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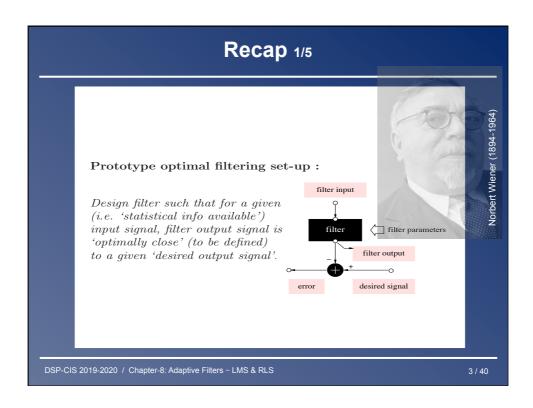
Part-III: Optimal & Adaptive Filters

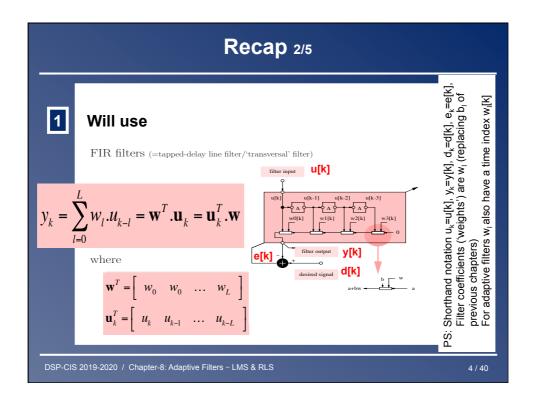
Chapter-8: Adaptive Filters - LMS & RLS

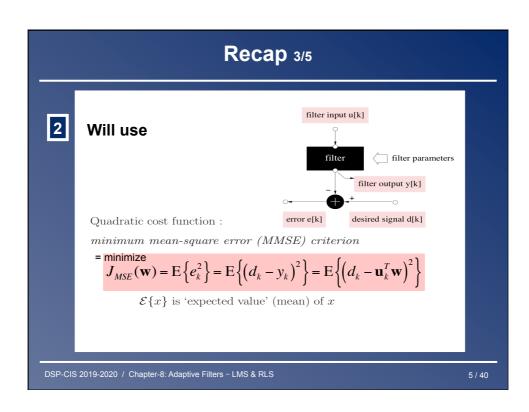
Marc Moonen

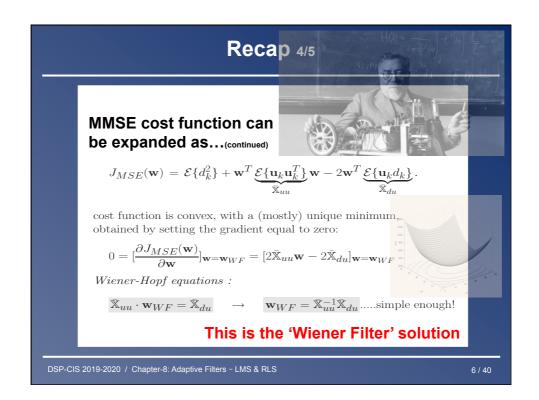
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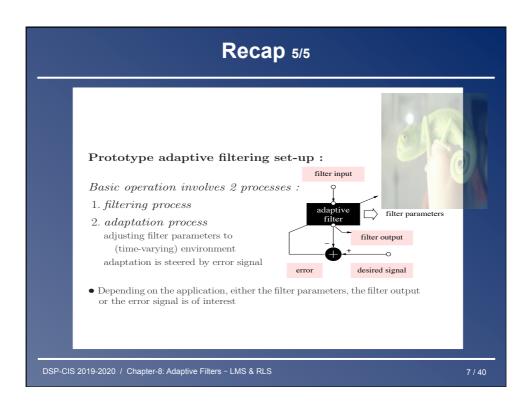
Part-III: Optimal & Adaptive Filters **Optimal Filters - Wiener Filters Chapter-7** • Introduction : General Set-Up & Applications Wiener Filters Adaptive Filters - LMS & RLS • Least Means Squares (LMS) Algorithm • Recursive Least Squares (RLS) Algorithm **Chapter-8 Square Root & Fast RLS Algorithms Chapter-9** · Square Root Algorithms Fast Algorithms **Chapter-10** Kalman Filters • Introduction - Least Squares Parameter Estimation · Standard Kalman Filter Square-Root Kalman Filter DSP-CIS 2019-2020 / Chapter-8: Adaptive Filters - LMS & RLS

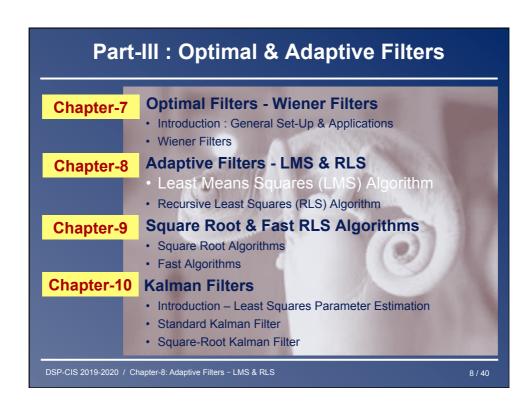


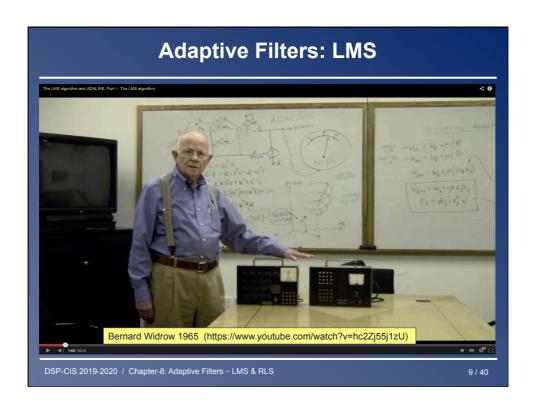












How do we compute the Wiener filter?

1) Cfr supra: By solving Wiener-Hopf equations (L+1 equations in L+1 unknowns)

$$\bar{\mathbb{X}}_{uu} \cdot \mathbf{w}_{WF} = \bar{\mathbb{X}}_{du}$$

2) Can also apply iterative procedure to minimize MMSE criterion, e.g.

Steepest-descent iterations:

$$\begin{aligned} \mathbf{w}(n+1) &= \mathbf{w}(n) + \frac{\mu}{2} \cdot \left[\frac{-\partial J_{MSE}(\mathbf{w})}{\partial \mathbf{w}} \right]_{\mathbf{w} = \mathbf{w}(n)} \\ &= \mathbf{w}(n) + \mu \cdot (\bar{\mathbb{X}}_{du} - \bar{\mathbb{X}}_{uu} \mathbf{w}(n)) \end{aligned}$$

here *n* is iteration index

μ is 'stepsize' (to be tuned..)

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Steepest-descent iterations:

$$\mathbf{w}(n+1) = \mathbf{w}(n) + \mu \cdot (\bar{\mathbb{X}}_{du} - \bar{\mathbb{X}}_{uu}\mathbf{w}(n))$$

$$\uparrow_{\bar{\mathbb{X}}_{uu} \cdot \mathbf{w}_{WF} = \bar{\mathbb{X}}_{du}}$$

Stability?

$$\begin{aligned} [\mathbf{w}(n+1) - \mathbf{w}_{WF}] &= (I - \mu \bar{\mathbb{X}}_{uu}) \cdot [\mathbf{w}(n) - \mathbf{w}_{WF}] \\ &= (I - \mu \bar{\mathbb{X}}_{uu})^{n+1} \cdot [\mathbf{w}(0) - \mathbf{w}_{WF}] \end{aligned}$$

stable iff $(\lambda_i = \text{eigenvalues of } \bar{\mathbb{X}}_{uu})$

$$-1 < 1 - \mu \lambda_i < 1 \quad \forall i$$

$$0 < \mu < \frac{2}{\lambda_{\text{max}}}$$

 $0 < \mu < \frac{2}{\lambda_{\max}}$ | \rightarrow large λ_{\max} implies a small stepsize

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Adaptive Filters: LMS

Convergence speed?

Transient behavior?

$$[\mathbf{w}(n+1) - \mathbf{w}_{WF}] = (I - \mu \bar{\mathbb{X}}_{uu})^{n+1} \cdot [\mathbf{w}(0) - \mathbf{w}_{WF}]$$

with (symmetric eigenvalue decomposition)

$$\bar{\mathbf{X}}_{uu} = Q_{uu} \Lambda_{uu} Q_{uu}^T \qquad Q_{uu}^T Q_{uu} = I$$

$$[\mathbf{w}(n+1) - \mathbf{w}_{WF}] = Q_{uu}(I - \mu \Lambda_{uu})^{n+1} Q_{uu}^T \cdot [\mathbf{w}(0) - \mathbf{w}_{WF}]$$

$$Q_{uu}^{T}[\mathbf{w}(n+1) - \mathbf{w}_{WF}] = \operatorname{diag}\{1 - \mu\lambda_{i}\}^{n+1}Q_{uu}^{T} \cdot [\mathbf{w}(0) - \mathbf{w}_{WF}]$$

error vector projected onto eigenvectors

i.e. $(1 - \mu \lambda_i)^n$ for 'mode' i (=projection on i-th eigenvector)

 \rightarrow small λ_i implies slow convergence (1- $\mu\lambda_i$ close to 1) for mode i

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Convergence speed?

Hence slowest convergence for $\lambda_i = \lambda_{min}$ With upper bound for μ (see p11):

$$1-2(\lambda_{min}/\lambda_{max}) \le 1-\mu\lambda_i \le 1$$

Hence λ_{min} << λ_{max} (i.e. large 'eigenvalue spread') implies **very** slow convergence

 λ_{min} << λ_{max} whenever input signal u[k] is very 'colored'

 $(\lambda_{min} = \lambda_{max}$ for 'white' input signal (i.e. autocorrelation matrix = I))

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Adaptive Filters: LMS

LMS is derived from WF steepest-descent iterations as follows Replace n+1 by n for convenience...

$$\mathbf{w}(n) = \mathbf{w}(n-1) + \mu.(\mathbf{E}\{\mathbf{u}_k.d_k\} - \mathbf{E}\{\mathbf{u}_k.\mathbf{u}_k^T\}.\mathbf{w}(n-1))$$

Then replace iteration index n by time index k (i.e. perform 1 iteration per sampling interval)

$$\mathbf{w}[k] = \mathbf{w}[k-1] + \mu.(\mathbf{E}\{\mathbf{u}_k.d_k\} - \mathbf{E}\{\mathbf{u}_k.\mathbf{u}_k^T\}.\mathbf{w}[k-1])$$

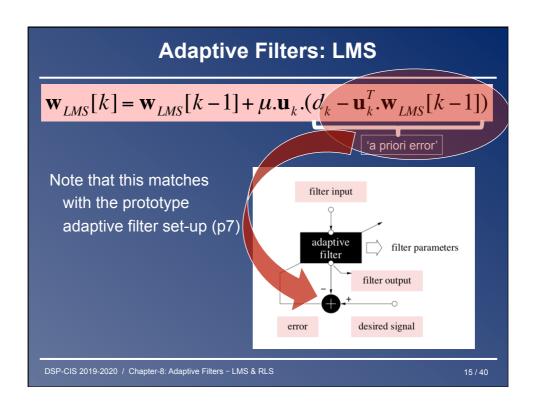
Then leave out expectation operators

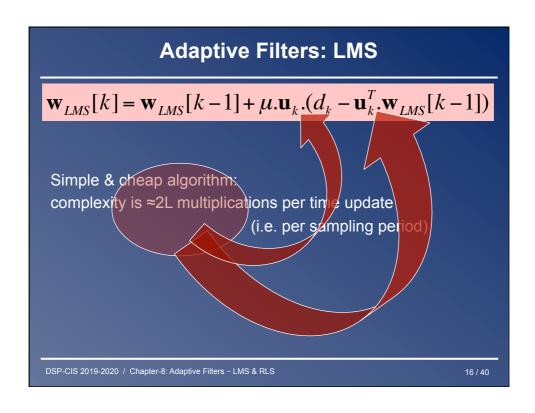
(i.e. replace expected values by instantaneous estimates)

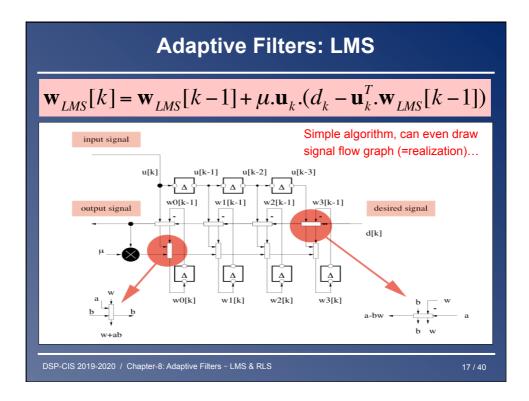
$$\mathbf{w}_{LMS}[k] = \mathbf{w}_{LMS}[k-1] + \mu \cdot \mathbf{u}_{k} \cdot (d_{k} - \mathbf{u}_{k}^{T} \cdot \mathbf{w}_{LMS}[k-1])$$

'a priori error'

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LMS analysis in a nutshell

LMS: stability/covergence? (proofs/details omitted)

- 'expected behavior'
 - = average over ∞ runs
 - = steepest-descent behavior

hence

$$0 < \mu < \frac{2}{\lambda_{\text{max}}}$$

• 'noisy gradients' (next page)

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LMS analysis in a nutshell

'Noisy gradients'

Whenever LMS has reached the WF solution the expected value of

$$\mathbf{u}_{k}.(d_{k} - \mathbf{u}_{k}^{T}.\mathbf{w}_{LMS}[k-1])$$
 (=estimated gradient in update formula)

but the instantaneous value

is generally non-zero (=noisy),

and hence LMS will again move away from the WF solution!

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Adaptive Filters: LMS

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LMS analysis in a nutshell

• 'noisy gradients' $\to J_{MSE}(\mathbf{w}[\infty]) > J_{MSE}(\mathbf{w}_{WF})$ results in excess MSE $J_{ex}(\infty)$ and mismatch \mathcal{M} :

$$J_{MSE}(\mathbf{w}[\infty]) = J_{MSE}(\mathbf{w}_{WF}) + \underbrace{J_{ex}(\mathbf{w}[\infty])}_{\approx J_{MSE}(\mathbf{w}_{WF}) \cdot \underbrace{\frac{\mu}{2} \sum_{i=0}^{L} \lambda_{i}}_{M}$$

PS: FIR case
$$\sum_{i=0}^{L} \lambda_i = \operatorname{trace}\{\bar{\mathbb{X}}_{uu}\} = L \bar{x}_{uu}(0) = L \mathcal{E}\{u_k^2\}$$

EX: for max 10% excess MSE: $\mu < \frac{0.2}{L.E\{u_k^2\}}$

means step size has to be much smaller...!

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LMS is an extremely popular algorithm many LMS-variants have been developed (cheaper/faster/...)...

• Normalized LMS (see p19-20)

$$\mathbf{w}_{NLMS}[k] = \mathbf{w}_{NLMS}[k-1] + \frac{\overline{\mu}}{\alpha + \mathbf{u}_{k}^{T} \cdot \mathbf{u}_{k}} \cdot \mathbf{u}_{k} \cdot (d_{k} - \mathbf{u}_{k}^{T} \cdot \mathbf{w}_{NLMS}[k-1])$$

- Transform domain LMS
- ullet Block LMS: K is block index, L_B is block size

$$\mathbf{w}_{BLMS}[K] = \mathbf{w}_{BLMS}[K-1] + \frac{\mu}{L} \cdot \sum_{i=1}^{L_B} \mathbf{u}_{(K-1), L_B+i} \cdot (d_{(K-1), L_B+i} - \mathbf{u}_{(K-1), L_B+i}^T \cdot \mathbf{w}_{BLMS}[K-1])$$

• Frequency domain LMS

(see Chapter-13

• Subband (LMS) adaptive filtering

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Adaptive Filters: LMS

normalized LMS (NLMS) = LMS with normalized step size (mostly used in practice)

$$\mathbf{w}_{NLMS}[k] = \mathbf{w}_{NLMS}[k-1] + \frac{\overline{\mu}}{\alpha + \mathbf{u}_{k}^{T}.\mathbf{u}_{k}}.\mathbf{u}_{k}.(d_{k} - \mathbf{u}_{k}^{T}.\mathbf{w}_{NLMS}[k-1])$$

Computational complexity is larger

≈3L instead of ≈2L multiplications per time update (except when $u_k^T u_k$ is computed recursively)

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normalized LMS (NLMS) = LMS with normalized step size (mostly used in practice)

$$\mathbf{w}_{NLMS}[k] = \mathbf{w}_{NLMS}[k-1] + \frac{\overline{\mu}}{\alpha + \mathbf{u}_{k}^{T} \cdot \mathbf{u}_{k}} \cdot \mathbf{u}_{k} \cdot (d_{k} - \mathbf{u}_{k}^{T} \cdot \mathbf{w}_{NLMS}[k-1])$$

Step size tuning for NLMS is much easier...

• stability/convergence ? : convergence if $0 < \bar{\mu} < 2$

max. 10% excess MSE obtained with $\bar{\mu} < 0.2$

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Adaptive Filters: LMS

normalized LMS (NLMS) = LMS with normalized step size (mostly used in practice)

$$\mathbf{w}_{NLMS}[k] = \mathbf{w}_{NLMS}[k-1] + \frac{\overline{\mu}}{\alpha + \mathbf{u}_{k}^{T} \cdot \mathbf{u}_{k}} \cdot \mathbf{u}_{k} \cdot (d_{k} - \mathbf{u}_{k}^{T} \cdot \mathbf{w}_{NLMS}[k-1])$$

• NLMS (for $\bar{\mu} = 1$) also solves a specific optimization problem:

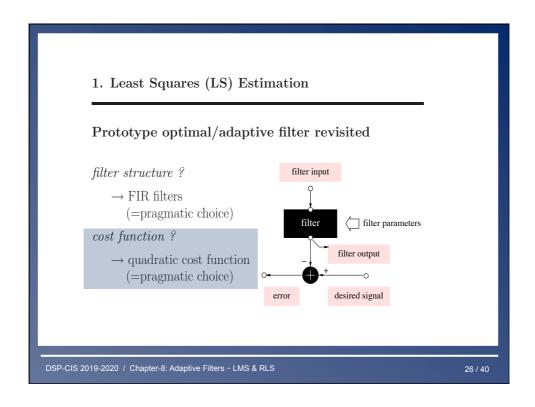
$$\min_{\mathbf{w}[k]} \tilde{J}(\mathbf{w}[k]) = \alpha \cdot \left\| \mathbf{w}[k] - \mathbf{w}_{NLMS}[k-1] \right\|_{2}^{2} + \left(d_{k} - \mathbf{u}_{k}^{T} \cdot \mathbf{w}[k] \right)^{2}$$

'a posteriori error'

For instance with (normalized step size=1 and) $\alpha \rightarrow 0$, the NLMS solution at time k sets the a posteriori error to zero, with minimal change with respect to previous NLMS solution at time k-1

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1. Least Squares (LS) Estimation

Quadratic cost function

MMSE:

$$J_{MSE}(\mathbf{w}) = \mathbf{E}\left\{e_k^2\right\} = \mathbf{E}\left\{\left(d_k - y_k\right)^2\right\} = \mathbf{E}\left\{\left(d_k - \mathbf{u}_k^T \mathbf{w}\right)^2\right\}$$

Least-squares(LS) criterion :

if statistical info is not available, may use an alternative 'data-based' criterion...

$$J_{LS}(\mathbf{w}) = \sum_{l=1}^{k} e_l^2 = \sum_{l=1}^{k} (d_l - y_l)^2 = \sum_{l=1}^{k} (d_l - \mathbf{u}_l^T \mathbf{w})^2$$

Interpretation? : see below

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1. Least Squares (LS) Estimation

filter input sequence : $\mathbf{u}_1,\mathbf{u}_2,\mathbf{u}_3,\dots$ \mathbf{u}_k corresponding desired response sequence is : d_1,d_2,d_3,\dots , d_k

$$\begin{bmatrix} e_1 \\ e_2 \\ \vdots \\ e_k \end{bmatrix} = \begin{bmatrix} d_1 \\ d_2 \\ \vdots \\ d_k \end{bmatrix} - \begin{bmatrix} \mathbf{u}_1^T \\ \mathbf{u}_2^T \\ \vdots \\ \mathbf{u}_k^T \end{bmatrix} \cdot \begin{bmatrix} w_0 \\ w_1 \\ \vdots \\ w_L \end{bmatrix}$$
error signal \mathbf{e}

$$\mathbf{d}$$

$$\mathbf{U}$$

cost function
$$J_{LS}(\mathbf{w}) = \sum_{l=1}^{k} e_l^2 = \|\mathbf{e}\|_2^2 = \|\mathbf{d} - U\mathbf{w}\|_2^2$$

 $\rightarrow linear\ least\ squares\ problem: \min_{\mathbf{w}} \|\mathbf{d} - U\mathbf{w}\|_2^2$

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1. Least Squares (LS) Estimation

$$J_{LS}(\mathbf{w}) = \sum_{l=1}^{k} e_l^2 = \|\mathbf{e}\|_2^2 = \mathbf{e}^T \cdot \mathbf{e} = \|\mathbf{d} - U\mathbf{w}\|_2^2$$

minimum obtained by setting gradient = 0:

$$0 = \left[\frac{\partial J_{LS}(\mathbf{w})}{\partial \mathbf{w}}\right]_{\mathbf{w} = \mathbf{w}_{LS}} = \left[\frac{\partial}{\partial \mathbf{w}} (\mathbf{d}^T \mathbf{d} + \mathbf{w}^T U^T U \mathbf{w} - 2\mathbf{w}^T U^T \mathbf{d})\right]_{\mathbf{w} = \mathbf{w}_{LS}}$$
$$= \left[2\underbrace{U^T U}_{\mathbb{X}_{uu}} \mathbf{w} - 2\underbrace{U^T \mathbf{d}}_{\mathbb{X}_{du}}\right]_{\mathbf{w} = \mathbf{w}_{LS}}$$

 $\mathbb{X}_{uu}\cdot\mathbf{w}_{LS}=\mathbb{X}_{du}$ \to $\mathbf{w}_{LS}=\mathbb{X}_{uu}^{-1}\mathbb{X}_{du}$ This is the 'Least Squares Solution'

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1. Least Squares (LS) Estimation

Note: correspondences with Wiener filter theory?

 \clubsuit estimate $\bar{\mathbb{X}}_{uu}$ and $\bar{\mathbb{X}}_{du}$ by time-averaging (ergodicity!)

estimate
$$\left\{ \overline{\aleph}_{uu} \right\} = \frac{1}{k} \cdot \sum_{l=1}^{k} \mathbf{u}_{l} \cdot \mathbf{u}_{l}^{T} = \frac{1}{k} \cdot U^{T}U = \frac{1}{k} \cdot \aleph_{uu}$$

estimate
$$\left\{ \overline{\aleph}_{du} \right\} = \frac{1}{k} \cdot \sum_{l=1}^{k} \mathbf{u}_{l} \cdot d_{l} = \frac{1}{k} \cdot U^{T} \mathbf{d} = \frac{1}{k} \cdot \aleph_{du}$$

leads to same optimal filter:

$$\text{estimate}\{\mathbf{w}_{WF}\} = (\tfrac{1}{k} : \mathbb{X}_{uu})^{-1} \cdot (\tfrac{1}{k} \mathbb{X}_{du}) = \mathbb{X}_{uu}^{-1} \cdot \mathbb{X}_{du} = \mathbf{w}_{LS}$$

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1. Least Squares (LS) Estimation

Note: correspondences with Wiener filter theory? (continued)

♣ Furthermore (for ergodic processes!) :

$$\overline{\aleph}_{uu} = \lim_{k \to \infty} \frac{1}{k} . \sum_{l=1}^{k} \mathbf{u}_{l} . \mathbf{u}_{l}^{T} = \lim_{k \to \infty} \frac{1}{k} . \aleph_{uu}$$

$$\overline{\aleph}_{du} = \lim_{k \to \infty} \frac{1}{k} \cdot \sum_{l=1}^{k} \mathbf{u}_{l} \cdot d_{l} = \lim_{k \to \infty} \frac{1}{k} \cdot \aleph_{du}$$

so that

$$\lim_{k\to\infty} \mathbf{w}_{LS} = \mathbf{w}_{WF}$$

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Least Squares (LS) Estimation

In words:

Whenever statistical info (autocorrelation and crosscorrelation) is missing, this can be estimated from observed data (assuming ergodicity)

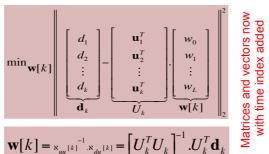
The Wiener filter solution, with true statistical quantities replaced by estimated quantities, then turns out to be the same as the LS solution

LS approach in itself optimizes a different (LS) criterion, without any need for statistical assumptions (e.g. ergodicity..)

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2. Recursive Least Squares (RLS)

For a fixed data segment 1... k least squares problem is



 $\mathbf{w}[k] = \mathbf{w}_{uu^{[k]}}^{-1} \cdot \mathbf{w}_{du^{[k]}} = \left[U_k^T U_k\right]^{-1} \cdot U_k^T \mathbf{d}_k$

Wanted: recursive/adaptive algorithms

Can LS solution @ time k be computed from solution @ time k-1?

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2.1 Standard RLS

It is observed that $\aleph_{uu}[k] = \aleph_{uu}[k-1] + \mathbf{u}_k \cdot \mathbf{u}_k^T$ (and $\aleph_{du}[k] = \aleph_{du}[k-1] + \mathbf{u}_k \cdot d_k$)

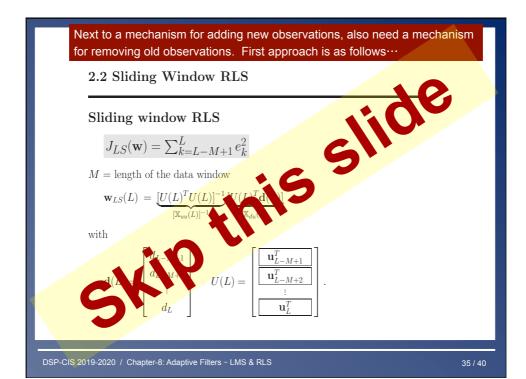
The matrix inversion lemma states that (check 'matrix inversion lemma' in Wikipedia)

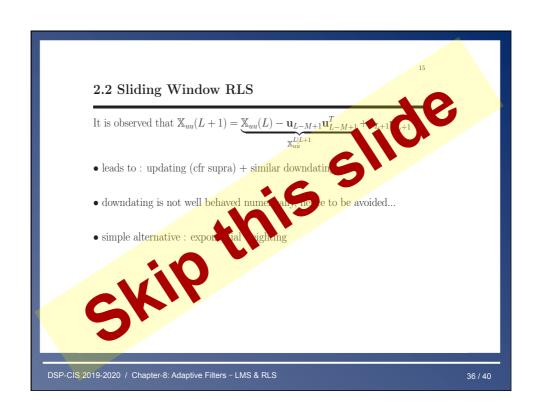
$$\aleph_{uu}[k]^{-1} = \aleph_{uu}[k-1]^{-1} - (\frac{1}{1 + \mathbf{u}_{k}^{T} \aleph_{uu}[k-1]^{-1} \mathbf{u}_{k}}) \cdot \mathbf{k}_{k} \mathbf{k}_{k}^{T} \quad \text{with} \quad \mathbf{k}_{k} = \aleph_{uu}[k-1]^{-1} \mathbf{u}_{k}^{T}$$

With this it is proved that: $\mathbf{w}_{LS}[k] = \mathbf{w}_{LS}[k-1]^{-1} - (\frac{1}{1 + \mathbf{u}_k^T \aleph_{uu}[k-1]^{-1} \mathbf{u}_k}) \cdot \mathbf{k}_k \mathbf{k}_k^T \quad \text{with} \quad \mathbf{k}_k = \aleph_{uu}[k-1]^{-1} \mathbf{u}_k$ $\mathbf{w}_{LS}[k] = \mathbf{w}_{LS}[k-1] + \underbrace{\frac{1}{\aleph_{uu}[k-1]^{-1} \mathbf{u}_k}}_{-(\frac{1}{1 + \mathbf{u}_k^T \aleph_{uu}[k-1]^{-1} \mathbf{u}_k}) \cdot \mathbf{k}_k}. \quad \text{'a priori residual'}$

= standard recursive least squares (RLS) algorithm

Remark: $O(L^2)$ instead of $O(L^3)$ operations per time update





Next to a mechanism for adding new observations, also need a mechanism for removing old observations. Simpler approach is as follows…

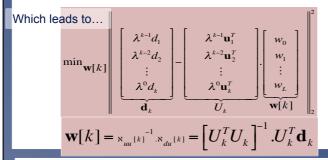
2.3 Exponentially Weighted RLS

Exponentially weighted RLS: Goal is to give a smaller weight to 'older' data, i.e.

$$J_{LS}(\mathbf{w}) = \sum_{l=1}^{k} \lambda^{2(k-l)} e_l^2$$

 $0 < \lambda < 1$ is weighting factor or forget factor

 $\frac{1}{1-\lambda}$ is a 'measure of the memory of the algorithm'



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2.3 Exponentially Weighted RLS

It is observed that $\aleph_{uu}[k] = \lambda^2.\aleph_{uu}[k-1] + \mathbf{u}_k.\mathbf{u}_k^T$ (and $\aleph_{du}[k] = \lambda^2.\aleph_{du}[k-1] + \mathbf{u}_k.d_k$)

hence

$$\aleph_{uu}[k]^{-1} = \frac{1}{\lambda^{2}} \aleph_{uu}[k-1]^{-1} - (\frac{1}{1 + \frac{1}{\lambda^{2}} \mathbf{u}_{k}^{T} \aleph_{uu}[k-1]^{-1} \mathbf{u}_{k}}) \cdot \mathbf{k}_{k} \mathbf{k}_{k}^{T} \quad \text{with} \quad \mathbf{k}_{k} = \frac{1}{\lambda^{2}} \aleph_{uu}[k-1]^{-1} \mathbf{u}_{k}$$

$$\mathbf{w}_{LS}[k] = \mathbf{w}_{LS}[k-1] + \aleph_{uu}[k]^{-1} \mathbf{u}_{k} \cdot (d_k - \mathbf{u}_k^T \mathbf{w}_{LS}[k-1])$$

i.e. exponential weighting hardly changes RLS formulas.. (easy!)

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Recursive Least Squares (RLS) Algorithm

Computational Complexity:

Standard RLS algorithm (with exponential weighting) has **O(L**²) computational complexity per time update

Compare to **O(L)** for LMS (=cheaper, but slow convergence)

In Chapter-9, will present 'Fast RLS' algorithms with **O(L)** computational complexity (and without compromising convergence properties)

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Recursive Least Squares (RLS) Algorithm

Numerical Analysis/Stability:

Standard RLS algorithm (even with exponential weighting) has been shown to have unstable quantization error propagation (in low-precision implementation)

In Chapter-9, will present 'Square Root RLS' algorithms which are shown to be perfectly stable numerically (without compromising complexity & convergence properties)

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