\left(\frac{j2\pi n}{N-1}\right) - 0.25\exp\left(-\frac{j2\pi n}{N-1}\right)$$

$$= 0.3 - 0.25 \exp\left(\frac{1}{N-1}\right) - 0.25 \exp\left(-\frac{1}{N-1}\right)$$
$$= 0.5 \left[1 - \cos\left(\frac{2\pi n}{N-1}\right)\right], \quad 0 \le n \le N-1$$

Windowing (7)

3. Hann window (= cosine window):

- main lobe width = $8\pi/N$
- side lobe level = -32 dB
- example: N = 41
- note: two end points = 0
- Modified Hann window:
 - start from length-(N+2)
 rectangular window
 - delete zero end points

$$w_{\rm hn}[n] = 0.5 \left\{ 1 - \cos\left[\frac{2\pi(n+1)}{N+1}\right] \right\},\$$
$$0 \le n \le N-1$$



Windowing (8)

4. Hamming window (= raised cosine window):

- rationale: superposition of 3 frequency-shifted Dirichlet kernels to reduce side lobe level
- implementation:
 - similar to Hann window
 - weights chosen to minimize side lobe level



$$W_{\rm hm}^{\rm f}(\theta) = 0.54W_{\rm r}^{\rm f}(\theta) - 0.23W_{\rm r}^{\rm f}\left(\theta - \frac{2\pi}{N-1}\right) - 0.23W_{\rm r}^{\rm f}\left(\theta + \frac{2\pi}{N-1}\right)$$
$$w_{\rm hm}[n] = 0.54 - 0.46\cos\left(\frac{2\pi n}{N-1}\right), \ 0 \le n \le N-1$$

Windowing (9)

4. Hamming window (= raised cosine window):

- main lobe width = $8\pi/N$
- side lobe level = -43 dB
- example: N = 41
- note: two end points $\neq 0$



Windowing (10)

- 5. Blackman window:
 - rationale: superposition of 5 frequency-shifted Dirichlet kernels to reduce side lobe level
 - implementation:
 - similar to Hamming window
 - 5 instead of 3 Dirichlet kernels

$$W_{\rm b}^{\rm f}(\theta) = 0.42W_{\rm r}^{\rm f}(\theta) - 0.25W_{\rm r}^{\rm f}\left(\theta + \frac{2\pi}{N-1}\right) - 0.25W_{\rm r}^{\rm f}\left(\theta - \frac{2\pi}{N-1}\right) + 0.04W_{\rm r}^{\rm f}\left(\theta + \frac{4\pi}{N-1}\right) + 0.04W_{\rm r}^{\rm f}\left(\theta - \frac{4\pi}{N-1}\right) \\ w_{\rm b}[n] = 0.42 - 0.5\cos\left(\frac{2\pi n}{N-1}\right) + 0.08\cos\left(\frac{4\pi n}{N-1}\right), \ 0 \le n \le N-1$$

Windowing (11)

5. Blackman window:

- main lobe width = $12\pi/N$
- side lobe level = -57 dB
- example: N = 41
- note: two end points = 0



Modified Blackman window:

- start from length-(N+2)
 rectangular window
- delete zero end points



Windowing (12)

6. Kaiser window:

 rationale: calculate family of windows as solution to constrained optimization problem

 $\begin{array}{ll} \min_{w_{k}[n]} & \text{main lobe width} \\ \text{s.t.} & \underline{\text{side lobe energy}} < \beta \end{array}$

- solution (with I_0 = modified Bessel function of order zero):

$$w_{\mathbf{k}}[n] = \frac{I_0 \left[\alpha \sqrt{1 - \left(\frac{|2n-N+1|}{N-1}\right)^2} \right]}{I_0[\alpha]}, \ 0 \le n \le N-1$$

 parameter α determines trade-off between main lobe width and side lobe energy
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Windowing (13)

6. Kaiser window (α = 12):

- main lobe width = $16\pi/N$
- side lobe level = -90 dB (!)
- example: N = 41







Deterministic signals: overview

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Frequency measurement (1)

- Frequency measurement of single complex sinusoid:
 - in theory, frequency of single complex sinusoid

$$y[n] = Ae^{j(\theta_0 n + \phi_0)}, \ 0 \le n \le N - 1$$

can be uniquely determined from its Fourier transform

$$Y^{f}(\theta) = Ae^{-j[0.5(\theta - \theta_{0})(N-1) - \phi_{0}]}D(\theta - \theta_{0}, N)$$

by finding maximum of magnitude spectrum

$$\theta_0 = \arg\min_{\theta} |Y^{\mathrm{f}}(\theta)|$$



Frequency measurement (2)

- Frequency measurement of single complex sinusoid:
 - in practice, however, DFT is calculated on frequency grid generally not including θ_0
 - simple frequency measurement:
 - find frequency bin with maximum magnitude

$$k_0 = \arg\min_k |Y^{d}[k]| \Rightarrow \theta[k_0] = \frac{2\pi k_0}{N}$$

- improved frequency measurement:
 - zero pad sequence y[n] and increase DFT length



Frequency measurement (3)

Frequency measurement of two complex sinusoids:

sum of two complex sinusoids and its frequency spectrum

$$y[n] = A_1 e^{j(\theta_1 n + \phi_1)} + A_2 e^{j(\theta_2 n + \phi_2)}, \ 0 \le n \le N - 1$$

$$Y^{f}(\theta) = A_1 e^{-j[0.5(\theta - \theta_1)(N - 1) - \phi_1]} D(\theta - \theta_1, N)$$

$$+ A_2 e^{-j[0.5(\theta - \theta_2)(N - 1) - \phi_2]} D(\theta - \theta_2, N)$$

- evaluation of Fourier transform at one sinusoid frequency θ_1

$$Y^{f}(\theta_{1}) = NA_{1}e^{j\phi_{1}} + A_{2}e^{-j[0.5(\theta_{1}-\theta_{2})(N-1)-\phi_{2}]}D(\theta_{1}-\theta_{2},N)$$

- frequency measurement only feasible when

$$|A_2 D(\theta_1 - \theta_2, N)| \ll NA_1 \Leftrightarrow \begin{cases} |\theta_2 - \theta_1| \geq 2\pi/N \\ O(A_2) = O(A_1) \end{cases}$$

Frequency measurement (4)

- Frequency measurement of two complex sinusoids:
 - example:



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Frequency measurement (5)

- Frequency measurement of two complex sinusoids:
 - windowed sum of two complex sinusoids x[n] = y[n]w[n] has Fourier transform

$$X^{\mathrm{f}}(\theta) = A_1 e^{j\phi_1} W^{\mathrm{f}}(\theta - \theta_1, N) + A_2 e^{j\phi_2} W^{\mathrm{f}}(\theta - \theta_2, N)$$

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- evaluation of Fourier transform at one sinusoid frequency θ_1

$$X^{f}(\theta_{1}) = A_{1}e^{j\phi_{1}}W^{f}(0,N) + A_{2}e^{j\phi_{2}}W^{f}(\theta_{1}-\theta_{2},N)$$

- frequency measurement feasible with window for which $|A_2 W^{\rm f}(\theta_1 \theta_2, N)| \ll A_1 \sum_{n=0}^{N-1} w[n]$
 - $| heta_2 heta_1|$ greater than kernel main lobe width
 - $20 \log_{10}(A_1/A_2)$ larger than side-lobe level

Frequency measurement (6)

- Frequency measurement of two complex sinusoids:
 - example: Hann window



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Frequency measurement (7)

- Frequency measurement of *M* real sinusoids:
 - real sinusoids can be decomposed as sum of complex sinusoids, hence previous results still hold
 - frequency measurement is feasible without windowing if
 - sinusoid frequencies are separated by at least $2\pi/N$
 - $\pi/N < \text{sinusoid frequencies} < \pi(1 1/N)$
 - $O(A_1) = \ldots = O(A_M)$
 - frequency measurement is feasible with windowing if
 - sinusoid frequencies are separated by at least $\frac{1}{2}$ main lobe width
 - $\frac{1}{2}$ main lobe width < sinusoid frequencies < $\pi \frac{1}{2}$ main lobe width
 - sinusoid amplitude differences > side lobe level

Frequency measurement (8)

- Practice of frequency measurement
 - 1. multiply sampled sequence by window
 - 2. compute DFT (using FFT algorithm)
 - 3. search for local maxima in magnitude spectrum



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Spectrogram

- Spectrogram = 2-D color plot of DFT log-magnitude for number of overlapping windowed signal segments
 - example: speech signal [Rice University]



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MATLAB exercise (1)

Consider signal $y[n] = \cos 0.35\pi n + \cos 0.4\pi n + 0.25\cos 0.8\pi n$

- 1. Draw magnitude spectrum by hand
- 2. Generate length-*N* segment of this signal in Matlab
- 3. Calculate and plot magnitude spectrum for N = 21
- 4. Calculate and plot magnitude spectrum for *N* = 21 with zero padding up till 2048 samples
- 5. Calculate and plot magnitude spectrum for N = 81 with zero padding up till 2048 samples
- 6. Calculate and plot magnitude spectrum for N = 81 with zero padding up till 2048 samples and Hamming windowing
- 7. Compare and explain results



MATLAB exercise (2)

Consider signal $y[n] = \cos 0.35\pi n + \cos 0.4\pi n + 0.25\cos 0.8\pi n$

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Les 6: Spectrale analyse

Deterministic signals

short-time DFT, windowing, frequency measurement, spectrogram...

+ MATLAB exercise

Random signals

periodogram, periodogram averaging, periodogram smoothing + MATLAB exercise

Random signals: overview

- Introduction: motivating example
- Averaged periodogram
- Smoothed periodogram

+ MATLAB exercise



Introduction: motivating example (1)

- Example from oceanography
 - measurement of time variation in height of ocean waves



- signal shows oscillatory behavior but not sinusoidal behavior
- repeating measurement yields similar but not same signal

Introduction: motivating example (2)

- Example from oceanography
 - how do we perform spectral analysis of random signals?
 - short-time spectral analysis (as for deterministic signals)? No!
 - random signals have random Fourier transform
 - DFT magnitude spectrum will look very noisy
 - spectrum details vary from experiment to experiment
 - zero padding does not solve this problem



some sort of averaging or smoothing is required

Random signals: overview

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+ MATLAB exercise



Averaged periodogram (1)

- Averaged periodogram
 - periodogram = square magnitude of DFT
 - divide length-NL signal x[n] into L length-N segments
 - averaged periodogram (AP) over L segments:

$$\hat{K}_{x}^{f}(\theta) = \frac{1}{L} \sum_{l=0}^{L-1} \left\{ \frac{1}{N} \left| \sum_{n=0}^{N-1} x[n+lN] e^{-j\theta n} \right|^{2} \right\}$$

key property: AP converges to power spectral density (PSD)

$$\lim_{\substack{N \to \infty \\ L \to \infty}} \hat{K}_x^{\mathrm{f}}(\theta) = K_x^{\mathrm{f}}(\theta) = E \left\{ \frac{1}{N} \left| \sum_{n=0}^{N-1} x[n+lN] e^{-j\theta n} \right|^2 \right\}$$

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– note: why average periodogram instead of DFT?

Averaged periodogram (2)

- Windowed averaged periodogram
 - since $N, L \neq \infty$ averaged periodogram is still (slightly) random
 - increasing N makes AP more detailed and more random
 - increasing L makes AP smoother with less randomness
 - additional smoothing can be obtained by windowing

$$\hat{K}_{x}^{f}(\theta) = \frac{1}{L} \sum_{l=0}^{L-1} \left\{ \frac{1}{N} \left| \sum_{n=0}^{N-1} w[n]x[n+lN]e^{-j\theta n} \right|^{2} \right\}$$

- example: ocean wave AP
 - *L* = 100
 - *N* = 500
 - *w*[*n*] = Hann window



Averaged periodogram (3)

- Welch periodogram
 - windowing reduces importance of end-of-segment samples
 - compensated by using (50%) overlap in signal segmentation



Welch periodogram = AP with windowing and 50% overlap

$$\hat{K}_{x}^{f}(\theta) = \frac{1}{L} \sum_{l=0}^{L-1} \left\{ \frac{1}{N} \left| \sum_{n=0}^{N-1} w[n]x[n+0.5lN]e^{-j\theta n} \right|^{2} \right\}$$

= standard tool for spectral analysis of stationary random signals

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Smoothed periodogram (1)

- Smoothed periodogram
 - consider case when data sequence is too short for averaging
 - because limited amount of data measurement is available
 - because signal has non-stationary behavior
 - smoothed periodogram (without segmentation or averaging)
 - frequency-domain convolution of periodogram and window kernel

$$\hat{K}_x^{\mathrm{f}}(\theta) = \frac{1}{2\pi} \int_{-\pi}^{\pi} \frac{1}{N} |X^{\mathrm{f}}(\theta - \lambda)|^2 W^{\mathrm{f}}(\lambda) d\lambda$$

• time-domain implementation by length-(2*M*+1) windowing

$$\hat{K}_x^{\mathrm{f}}(\theta) = \sum_{m=-M}^M \hat{\kappa}_x[m]w[m]e^{-j\theta m}$$

Smoothed periodogram (2)

- Smoothed periodogram
 - $\hat{\kappa}_x[m] \sim \text{inverse Fourier transform of periodogram } |X^{\mathbf{f}}(\theta)|^2$
 - $\hat{\kappa}_x[m]$ is estimate of covariance $\kappa_x[m]$ of random signal x[n]

$$\hat{\kappa}_x[m] = \frac{1}{N} \sum_{i=0}^{N-1-|m|} x[i]x[i+|m|] \approx E\left\{x[i]x[i+|m|]\right\} = \kappa_x[m]$$

- smoothed periodogram computational procedure:
 - 1. estimate covariance
 - 2. multiply with window
 - 3. compute DFT
- window length 2(M+1) should always be smaller than 2N-1, and typically 0.2 < M/N < 0.5

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Smoothed periodogram (3)

• Example: sunspot statistics



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MATLAB exercise

- Generate zero-mean unit-variance Gaussian white noise signal (length NL=2¹⁶)
- 2. Filter this signal with FIR filter $H(z) = 1 + 0.8z^{-1}$
- 3. Calculate and plot periodogram of original and filtered signal
- 4. Calculate and plot averaged periodogram of original and filtered signal (try out different combinations of *N* and *L*)