

#### **Aalto University**

# School of Electrical Engineering Department of Signal Processing and Acoustics

#### S-88.4212 Signal Processing in Telecommunications II Fall 2013

Lecture 2: ML Estimation Principles

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#### **Timetable**

| L1 | Introduction; models | for channels and | comms. systems |
|----|----------------------|------------------|----------------|
|----|----------------------|------------------|----------------|

#### L2**ML Estimation principles**

- **L3** Synchronization: Overview
- **L4** Carrier frequency estimation I
- **L5** Carrier frequency estimation II
- **L6** Carrier phase estimation I
- **L7** Carrier phase estimation II
- **L8** Symbol timing estimation I
- L9 Symbol timing estimation II
- L10 Channel estimation I
- L11 Channel estimation II, course review

**Exam 12.12. Exam Thursday** 12-15 (Check!)

#### Contents of Lecture 2

I What is synchronization

II ML estimation principles

III ML example: sinusoid phase estimation

IV Cramer-Rao lower bound

The material in this lecture is mostly based on the book

S. Kay: Fundamentals of Statistical Signal Processing

- Estimation Theory, Prentice-Hall 1993.

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# I. What is synchronization

#### What is synchronization

#### Synchronous digital transmission

- Information is carried by uniformly spaced pulses
- Signal is known except for:
  - Data symbols
  - Reference parameters
- **Baseband** pulse amplitude modulation:
  - Matched filtering & symbol-rate sampling
  - Optimum sampling time at pulse peaks for max eye opening and min errors
  - Timing (clock) synchronizer

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# What is synchronization...

- Coherent passband transmission:
  - Signal is modulated to a sinusoidal carrier frequency
  - For coherent demodulation, a local reference sinusoid is needed with the same *frequency* and *phase*
  - Requires carrier frequency and phase estimation
- Alternatives to coherent demodulation
  - Differentially coherent: phase difference between consecutive symbol samples sufficient
  - Noncoherent: decisions based on carrier envelope only
  - Simpler to implement, worse performance
  - Not considered in this course!

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# What is synchronization...

- · Higher levels of synchronization
  - Block coding: word synchronizers needed for identification of block boundaries
  - Convolutional coding: fixed symbol segments used, node synchronizers needed to mark the start of each segment
  - Frame synchronizers for time-shared channels (like TDMA)
  - Network synchronization: transmitter timing adjusted in e.g.
     PCM networks
  - Also beyond our scope!

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# What is our synchronization

- Concentrate on the estimation of
  - timing
  - carrier frequency and
  - carrier phase parameters

in passband transmission

Review ML estimation principles first

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II. Maximum Likelihood (ML) Estimation

# Maximum Likelihood (ML) Estimation

- Historical development of digital synchronization algorithms:
  - heuristic methods
  - application of ML estimation methods
- We want to employ the ML method systematically for different synchronization tasks

- Problem of extracting values of parameters from a discrete-time (continuous) waveform or a data-set
- Mathematically, we have the N-point data set  $\{r(0), r(1), \dots r(N-1)\}\$ , which depends on our unknown synchronization parameters, v
- Determine v based on the data, or define an estimator

$$\hat{\mathbf{v}} = g[r(0), r(1), \dots, r(N-1)]$$

where g is some function

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#### ML Estimation...

- To determine good estimators the first step is to mathematically model the data
- Data is inherently  $random \Rightarrow$  we describe it by its probability density function (PDF)

$$p(\mathbf{r}; \mathbf{v}) = p[r(0), r(1), \dots, r(N-1); \mathbf{v}]$$

- PDF is parameterized by the unknown parameter v (denoted by a semicolon)
  - We have a class of PDFs where each one is different due to a different value of v

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• Simple example: N = 1 sample, v denote the mean, the PDF might be

$$p[r(0);v] = \frac{1}{\sqrt{2\pi\sigma^2}} \exp\left\{-\frac{1}{2\sigma^2} [r(0)-v]^2\right\}$$

- Since the value of v affects the probability of r(0), we should be able to guess v from r(0)
- In an actual problem we must choose a proper PDF
  - Should be consistent with the problem constraints
  - Mathematically tractable
- In this course the PDF will take form above (AWGN channel)

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#### ML Estimation...

 Transmitted signal: (baseband model, linear modulation like PAM, MPSK etc.)

$$x(t) = h_{\mathrm{T}}(t) * \sum_{k = -\infty}^{\infty} a_k \delta(t - kT) = \sum_{k = -\infty}^{\infty} a_k h_{\mathrm{T}}(t - kT)$$

 $a_k$  = data symbols to be transmitted

 $h_{\rm T}(t)$  = transmitted continuous-time waveform

 $\delta(t)$  = Dirac delta function

• Received continuous-time signal:

$$r(t) = x(t, \mathbf{v}) + w(t)$$

v = vector of unknown sync parameters

r(t) = received signal waveform

w(t) = additive noise

Consider different trial sets v<sub>1</sub>,v<sub>2</sub> for the sync parameters and the corresponding set of realizations r (vector representation of r(t)) for the received signal

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#### ML Estimation...

• Define probability density function (PDF) of  ${\bf r}$  with the parameter set  ${\bf v}_1$ 

$$p(\mathbf{r}; \mathbf{v}_1)$$

• The parameter set  $\mathbf{v}_1$  is more likely than  $\mathbf{v}_2$  if

$$p(\mathbf{r}; \mathbf{v}_1) > p(\mathbf{r}; \mathbf{v}_2)$$

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• The maximum likelihood (ML) solution: find such parameter set **v** that the probability density function

$$p(\mathbf{r}; \mathbf{v})$$

is maximum, or

$$\hat{\mathbf{v}}_{\mathrm{ML}}(\mathbf{r}) = \arg \left\{ \max_{\mathbf{v}} \left\{ p(\mathbf{r}; \mathbf{v}) \right\} \right\}$$

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#### ML Estimation...

- Sometimes only a subset u of all the parameters v is of interest (and the remaining set u<sub>R</sub> are unwanted)
- Total probability theorem gives the result

$$p(\mathbf{r}; \mathbf{u}) = \int_{-\infty}^{\infty} p(\mathbf{r}; \mathbf{v}) p(\mathbf{u}_{R}) d\mathbf{u}_{R}$$

i.e., the unwanted parameters are *integrated away* (practical approximation: *averaging*)

• The modified ML estimate is then

$$\hat{\mathbf{u}}_{\mathrm{ML}}(\mathbf{r}) = \arg \max_{\mathbf{u}} \{p(\mathbf{r}; \mathbf{u})\}$$

# Scalar parameter ML Estimation

 Simple example: Discrete-time received signal is a DC level in zero-mean additive white Gaussian noise (AWGN):

$$r(k) = A + w(k),$$
  $k = 0,1,...,N-1$ 

where the DC level A is *unknown* but the noise variance  $\sigma^2$  is *known*. The PDF of the observation vector **r** is

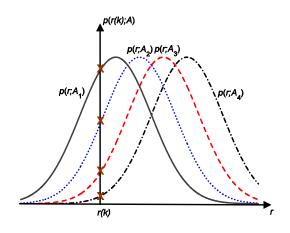
$$p(\mathbf{r}; A) = \frac{1}{(2\pi\sigma^2)^{N/2}} \exp\left[\frac{-1}{2\sigma^2} \sum_{k=0}^{N-1} (r(k) - A)^2\right]$$

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# Scalar parameter ML Estimation...



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#### Scalar parameter ML Estimation...

- PDF is viewed as a continuous function of the unknown parameter (*likelihood function*, LF)
- For an ML estimate we need to find the *A* that maximizes the likelihood function
- For easier optimization, the likelihood function is usually replaced by the log-likelihood function (LLF)

$$\Lambda(\mathbf{r}; A) = \ln p(\mathbf{r}; A) = -\frac{N}{2} \ln(2\pi\sigma^2) - \frac{1}{2\sigma^2} \sum_{k=0}^{N-1} (r(k) - A)^2$$

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#### Scalar parameter ML Estimation...

The maximum is found by setting the first derivative to zero:

$$\frac{\partial}{\partial A} \Lambda(\mathbf{r}; A) = \frac{-1}{2\sigma^2} \sum_{k=0}^{N-1} (-2) (r(k) - A) = 0$$

$$\iff \sum_{k=0}^{N-1} r(k) = NA$$

$$\Leftrightarrow \hat{A} = \frac{1}{N} \sum_{k=0}^{N-1} r(k)$$

• The conventional *averager* thus gives the ML estimate for a DC level in the AWGN case.

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# Vector parameter ML Estimation

• Let us modify the DC-in-AWGN example so that both A and  $\sigma^2$  are unknown. The likelihood function looks the same

$$p(\mathbf{r}; \mathbf{v}) = \frac{1}{(2\pi\sigma^2)^{N/2}} \exp\left[\frac{-1}{2\sigma^2} \sum_{k=0}^{N-1} (r(k) - A)^2\right]$$

but it is now considered as a function of *two* parameters, or a function of the *vector* parameter

$$\mathbf{v} = \left[ A \, \sigma^2 \, \right]^{\mathrm{T}}$$

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# Vector parameter ML Estimation...

• The zero of the derivative of the LLF with respect to (w.r.t.) *A* gives the same equation as before:

$$\frac{\partial}{\partial A} \Lambda(\mathbf{r}; \mathbf{v}) = \frac{-1}{2\sigma^2} \sum_{k=0}^{N-1} (-2) (r(k) - A) = 0$$

• Differentiating w.r.t. to  $\sigma^2$  yields

$$\frac{\partial}{\partial \sigma^2} \Lambda(\mathbf{r}; \mathbf{v}) = \frac{-N}{2\sigma^2} + \frac{1}{2\sigma^4} \sum_{k=0}^{N-1} (r(k) - A)^2 = 0$$

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## Vector parameter ML Estimation...

 Solving these two equations simultaneously gives the ML estimates for the DC level and AWGN variance as

$$\hat{A} = \bar{r} = \frac{1}{N} \sum_{k=0}^{N-1} r(k)$$

$$\hat{\sigma}^{2} = \frac{1}{N} \sum_{k=0}^{N-1} (r(k) - \overline{r})^{2}$$

which are the familiar results for sample mean and variance.

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# Construction of MLE algorithms

- In principle, the design of MLE algorithms is easy:
  - 1) Express the LLF, preferably using AWGN model
  - 2) Find the maximum of the LLF
    - Differentiate w.r.t. parameters and set derivatives to zero
  - 3) Solve for desired parameters
- The solution always in terms of received data samples
- The estimator has good properties (unbiased, efficient)
- Not always in neat closed form though!
- Problems: nonlinear dependencies of parameters
  - Try linearizations & other tricks

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III. ML example: sinusoid phase estimation

# ML examples

Consider sinusoidal signal

$$r(k) = A\cos(2\pi f_0 k + \theta) + w(k), \qquad k = 0,1,...,N-1$$

where w(k) is AWGN with known variance  $\sigma^2$ , and the sine amplitude A and frequency  $f_0$  are known as well.

• The LF is thus a function of the *unknown phase*  $\theta$ :

$$p(\mathbf{r};\theta) = \frac{1}{(2\pi\sigma^2)^{N/2}} \exp\left[\frac{-1}{2\sigma^2} \sum_{k=0}^{N-1} (r(k) - A\cos(2\pi f_0 k + \theta))^2\right]$$

#### ML examples...

• The log-likelihood function:

$$\Lambda(\mathbf{r};\theta) = C_1 - \frac{1}{2\sigma^2} \sum_{k=0}^{N-1} (r(k) - A\cos(2\pi f_0 k + \theta))^2$$

• Minimization: differentiate LLF w.r.t.  $\theta$ :

$$\frac{\partial \Lambda(\mathbf{r};\theta)}{\partial \theta} = \frac{-2}{2\sigma^2} \sum_{k=0}^{N-1} (r(k) - A\cos(2\pi f_0 k + \theta)) A\sin(2\pi f_0 k + \theta) = 0$$

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# ML examples...

This gives equation

$$\sum_{k=0}^{N-1} r(k) \sin(2\pi f_0 k + \hat{\theta}) = A \sum_{k=0}^{N-1} \sin(2\pi f_0 k + \hat{\theta}) \cos(2\pi f_0 k + \hat{\theta})$$

• Simplify:

$$\sum_{k=0}^{N-1} \sin(2\pi f_0 k + \hat{\theta}) \cos(2\pi f_0 k + \hat{\theta}) = \frac{1}{2} \sum_{k=0}^{N-1} \sin(4\pi f_0 k + 2\hat{\theta}) \approx 0$$

when N is large enough and  $f_0$  is not too close to 0 or 0.5.

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## ML examples...

• The approximate MLE can then be solved from

$$\sum_{k=0}^{N-1} r(k) \sin(2\pi f_0 k + \hat{\theta}) = 0$$

• Separate the phase:

$$\sum_{k=0}^{N-1} r(k) \sin(2\pi f_0 k) \cos \hat{\theta} = -\sum_{k=0}^{N-1} r(k) \cos(2\pi f_0 k) \sin \hat{\theta}$$

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# ML examples...

• The approximate MLE for the phase is then solved as:

$$\hat{\theta} = -\arctan\left(\frac{\sum_{k=0}^{N-1} r(k)\sin(2\pi f_0 k)}{\sum_{k=0}^{N-1} r(k)\cos(2\pi f_0 k)}\right)$$

• Closed-form (approximate) solution!

# IV. Cramer-Rao lower bound

#### Cramer-Rao lower bound

- How good is the MLE? How good can *any* estimate be?
- The goodness of any *unbiased* estimate can be measured by the *variance* of the estimate
- The variance depends on PDF (or LF) and its *sensitivity* to the parameter in question
- If the PDF depends only weakly (or not at all!) on the parameter, we cannot expect to get a good estimate from data samples by using any technique

#### Cramer-Rao lower bounds...

 In the DC-in-AWGN example we obtained the LLF derivative w.r.t. A as

$$\frac{\partial}{\partial A}\Lambda(\mathbf{r};A) = \frac{1}{\sigma^2} \sum_{k=0}^{N-1} (r(k) - A)$$

• By setting this to zero we got the averager as the MLE:

$$\hat{A} = \bar{r} = \frac{1}{N} \sum_{k=0}^{N-1} r(k)$$

• The *second* derivative of the LLF is

$$\frac{\partial^2}{\partial A^2} \Lambda(\mathbf{r}; A) = \frac{-N}{\sigma^2}$$

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#### Cramer-Rao lower bounds...

• The second derivative is the negative inverse of the estimator variance:

$$\operatorname{var}(\hat{A}) = \frac{\sigma^2}{N} = \frac{-1}{\frac{\partial^2}{\partial A^2} \Lambda(\mathbf{r}; A)}$$

• It can be shown that this is a lower bound for *all* estimators and *all* PDFs!

#### Cramer-Rao lower bound

• *Cramer-Rao Lower Bound (Scalar parameter):* For any unbiased estimator of parameter *A*, the estimator variance is lower bounded by

$$\operatorname{var}(\hat{A}) \ge \frac{-1}{\operatorname{E}\left[\frac{\partial^2}{\partial A^2} \Lambda(\mathbf{r}; A)\right]} = CRB$$

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#### Cramer-Rao lower bound...

• *Cramer-Rao Lower Bound (Vector parameter):* For any unbiased estimator of parameter vector **v**, the estimator variance is lower bounded by

$$\operatorname{var}(\hat{v}_i) \ge \left[\mathbf{I}^{-1}(\mathbf{v})\right]_{ii} = CRB$$

where

$$\left[\mathbf{I}(\mathbf{v})\right]_{ij} = -\mathbf{E}\left[\frac{\partial^2 \Lambda(\mathbf{r}; \mathbf{v})}{\partial v_i \partial v_j}\right]$$

• I(v) is the *Fisher information matrix* 

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#### CRB example: phase estimation

• Consider the CRB for the sine phase estimation problem:

$$r(k) = A\cos(2\pi f_0 k + \theta) + w(k), \qquad k = 0,1,...,N-1$$

• The LLF:

$$\Lambda(\mathbf{r};\theta) = C_1 - \frac{1}{2\sigma^2} \sum_{k=0}^{N-1} (r(k) - A\cos(2\pi f_0 k + \theta)^2)$$

• The first derivative of LLF:

$$\begin{split} \frac{\partial \Lambda(\mathbf{r};\theta)}{\partial \theta} &= \frac{-1}{\sigma^2} \sum_{k=0}^{N-1} \left( r(k) - A \cos(2\pi f_0 k + \theta) \right) A \sin(2\pi f_0 k + \theta) \\ &= \frac{-A}{\sigma^2} \sum_{k=0}^{N-1} \left[ r(k) \sin(2\pi f_0 k + \theta) - \frac{A}{2} \sin(4\pi f_0 k + 2\theta) \right] \end{split}$$

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#### CRB example: phase estimation...

• The second derivative of LLF:

$$\frac{\partial^2 \Lambda(\mathbf{r}; \theta)}{\partial \theta^2} = \frac{-A}{\sigma^2} \sum_{k=0}^{N-1} \left[ r(k) \cos(2\pi f_0 k + \theta) - A \cos(4\pi f_0 k + 2\theta) \right]$$

• Negative expected value:

$$-\mathbf{E}\left[\frac{\partial^{2} \Lambda(\mathbf{r};\theta)}{\partial \theta^{2}}\right] = \frac{A}{\sigma^{2}} \sum_{k=0}^{N-1} \left[A \cos^{2}(2\pi f_{0}k + \theta) - A \cos(4\pi f_{0}k + 2\theta)\right]$$
$$= \frac{A^{2}}{\sigma^{2}} \sum_{k=0}^{N-1} \left[\frac{1}{2} + \frac{1}{2} \cos(4\pi f_{0}k + 2\theta) - \cos(4\pi f_{0}k + 2\theta)\right]$$
$$\approx \frac{NA^{2}}{2\sigma^{2}}$$

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## CRB example: phase estimation...

• Hence, the Cramer-Rao bound is obtained as:

$$\operatorname{var}(\hat{\theta}) \ge CRB = \frac{-1}{\operatorname{E}\left[\frac{\partial^2}{\partial A^2} \Lambda(\mathbf{r}; \theta)\right]} = \frac{2\sigma^2}{NA^2} = \frac{1}{N \times SNR}$$

• Better phase estimates (= small variance) can be obtained with high SNR and by increasing the no. of samples

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#### **Modified CRB**

- The true Cramer-Rao bound is often difficult to compute
- Requires explicit elimination of nuisance parameters
- Alternative: Replace 2nd derivative with time-domain integral:

$$MCRB(v_i) = \frac{N_0 / 2}{E_u \left[ \int_0^{T_0} \left| \partial s(t, \mathbf{v}, \mathbf{u}) / \partial v_i \right|^2 dt \right]}$$

where s(t) is (passband) signal,  $\mathbf{v} = \text{sync}$  parameters,

 $\mathbf{u}$  = unwanted parameters

and  $v_i$  is the *i*th element in sync parameter vector  $\mathbf{v}$ .

• Always MCRB ≤ CRB ('looser bound')

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# **Summary**

Today we discussed

I What is synchronization

II ML estimation principles

III ML example: sinusoid phase estimation

IV Cramer-Rao lower bound

Next: Synchronization: overview

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