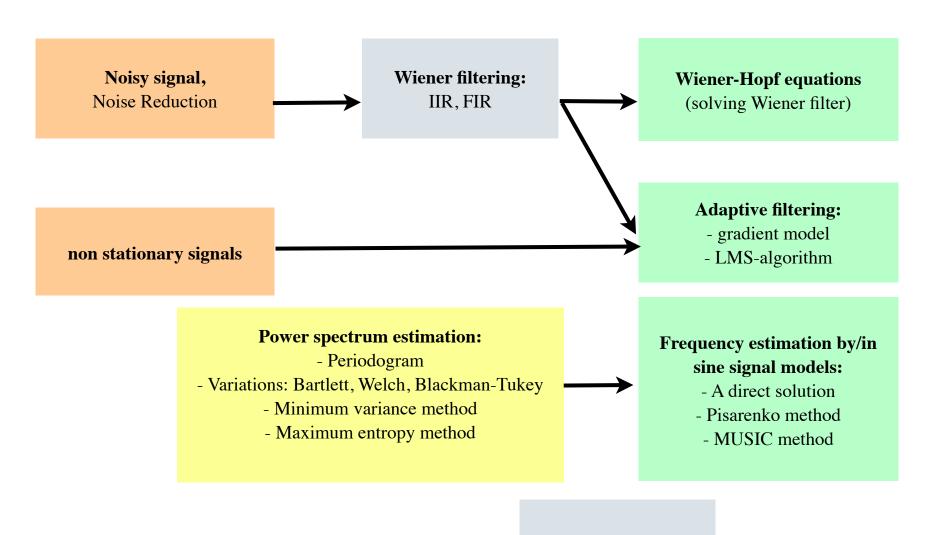


T61.3040

Parametric methods for estimating the power spectrum and Frequency Estimation

Diagram of content of the final part of the course



Estimation of variance:

ARCH models

Today

Power spectrum estimation:

- Periodogram
- Variations: Bartlett, Welch, Blackman-Tukey
 - Minimum variance method
 - Maximum entropy method

Frequency estimation by/in sine signal models:

- A direct solution
- Pisarenko method
- MUSIC method

- Periodogram and other Fourier methods: the process is assumed to be WSS and ergodic in the autocorrelation
- Periodogram resolution is of the order 1/N, when N observations are available
- In this case, for example, two sine signal whose frequency difference is less than 1/N are not distinguished in the periodogram
- In addition, the variance of the periodogram is large



- In parametric methods, the estimated numbers of "quantities" does not grow when the number of observations *N* increases
- If the process x(n) is assumed to have a parametric model then often we can calculated the power spectrum model as a function of parameters
- Especially using sine-signal model provides efficient methods for sine signal spectrum estimation
- Let's first consider two parametric methods, which are not derived from the process model



• We want to calculate the power of process x(n) for frequency ω_i ie $P_x(\exp(j\omega_i))$

- Filtered x(n) by bandpass filter $g_i(n)$, whose passband is $\omega_i \pm \Delta/2$
- Response is $y_i(n) = x(n) * g_i(n)$ the power spectrum is $P_{y_i} = P_x |G_i|^2$



• The total power of the filtered process $y_i(n)$ is approximately

$$E(|y_i(n)|^2) = \frac{1}{2\pi} \int_{\omega_i - \Delta/2}^{\omega_i + \Delta/2} P_x(\exp(j\omega)) d\omega$$
$$\approx P_x(\exp(j\omega_i)) \frac{\Delta}{2\pi}$$

This provides a power spectrum value

$$\hat{P}_x(\exp(j\omega_i)) = \frac{\mathrm{E}(|y_i(n)|^2)}{\Delta/2\pi}$$



- Assume that $g_i(n)$ is FIR filter with p+1 coefficients
- Minimum variance method:
 - select $g_i(n)$ so that the power $E(|y_i(n)|^2)$ as small as possible with condition $|G_i(\exp(j\omega_i))| = 1$
 - The MV estimate of Power spectrum is $E(|y_i(n)|^2)/(\Delta/2\pi)$
 - We calculate the passband bandwidth ∆ so that the results are correct for a white noise



· Let's define

$$g_i = [g_i(0), g_i(1), \dots, g_i(p)]^T$$

$$e_i = [1, \exp(j\omega_i), \exp(j2\omega_i), \dots, \exp(jp\omega_i)]^T$$

This optimization problem must be solved

$$\min_{g_i} E(|y_i(n)|^2) = \min_{g_i} g_i^H R_x g_i, \quad g_i^H e_i = 1$$



• Solution (derived in the book) is

$$g_{i} = \frac{R_{x}^{-1}e_{i}}{e_{i}^{H}R_{x}^{-1}e_{i}}$$

$$\min_{g_{i}} E(|y_{i}(n)|^{2}) = \frac{1}{e_{i}^{H}R_{x}^{-1}e_{i}}$$

The lower formula is the necessary power for the filtered signal



For white noise the

$$\hat{P}_{x} = \frac{E(|y_{i}(n)|^{2})}{\Delta/2\pi} = \frac{2\pi/\Delta}{e_{i}^{H}R_{x}^{-1}e_{i}} = \frac{\sigma^{2}2\pi/\Delta}{e_{i}^{H}e_{i}} = \frac{\sigma^{2}2\pi/\Delta}{p+1}$$

- Required: $\hat{P}_x = P_x = \sigma^2$ then $2\pi/\Delta = p+1$
- Using this in the general Minimum Variance solution

$$\hat{P}_{MV}(\exp(j\omega_i)) = \frac{E(|y_i(n)|^2)}{\Delta/2\pi} = \frac{p+1}{e_i^H R_x^{-1} e_i}$$



In the periodogram we estimate autocorrelation,

$$\hat{r}_x(k)$$
, $|k| < N$

- The remaining autocorrelations were set to zero
- Unknown variables set equal to zero is not necessarily optimal
- Another way: selecting $\hat{r}_x(k)$, $|k| \ge N$ so that the obtained autocorrelation sequence corresponds a random process



- Measuring randomness by means of entropy (not discussed further in this course)
- It can be shown that for a normally distributed WSS process x(n), the entropy is maximized by the estimates $\hat{r}_x(k)$, |k| > pobtained from the AR(p)-process
- YW equations can be solved when values are known



$$r_x(0),\ldots,r_x(p)$$
:

$$\begin{bmatrix} r_x(0) & r_x^*(1) & \dots & r_x^*(p) \\ r_x(1) & r_x(0) & \dots & r_x^*(p-1) \\ \vdots & \vdots & \ddots & \vdots \\ r_x(p) & r_x(p-1) & \dots & r_x(0) \end{bmatrix} \begin{bmatrix} 1 \\ a_p(1) \\ \vdots \\ a_p(p) \end{bmatrix} = \begin{bmatrix} |b(0)|^2 \\ 0 \\ \vdots \\ a_p(p) \end{bmatrix}$$

• Autocorrelation $r_x(p+1)$, $r_x(p+2)$, ... will be obtained by solving the parameters $a_p(k)$ from YW equations



• The power spectrum estimate of the process x(n) is

$$\hat{P}_{mem}(\exp(j\omega)) = \frac{|b(0)|^2}{\left|1 + \sum_{k=1}^{p} a_p(k) \exp(-jk\omega)\right|^2}$$

- This is also the power spectrum of AR(p)-process
- · justification is more general than the assumption AR-model
- the power spectrum of a process that is as random as possible given that part of the autocorrelation is known



- The above analysis was based on the assumption that the estimated autocorrelation values are correct
- As previously, the autocorrelation is estimated from the observations, and this can be done for example by autocorrelation or covariance methods
- Covariance methods generally provide better resolution than the autocorrelation method
- Autocorrelation method has the advantage that R_x has a Toeplitz matrix structure, which can be used when solving the parameters of the AR process



• Example: estimate the power spectrum of a AR (4)-process with different methods. the system generating the process has poles at the points

$$z_{1,2} = 0.98 \exp(\pm j(0.2\pi))$$

$$z_{3,4} = 0.98 \exp(\pm j(0.3\pi))$$

- N = 128 samples.
- Demo: mem.R



Consider a complex sine signal model

$$x(n) = \sum_{i=1}^{p} A_i \exp(jn\omega_i) + v(n), \quad A_i = |A_i| \exp(j\phi_i)$$

- ω_i frequencies and amplitudes $|A_i|$ are unknown constants
- Phases ϕ_i are uncorrelated and uniformly distributed in the interval $[-\pi,\pi)$



• The process x(n) has the power spectrum

$$P_{x}(\exp(j\omega)) = \sum_{i=1}^{p} 2\pi |A_{i}|^{2} \delta(\omega - \omega_{i}) + \sigma^{2}$$

- In the power spectrum, there is some impulse δ at ω_i which is the frequency of the sine signal
- Elsewhere, the value of power spectrum is the noise variance σ^2

- Frequencies ω_i could be estimated by peaks in the periodogram
- This is not necessarily a good solution because of the poor resolution in the periodogram
- Process model provides significantly better methods in the case of sine signal



• One sine signal with noise: $x(n) = A_1 \exp(jn\omega_1) + v(n)$ $= s(n) + v(n), \operatorname{var}(v(n)) = \sigma_n^2$

- Consider M successive values of the processes
- at time n, the observations of the sine sequence s(n) form a vector

$$\underline{s}(n) = A_1[e^{jn\omega_1}, e^{j(n-1)\omega_1}, \dots, e^{j(n-M+1)\omega_1}]^T$$



At the time n + 1 we obtain

$$\underline{s}(n+1) = A_1[e^{j(n+1)\omega_1}, e^{jn\omega_1}, \dots, e^{j(n-M+2)\omega_1}]^T$$
$$= \exp(j\omega_1)\underline{s}(n)$$

- Vectors s(n) and s(n+1) are in the same direction
- The direction of the vector depends on the frequency ω_i

Needed in the future signal vector,

$$e_i = \left[1, e^{j\omega_i}, e^{j2\omega_i}, \dots, e^{j(M-1)\omega_i}\right]^T$$

- Which can be thought as a sine signal with frequency ω_i and complex amplitude A = 1
- In the case of one sine, this vector defines the subspace in which there is the noise-free part of the the process
- (the time index is the other way around as in the example above)



• The process x(n) has autocorrelation $r_x(k)$

$$r_x(k) = P_1 \exp(jk\omega_1) + \sigma_v^2 \delta(k)$$

= $r_s(k) + r_n(k)$, $P_1 = |A_1|^2$

The autocorrelation matrix R_x can be presented as the

$$R_x = R_s + R_n$$

 In other words, the noisy sine signal autocorrelation can be separated into the signal and noise autocorrelations



Using the signal vector, we can write

$$R_s = |A_1|^2 e_1 e_1^H$$

This is easy to verify by calculating the element

$$R_s[k, l] = |A_1|^2 [e_1]_k [e_1^*]_l$$

= $|A_1|^2 \exp(j(k-l)\omega_1)$
= $r_s(k-l)$



- Properties of the matrix R_s (one sine):
 - 1. matrix R_s has one nonzero eigenvalue $\lambda_1^s = MP_1$
 - 2. the corresponding eigenvector v_1 is e_1
 - 3. eigenvectors v_1 , ... v_M are orthogonal with e_1
 - 4. corresponding eigenvalues $\lambda_2^s, \ldots, \lambda_M^s$ are zero



• R_x has the same eigenvectors than R_s :

$$R_x v_i = (R_s + \sigma_v^2 I) v_i = (\lambda_i^s + \sigma_v^2) v_i$$

At the same time, we obtained the eigenvalues

$$\lambda_i = \lambda_i^s + \sigma_v^2$$

- So is the largest eigenvalue is $\lambda_{max} = MP_1 + \sigma_v^2$
- And all other eigenvalues are σ_v^2



- Eigenvalues and vectors contain the necessary information to estimate the power P_1 , the variance σ_v^2 and frequency ω_1 :
 - 1. Calculate the eigenvalues and eigenvectors of the matrix R_x . The largest eigenvalue is $MP_1 + \sigma_v^2$ and the rest are σ_v^2
 - 2. Noise variance is $\sigma_v^2 = \lambda_{min}$ and the sine signal power is

$$P_{1} = \frac{1}{M} \left(\lambda_{max} - \lambda_{min} \right)$$

3. Sine signal frequency can be solved from the eigenvector v_{max} corresponding to the largest eigenvalue, which is

$$v_{max} \propto [1, \exp(j\omega_1), \dots, \exp(j(M-1)\omega_1)]^T$$



Two sine signals:

$$x(n) = A_1 \exp(jn\omega_1) + A_2 \exp(jn\omega_2) + v(n)$$

The autocorrelation function is

$$r_x(k) = P_1 \exp(jk\omega_1) + P_2 \exp(jk\omega_2) + \sigma_v^2 \delta(k)$$

Autocorrelation matrix can be written as the sum

$$R_x = P_1 e_1 e_1^H + P_2 e_2 e_2^H + \sigma_v^2 I$$



• By assembling the vectors e_i and powers P_i as matrices

$$E = [e_1, e_2], \quad P = \begin{bmatrix} P_1 & 0 \\ 0 & P_2 \end{bmatrix}$$

· We can write

$$R_x = EPE^H + \sigma_v^2 I$$
$$= R_s + R_n$$

Where

$$R_s = EPE^H$$

Matrix R_x has M a eigenvalues

$$\lambda_1 \geq \lambda_2 \geq \lambda_3 \geq \cdots \geq \lambda_M \geq 0$$

- Since the matrix R_s has only two nonzero eigenvalues, λ_1 and λ_2 are greater than σ_v^2
- Other eigenvalues are $\lambda_3 = \cdots = \lambda_M = \sigma_v^2$



- The two largest eigenvalues correspond to signal eigenvectors
 v₁ and v₂ which span a signal subspace
- Noise eigenvectors v₃ ... v_M span a M-2 dimensional noise subspace
- Spaces are orthogonal because the R_x is a Hermitian matrix, i.e. its eigenvectors form a orthonormal set



- Is v_1 parallel to e_1 , such as the case of one sine?
- Generally not: However, e₁ and e₂ are in the signal subspace spanned by v_1 and v_2
- It follows that e₁ and e₂ are orthogonal to the eigenvectors v₃ ... V_{M}



- Let's check that the e₁ and e₂ are in the signal subspace
- Because the signal subspace is spanned by the orthogonal vectors v_1 and v_2 , then subspace dimension is 2
- Because $R_s v_i = \lambda_i v_i$ so

$$v_i = \frac{1}{\lambda_i} P_1 e_1(e_1^H v_i) + \frac{1}{\lambda_i} P_2 e_2(e_2^H v_i)$$

= $c_1 e_1 + c_2 e_2$, $i = 1, 2$

 So both of the vectors v₁, v₂ can be represented as a linear combination of vectors e₁, e₂



- Then the vectors v_1 , v_2 are in the subspace spanned by the signal vectors
- But this subspace dimension is also 2, so it must be the same subspace which is spanned by vectors v_1 , v_2
- It also follows that e₁, e₂ are orthogonal to the noise eigenvectors



• The general case, with *p* sine signal:

$$x(n) = \sum_{i=1}^{p} A_i \exp(jn\omega_i) + v(n)$$

$$r_x(k) = \sum_{i=1}^p P_i \exp(jk\omega_i) + \sigma_v^2 \delta(k)$$

$$R_x = R_s + R_n = \sum_{i=1}^{p} P_i e_i e_i^H + \sigma_v^2 I$$

$$R_x = EPE^H + \sigma_v^2 I$$



- Changing the number of sines from 2 to p there is no surprise:
- R_x eigenvectors v_1, \ldots, v_p span the signal subspace which includes all signal vectorst e_1, \ldots, e_p
 - Eigenvalues $\lambda_1, \ldots, \lambda_p > \sigma_v^2$
 - The rest of the eigenvalues $\lambda_{p+1} = \lambda_{p+2} = \cdots = \lambda_M = \sigma_v^2$
 - signal vectors e_1, \ldots, e_p are orthogonal to the noise subspace spanned by $v_{p+1}, v_{p+2}, \ldots v_{M}$



Pisarenko method

- we assume that the number p of sine signals is known (can also be estimated from the observations)
- We estimate R_x which has size $p + 1 \times p + 1$
- Noise-subspace is then spanned by one vector V_{p+1}
- All signal vectors are orthogonal to the vector V_{p+1}



Pisarenko method

This provide the Pisarenko method. Calculate the pseudo-

spectrum

$$\hat{P}_{PHD}(\exp(j\omega)) = \frac{1}{|e^{H}v_{p+1}|^{2}}$$

- Where $e = [1, \exp(j\omega), ..., \exp(jp\omega)]^T$
- The term $|e^{H}v_{p+1}|^{2}$ measures the projection of the signal vector e to the noise subspace



Pisarenko method

- When $w = w_i$ for some i, the projection is zero
- Otherwise the projection has a positive absolute value
- In the pseudo-spectrum, there appears peaks at the right frequencies, because we calculate the inverse (1/value) of the projection
- So the pseudo-spectrum indicates the frequencies of the sine signals, but it is not an estimate of the actual power spectrum



MUSIC method

- Pisarenko, one noise eigenvector
- MUSIC (multiple sinusoid classification): Several noise eigenvectors
- Correlation matrix has size $M \times M$, where M > p
- where there are M-p noise eigenvector v_{p+1}, \ldots, v_{M}
- which are all orthogonal to the signal vectors e_1, \ldots, e_p



MUSIC method

This provides the MUSIC estimator

$$\hat{P}_{MU}(\exp(j\omega)) = \frac{1}{\sum_{i=p+1}^{M} |e^{H}v_{i}|^{2}}$$

- Now, in the denominator is measured the projection of the signal vector onto a M-p dimensional noise subspace
- Again, with the "real" frequencies, the projection is theoretically zero



Amplitude and phase estimation

 Pisarenko and MUSIC provide a pseudo-spectrum, from which can be estimated we can estimate sine signal frequency

• Signal model $x(n) = \sum_{i=1}^{p} A_i \exp(jn\omega_i) + v(n)$ where the frequencies are assumed to be known



Amplitude and phase estimation

We obtain the system of equations

$$\begin{bmatrix} x(n) \\ x(n-1) \\ \vdots \end{bmatrix} = \begin{bmatrix} e^{jn\omega_1} & \dots & e^{jn\omega_p} \\ e^{j(n-1)\omega_1} & \dots & e^{j(n-1)\omega_p} \\ \vdots & \vdots & \vdots \end{bmatrix} \begin{bmatrix} A_1 \\ A_2 \\ \vdots \\ A_p \end{bmatrix} + \begin{bmatrix} v(n) \\ v(n-1) \\ \vdots \\ A_p \end{bmatrix}$$

$$x = EA + v$$

• Minimize $||x - EA||^2 = ||v||^2$

which corresponds to the overdetermined solution to a system of linear equations



Amplitude and phase estimation

- Because the frequencies have been solved then the matrix E is known
- The result is a vector, whose components are $A_i = |A_i| \exp(j\phi_i)$
- i.e., the amplitude is the modulus of A_i and the phase is A_i 's phase
- Demo: music.R

