

## 2 Adaptive FIR filters

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### Some algorithms and their limitations

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- Wiener filtering.
- Stationary case description (steepest descent, quasi-Newton).
- Traditional updating schemes: LMS, RLS, QR.
- Convergence in the mean and mean square error variance.
- Convergence speed - correlation matrix conditioning trade off.
- Different realizations.

In a general framework, the *Mean Squared Error* (MSE)  $E\{e^2(n)\}$ , has the following quadratic form:

$$E\{e^2(n)\} = \rho - 2 \boldsymbol{\theta}^T \mathbf{p} + \boldsymbol{\theta}^T \mathbf{R}_x \boldsymbol{\theta}$$

where  $\mathbf{R}_x > 0$  and  $\mathbf{p}$  and  $\rho$  are assumed to be known in an ideal setting or, from a practical implementation point of view, some suitable estimates are at hand.

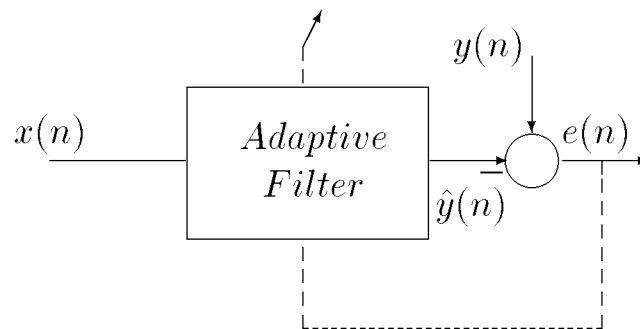
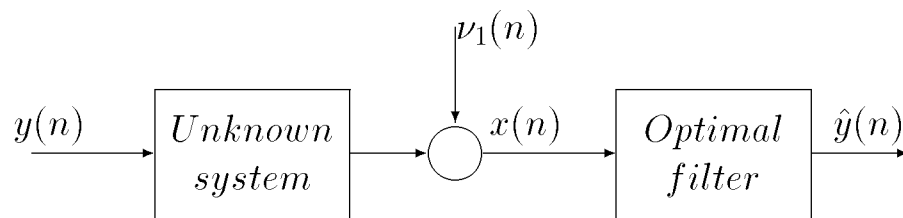


Figure 9: Adaptive filtering general framework.

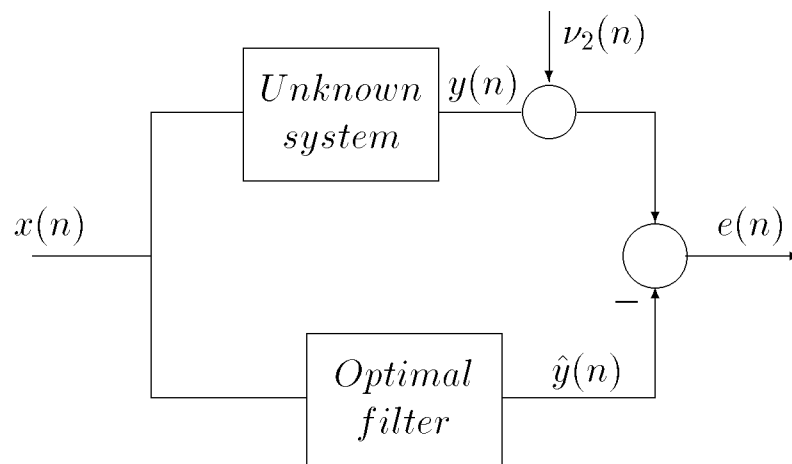
## 2.1 Wiener Filtering

### 2.1.1 Optimal filtering

- **Problem 1: Inverse filtering:** To design  $H(z)$ , the input (observable signal)  $x(n)$  has noise and the reference is not available. The idea is to design  $H(z)$  so that  $\hat{y}(n) = H(z)x(n)$  approximates  $y(n)$ .
- **Problem 2: Direct filtering or modeling:**  $y(n)$  is the not observable output of the filter to design  $H(z)$ . The idea is to design  $H(z)$  so that  $\hat{y}(n) = H(z)x(n)$  approximates  $y(n)$ .



a)



b)

Figure 10: a) Inverse filtering and b) direct filtering or modeling

- Assumption:  $x(n)$  and  $y(n)$  jointly wide sense stationary and have zero mean and are uncorrelated with the disturbance.
- The degree of approximation is measured by the *Mean squared error*,

$$E\{e^2(n)\} = E\{(y(n) - \hat{y}(n))^2\} \quad (3)$$

- Second order statistics known, i.e.,

$$\begin{aligned} p(k) &= E\{y(n-k)x(n)\} \\ r(k) &= E\{x(n-k)x(n)\} \\ \rho &= E\{y^2(n)\} \end{aligned}$$

### 2.1.2 The inverse filtering problem

$x(n) = s(n) + \nu_1(n)$ ,  $y(n) = s(n)$  ( $s(n)$  recoverable signal). Since  $\hat{y}(n) = \sum_k h(k)x(n-k)$ , three cases:

- Non causal case:  $x(n-k)$  known for all  $k$ , no constraints on  $h(n)$ , i.e.,  $\|h(n)\|^2 < \infty$ .
- Causal FIR case:  $x(n-k)$  known for  $0 \leq k \leq n$ , so  $h(k) = 0$  for  $k < 0$  and  $k > n$ .
- Causal IIR case:  $x(n-k)$  known for  $k \geq 0$ , so  $h(k) = 0$  for  $k < 0$  and  $\|h(n)\|^2 < \infty$ .

$$\begin{aligned} E[e^2(n)] &= E[y^2(n)] - 2 \sum_k h(k) E[y(n)x(n-k)] \\ &\quad + \sum_k \sum_l h(l)h(k) E[x(n-k)x(n-l)] \\ &= \rho - 2 \sum_k h(k)p(k) + \sum_k \sum_l h(k)h(l)r(l-k) \end{aligned}$$

OR

$$\begin{aligned} r_e(k) &= r_y(k) - r_{y\hat{y}}(k) - r_{\hat{y}y}(k) + r_{\hat{y}\hat{y}}(k) \\ &= \rho(k) - \sum_l [h(l)p(l+k) + p(l)h(l+k)] \\ &\quad + \sum_l \sum_j h(l)r(l+k-j)h(j) \\ S_e(e^{jw}) &= S_y(e^{jw}) - S_{yx}(e^{jw})H^*(e^{jw}) - S_{yx}^*(e^{jw})H(e^{jw}) + S_x(e^{jw})|H^*(e^{jw})|^2 \\ &= \left| H(e^{jw}) - \frac{S_{yx}(e^{jw})}{S_x(e^{jw})} \right|^2 S_x(e^{jw}) + \left[ S_y(e^{jw}) - \frac{|S_{yx}(e^{jw})|^2}{S_x(e^{jw})} \right] \end{aligned}$$

- **the non causal case:**  $S_{yx}(e^{j\omega}) = S_s(e^{j\omega})$  and  $S_x(e^{j\omega}) = S_s(e^{j\omega}) + S_\nu(e^{j\omega})$ , i.e.,

$$H(e^{j\omega}) = \frac{S_{y\nu}(e^{j\omega})}{S_x(e^{j\omega})} = \frac{S_s(e^{j\omega})}{S_s(e^{j\omega}) + S_\nu(e^{j\omega})}$$

- **the causal FIR case:**

$$\begin{aligned} E[e^2(n)] &= E[y^2(n)] - 2 \sum_{k=0}^N h(k) E[y(n)x(n-k)] \\ &\quad + \sum_{k=0}^N \sum_{l=0}^N h(l)h(k) E[x(n-k)x(n-l)] \\ &= \rho - 2 \sum_{k=0}^N h(k)p(k) + \sum_{k=0}^N \sum_{l=0}^N h(k)h(l)r(l-k) \end{aligned}$$

that is minimized for

$$\begin{aligned} p(k) &= \sum_{n=0}^N r(k-n)h(n), \quad \text{or} \\ 0 &= E[e(n)x(n-k)], \quad \text{for } 0 \leq k \leq N \end{aligned}$$

Let consider two cases:

- Filtering (basic equalization): if  $x(n) = s(n) + \nu(n)$  and  $y(n) = s(n - N)$ , then ( $r(n) = r_x(n) + r_\nu(n)$  and  $p(n) = r_s(n - N)$ ):

$$\mathbf{R}_s + \mathbf{R}_\nu \begin{bmatrix} h(0) \\ \vdots \\ h(N) \\ \vdots \\ h(2N) \end{bmatrix} = \mathbf{R}_s \begin{bmatrix} 0 \\ \vdots \\ 1 \\ \vdots \\ 0 \end{bmatrix}$$

whose solution is a linear phase FIR filter ( $h(n) = h(2N - n)$ ,  $n = 0, 1, \dots, 2N$ ).

- Prediction:

\* Forward: if  $\hat{y}(n) = \sum_{k=1}^N h_k x(n - k)$  and  $y(n) = x(n)$ , then:

$$\begin{bmatrix} r(1) & r(2) & \cdots & r(N) \\ r(2) & r(1) & \cdots & r(N-1) \\ & & \vdots & \\ r(N) & r(N-1) & \cdots & r(1) \end{bmatrix} \begin{bmatrix} h(1) \\ h(2) \\ \vdots \\ h(N) \end{bmatrix} = \begin{bmatrix} r(1) \\ r(2) \\ \vdots \\ r(N) \end{bmatrix}$$

\* Backward: if  $\hat{y}(n - N) = \sum_{k=1}^N g_k x(n - k + 1)$  and  $y(n) = x(n - M)$ , then:

$$\begin{bmatrix} r(1) & r(2) & \cdots & r(N) \\ r(2) & r(1) & \cdots & r(N-1) \\ & & \vdots & \\ r(N) & r(N-1) & \cdots & r(1) \end{bmatrix} \begin{bmatrix} g(1) \\ g(2) \\ \vdots \\ g(N) \end{bmatrix} = \begin{bmatrix} r(N) \\ r(N-1) \\ \vdots \\ r(1) \end{bmatrix}$$

The solutions are related by:

$$g_k = h_{N-k} \quad \text{for } k = 1, \dots, N - 1$$

- **the causal IIR case:** Here, in a similar form that for the FIR case, except for  $N \rightarrow \infty$ ,

$$p(k) = \sum_{n=0}^{\infty} r(k-n)h(n),$$

is the optimal condition, for  $0 \leq k < \infty$ . A frequency domain solution is obtained if

- $x(n) = G(z)u(n)$ ,  $u(n)$  white noise, i.e.,  $S_u(e^{jw}) = 1$  and  $G(z)$  is invertible.
- $y(n) = F(z)u(n) + \nu(n)$ ,  $\nu(n)$  colored noise uncorrelated with  $u(n)$ , i.e.,  $S_{u\nu}(e^{jw}) = 0$ .

then with  $S_{yx}(e^{jw})$  and  $S_y(e^{jw})$  respectively,

$$\begin{aligned} F(e^{jw}) &= \frac{S_{yx}(e^{jw})}{G(e^{jw})^*} \\ S_{\nu}(e^{jw}) &= S_y(e^{jw}) - |F(e^{jw})|^2 = S_y(e^{jw}) - \frac{|S_{yx}(e^{jw})|^2}{S_x(e^{jw})} \\ S_e(e^{jw}) &= |F(e^{jw}) - G(e^{jw})H(e^{jw})|^2 + S_{\nu}(e^{jw}) \end{aligned}$$

and  $H(z) = H_0(z)/G(z) = F_+(z)/G(z)$ , where  $F_+(z)$  is the causal part of  $F(z)$ .



### 2.1.3 The direct filtering or modeling problem

Here  $y(n) = H(z)x(n) + \nu_2(n)$ ,  $\hat{y}(n) = \hat{H}(z)x(n)$ .

- In the FIR case,

$$\begin{aligned}\hat{y}(n) &= b_0x(n) + b_1x(n-1) + \dots + b_Nx(n-N) \\ &= \boldsymbol{\theta}^T \mathbf{x}(n)\end{aligned}$$

where  $\boldsymbol{\theta} = [b_0 \dots b_N]^T$  and  $\mathbf{x}(n) = [x(n) \dots x(n-N)]^T$ .

Then in an stationary environment,

$$E\{e^2(n)\} = \rho - 2 \boldsymbol{\theta}^T \mathbf{p} + \boldsymbol{\theta}^T \mathbf{R}_x \boldsymbol{\theta}$$

where  $\mathbf{R}_x = E\{\mathbf{x}(n)\mathbf{x}^T(n)\}$  and  $\mathbf{p} = E\{\mathbf{x}(n)y(n)\}$  are known.

A quadratic function of  $\boldsymbol{\theta}(n)$  with

$$\begin{aligned}\nabla &= \frac{\partial E\{e^2(n)\}}{\partial \boldsymbol{\theta}} = \left[ \frac{\partial E\{e^2(n)\}}{\partial b_0} \quad \frac{\partial E\{e^2(n)\}}{\partial b_1} \quad \dots \quad \frac{\partial E\{e^2(n)\}}{\partial b_N} \right]^T \\ &= -2 \mathbf{p} + 2 \mathbf{R}_x \boldsymbol{\theta} = 0\end{aligned}$$

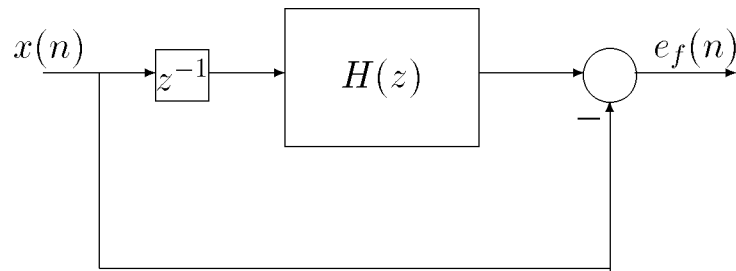
then the MSE is minimized when

$$\boldsymbol{\theta}_o = \mathbf{R}_x^{-1} \mathbf{p}$$

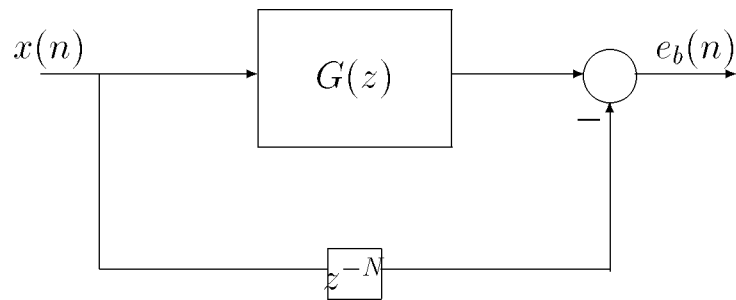
Note also that

$$\begin{aligned}\nabla &= \frac{\partial E\{e^2(n)\}}{\partial \boldsymbol{\theta}} = E\left[2 e(n) \frac{\partial e(n)}{\partial \boldsymbol{\theta}}\right] \\ &= -2E[e(n)\mathbf{x}(n)] = 0\end{aligned}$$

i.e., the *normal equation*.



a)



b)

Figure 11: Relationship between prediction and whitening filtering. a) forward predictor and whitening filter, b) backward predictor and whitening filter.

- Whitening a forward prediction filter: with  $e_f(n) = y(n) + \sum_{k=1}^N a_k y(n-k)$ , find  $A(z)$ , constrained to be a monic FIR filter ( $a(0) = 1$ ).

$$\begin{bmatrix} r(0) & r(1) & \cdots & r(N) \\ r(1) & r(0) & \cdots & r(N-1) \\ & & \vdots & \\ r(N) & r(N-1) & \cdots & r(0) \end{bmatrix} \begin{bmatrix} 1 \\ a(1) \\ \vdots \\ a(N) \end{bmatrix} = \begin{bmatrix} \alpha \\ 0 \\ \vdots \\ 0 \end{bmatrix}$$

The  $N$ -order (forward) prediction filter and the  $N+1$ -order whitening filter are related by  $A_{N+1}(z) = 1 - z^{-1}H_N(z)$ .

- Whitening a backward prediction filter: with  $e_b(n) = y(n-N) + \sum_{k=1}^N b_k y(n-k+1)$ , find  $B(z)$ , constrained to be a monic FIR filter ( $b(0) = 1$ ). Then

$$\begin{bmatrix} r(0) & r(1) & \cdots & r(N) \\ r(1) & r(0) & \cdots & r(N-1) \\ & & \vdots & \\ r(N) & r(N-1) & \cdots & r(0) \end{bmatrix} \begin{bmatrix} 1 \\ b(1) \\ \vdots \\ b(N) \end{bmatrix} = \begin{bmatrix} \beta \\ 0 \\ \vdots \\ 0 \end{bmatrix}$$

An important property: the outputs collection of whitening backward filters of orders 1 to  $N$  are orthogonal, i.e.,

$$E\{e_b^k e_b^m\} = \begin{cases} \beta_k & \text{if } m = k \\ 0 & \text{if } m \neq k \end{cases}$$

where  $\beta_k = E\{(e_b^k)^2\}$ .

This property can be used to obtain an useful decomposition (low Cholesky in this case) of the correlation matrix  $\mathbf{R}$ ,

$$\begin{aligned} \mathbf{D}_L &= \text{diag}[\beta_0, \dots, \beta_N] \\ &= E\{\mathbf{e}_b^N \mathbf{e}_b^N\} = E\{\mathbf{L}\mathbf{x}\mathbf{x}^T \mathbf{L}^T\} \\ \mathbf{L} &= \begin{bmatrix} 1 & 0 & 0 & 0 \\ b_N(1) & 1 & \cdots & \\ & & \vdots & \\ b_N(N) & \cdots & b_1(1) & 1 \end{bmatrix} \\ \mathbf{R}^{-1} &= \mathbf{L}\mathbf{D}_L^{-1}\mathbf{L}^T \end{aligned}$$

A similar factorization can be obtained but related to the whitening forward filtering, i.e., (upper Cholesky)

$$\begin{aligned} \mathbf{R}^{-1} &= \mathbf{U}\mathbf{D}_U^{-1}\mathbf{U}^T \\ \mathbf{U} &= \begin{bmatrix} 1 & a_1(1) & \cdots & a_N(N) \\ 0 & 1 & \cdots & a_N(N-1) \\ & & \vdots & a_N(1) \\ 0 & 0 & \cdots & 1 \end{bmatrix} \\ \mathbf{D}_U &= \text{diag}[\alpha_0, \dots, \alpha_N] \end{aligned}$$

where can be shown that  $\alpha_k = \beta_k$ .

## 2.2 Optimization in the ideal setting

### 2.2.1 Newton algorithm

$$\begin{aligned}\boldsymbol{\theta}(n+1) &= \boldsymbol{\theta}(n) - \mu \mathbf{R}_x^{-1} \nabla(n) \\ &= \boldsymbol{\theta}(n) + \mu \mathbf{R}_x^{-1} (-2\mathbf{p} + 2\mathbf{R}_x \boldsymbol{\theta}(n)) = (\mathbf{I} - 2\mu \mathbf{I}) \boldsymbol{\theta}(n) + 2\mu \boldsymbol{\theta}_o\end{aligned}$$

if  $\mu = 1/2$  the Wiener solution is reached in one step!.

### 2.2.2 Steepest Descent algorithm

Using  $\nabla(n) = 2(\mathbf{R}_x \boldsymbol{\theta}(n) - \mathbf{p})$ , then

$$\begin{aligned}\boldsymbol{\theta}(n+1) &= \boldsymbol{\theta}(n) - \mu \nabla(n) \\ &= \boldsymbol{\theta}(n) + 2\mu \mathbf{p} - 2\mu \mathbf{R}_x \boldsymbol{\theta}(n)\end{aligned}$$

With  $\tilde{\boldsymbol{\theta}}(n) = \boldsymbol{\theta}(n) - \boldsymbol{\theta}_o$ ,

$$\begin{aligned}\tilde{\boldsymbol{\theta}}(n+1) &= (\mathbf{I} - 2\mu \mathbf{R}_x) \tilde{\boldsymbol{\theta}}(n) \\ &= (\mathbf{I} - 2\mu \mathbf{R}_x)^{n+1} \tilde{\boldsymbol{\theta}}(0)\end{aligned}$$

Since  $\mathbf{R}_x > 0$ ,  $\mathbf{R}_x = \mathbf{Q} \boldsymbol{\Lambda} \mathbf{Q}^T$ , where  $\mathbf{Q}$  is an orthogonal and  $\boldsymbol{\Lambda}$  is the diagonal eigenvalue matrix. Then with  $\boldsymbol{\vartheta}(n) = \mathbf{Q}^T \tilde{\boldsymbol{\theta}}(n)$ ,

$$\begin{aligned}E \{ \tilde{\boldsymbol{\vartheta}}(n+1) \} &= [\mathbf{I} - \mu \boldsymbol{\Lambda}] E \{ \tilde{\boldsymbol{\vartheta}}(n) \} \\ &= [\mathbf{I} - \mu \boldsymbol{\Lambda}]^{n+1} \{ \tilde{\boldsymbol{\vartheta}}(0) \} \\ &= \begin{bmatrix} (1 - 2\mu \lambda_0)^{n+1} & 0 & \cdots & 0 \\ 0 & (1 - 2\mu \lambda_1)^{n+1} & & \vdots \\ \vdots & \vdots & \ddots & \vdots \\ 0 & 0 & & (1 - 2\mu \lambda_N)^{n+1} \end{bmatrix} \tilde{\boldsymbol{\vartheta}}(0)\end{aligned}$$

Then, SD algorithm converges to the Wiener solution if, for  $n \rightarrow \infty$ ,  $\mu$  satisfy

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

where  $\lambda_{max}$  is the maximum eigenvalue of  $\mathbf{R}_x$ .

### 2.2.3 Conjugate direction algorithm

Consider  $\mathbf{d}_j$ ,  $j = 0, \dots, N$ , that verify

$$\mathbf{d}_j^T \mathbf{R}_x \mathbf{d}_k = 0 \quad \text{if } j \neq k$$

as *conjugate directions*. Then:

Theorem: The sequence

$$\boldsymbol{\theta}(n+1) = \boldsymbol{\theta}(n) + \gamma_n \mathbf{d}_n$$

where  $\gamma_n = -(\mathbf{d}_n^T \mathbf{R}_x \mathbf{d}_n)^{-1} \mathbf{d}_n^T \nabla(n)$  and  $\nabla(n) = 2(\mathbf{R}_x \boldsymbol{\theta}(n) - \mathbf{p})$  converges to  $\boldsymbol{\theta}_o = \mathbf{R}_x^{-1} \mathbf{p}$  after  $N + 1$  steps, i.e.,  $\boldsymbol{\theta}(n) = \boldsymbol{\theta}_o$ .

Assuming to minimize the MSE with  $\boldsymbol{\theta}(n)$  constrained in  $\boldsymbol{\theta}(0) + \mathbf{D}_N$ , where  $\mathbf{D}_N = [\mathbf{d}_0 \mathbf{d}_1 \dots \mathbf{d}_{N-1}]$ . Then  $(\boldsymbol{\theta}(n) - \boldsymbol{\theta}_o)$  is given by the  $\mathbf{R}_x$ -orthogonal projection of  $(\boldsymbol{\theta}_o - \boldsymbol{\theta}(0))$  onto  $\mathbf{D}_N$ , i.e.,

$$\begin{aligned} \boldsymbol{\theta}(n) &= \boldsymbol{\theta}(0) + \mathbf{D}_N (\mathbf{D}_N^T \mathbf{R}_x \mathbf{D}_N)^{-1} \mathbf{D}_N^T \mathbf{R}_x (\boldsymbol{\theta}_o - \boldsymbol{\theta}(0)) \\ &= \boldsymbol{\theta}(0) + \sum_{k=0}^{N-1} \mathbf{d}_k (\mathbf{d}_k^T \mathbf{R}_x \mathbf{d}_k)^{-1} \mathbf{d}_k^T \mathbf{R}_x (\boldsymbol{\theta}_o - \boldsymbol{\theta}(0)) \end{aligned}$$

But  $\mathbf{d}_k^T \mathbf{R}_x \boldsymbol{\theta}_o = \mathbf{d}_k^T \mathbf{R}_x \boldsymbol{\theta}(k)$  so that with  $\mathbf{p} = \mathbf{R}_x \boldsymbol{\theta}_o$  is possible to verify

$$\mathbf{d}_k^T \mathbf{R}_x (\boldsymbol{\theta}_o - \boldsymbol{\theta}_k) = -\mathbf{d}_k^T \nabla(k)$$

that serves to justify the gain  $\gamma_k$ .

## 2.3 Updating algorithms

Two important properties related to estimation (updating) algorithms are in order:

- An estimate is *unbiased* if  $E\{\boldsymbol{\theta}(n)\} = \boldsymbol{\theta}_o$ .
- An estimate is *consistent* if  $\boldsymbol{\theta}(n) \rightarrow \boldsymbol{\theta}_o$  as  $n \rightarrow \infty$ .

Since second order statistics are not usually available, some simplifications in the ideal method are necessary.

### 2.3.1 The Least-Mean-Square (LMS) algorithm

When the gradient  $\nabla$  is not available, a suitable estimate is  $\nabla \approx -2e(n)\mathbf{x}(n)$ , the LMS algorithm

$$\boldsymbol{\theta}(n+1) = \boldsymbol{\theta}(n) + \mu \mathbf{x}(n)e(n) \quad (4)$$

where  $\mu > 0$ . As can be expected by the analysis of the ideal SD algorithm, this parameter is related to convergence speed and stability of the algorithm.

Some useful variants

$$\begin{aligned} \boldsymbol{\theta}(n+1) &= \boldsymbol{\theta}(n) + \mu \mathbf{x}(n) \operatorname{sgn}[e(n)] && \textit{Sign Error} \\ \boldsymbol{\theta}(n+1) &= \boldsymbol{\theta}(n) + \mu \operatorname{sgn}[\mathbf{x}(n)] e(n) && \textit{Sign Data} \\ \boldsymbol{\theta}(n+1) &= \boldsymbol{\theta}(n) + \mu \operatorname{sgn}[\mathbf{x}(n)] \operatorname{sgn}[e(n)] && \textit{Sign Sign} \end{aligned}$$

### 2.3.2 Convergence in the Mean and Error variance of the LMS

Using some simplifactory hypotesis, and by defining  $\tilde{\boldsymbol{\theta}}(n) = \boldsymbol{\theta}(n) - \boldsymbol{\theta}_o$ , and rewritten (4) as follows

$$\tilde{\boldsymbol{\theta}}(n+1) = [\mathbf{I} - \mu \mathbf{x}(n) \mathbf{x}^T(n)] \tilde{\boldsymbol{\theta}}(n) + \mu \mathbf{x}(n)(y(n) - \mathbf{x}^T(n) \boldsymbol{\theta}_o) \quad (5)$$

then

$$E \{ \tilde{\boldsymbol{\theta}}(n+1) \} = E \{ [\mathbf{I} - \mu \mathbf{x}(n) \mathbf{x}^T(n)] \tilde{\boldsymbol{\theta}}(n) \} + \mu E \{ \mathbf{x}(n)(y(n) - \mathbf{x}^T(n) \boldsymbol{\theta}_o) \} \quad (6)$$

Using the hypotesis

$$E \{ \tilde{\boldsymbol{\theta}}(n+1) \} = [\mathbf{I} - \mu \mathbf{R}_x] E \{ \tilde{\boldsymbol{\theta}}(n) \} \quad (7)$$

Using  $\mathbf{R}_x = \mathbf{Q} \boldsymbol{\Lambda} \mathbf{Q}^T$  (Cholesky) and pre-multiplying (7) by  $\mathbf{Q}^T$  and defining

$$\boldsymbol{\vartheta}(n) = \mathbf{Q}^T \tilde{\boldsymbol{\theta}}(n) \quad (8)$$

is possible to obtain

$$\begin{aligned} E \{ \tilde{\boldsymbol{\vartheta}}(n+1) \} &= [\mathbf{I} - \mu \boldsymbol{\Lambda}] E \{ \tilde{\boldsymbol{\vartheta}}(n) \} \\ &= [\mathbf{I} - \mu \boldsymbol{\Lambda}]^{n+1} E \{ \tilde{\boldsymbol{\vartheta}}(0) \} \end{aligned} \quad (9)$$

Then, in order that  $\boldsymbol{\theta}(n)$  converge in the mean to the Wiener solution

$$0 < \mu < \frac{1}{\lambda_{max}} \quad (10)$$

Since the gradient is noisy, some residual MSE after convergence can be expected. This residual error is called **Excess in the MSE** and is defined at time  $n$  by

$$\begin{aligned}\Delta\xi(n) = \xi(n) - \xi_{min} &= E\{\tilde{\boldsymbol{\theta}}^T(n)\mathbf{R}_x\tilde{\boldsymbol{\theta}}(n)\} \\ &= E\{tr(\mathbf{R}_x\tilde{\boldsymbol{\theta}}(n))\tilde{\boldsymbol{\theta}}^T(n)\} \\ &= tr\left(E\{\mathbf{R}_x\tilde{\boldsymbol{\theta}}(n)\tilde{\boldsymbol{\theta}}^T(n)\}\right)\end{aligned}$$

where  $tr(\mathbf{AB}) = tr(\mathbf{BA})$  was used.

Using this, and after some not trivial intermediate steps, it is possible to shown that

$$\begin{aligned}\Delta\xi(n) &\cong \frac{\mu\sigma_\nu^2\sum_{k=0}^N\lambda_k}{1 - \mu\sum_{k=0}^N\lambda_k} \\ &= \frac{\mu\sigma_\nu^2tr[\mathbf{R}]}{1 - \mu tr[\mathbf{R}]}\end{aligned}$$

where  $\sigma_\nu^2 = E\{\nu^2(n)\}$ . Finally, for  $n \rightarrow \infty$

$$\xi_{exc} = \lim_{n \rightarrow \infty} \Delta\xi(n) \cong \frac{\mu\sigma_\nu^2tr[\mathbf{R}]}{1 - \mu tr[\mathbf{R}]}$$

and assuming  $\mu$  small enough,

$$\xi_{exc} \cong \mu\sigma_n^2tr[\mathbf{R}] = \mu(N+1)\sigma_\nu^2\sigma_x^2$$

Note that  $\xi_{exc}$  is a relative quantity. In order to compare different algorithms a more suitable parameter is the **Misadjustment**:

$$M \triangleq \frac{\xi_{exc}}{\xi_{min}} = \frac{\mu tr[\mathbf{R}]}{1 - \mu tr[\mathbf{R}]}$$



### 2.3.3 MSE transient

The essential drawback related to the LMS algorithm is that convergence speed depends directly on the correlation matrix eigenvalue spread.

Using the expression of the MSE at time  $n$  it is not hard to show that

$$\begin{aligned}\xi(n) &= \xi_{min} + E\{\tilde{\mathbf{v}}^T(n)\mathbf{\Lambda}\tilde{\mathbf{v}}(n)\} \\ &= \xi_{min} + \sum_{k=0}^N \lambda_k \tilde{v}_k^2(n) \\ &= \xi_{min} + \sum_{k=0}^N \lambda_k (1 - \mu\lambda_k)^{2n} \tilde{v}_k^2(0)\end{aligned}$$

Then the transient that characterizes the behavior of the MSE convergence is related to  $N+1$  geometric ratios,  $r_k = 1 - 2\mu\lambda_k$ . Using the usual exponential envelope  $r_k = 1 - 2\mu\lambda_k \cong 1 - \frac{1}{\tau_k}$ , then

$$\tau_k \cong \frac{1}{2\mu\lambda_k}$$

with  $k = 0, \dots, N$ . This is the time constants related to parameter convergence. For MSE convergence speed

$$\tau_{xi_k} \cong \frac{1}{4\mu\lambda_k}$$

### 2.3.4 The Normalized LMS algorithm

- To optimize the convergence speed: a time variant convergence factor  $\mu(n)$  in the LMS algorithm.
- Consider the difference between the instantaneous squared error  $e^2(n)$  and the squared error obtained by  $\boldsymbol{\theta}_*(n) = \boldsymbol{\theta}(n) + \boldsymbol{\Delta}\boldsymbol{\theta}(n)$ , given by  $e_*^2(n)$ .
- Then

$$\begin{aligned}\Delta e^2(n) &= e_*^2(n) - e^2(n) \\ &= -2\boldsymbol{\Delta}\boldsymbol{\theta}^T(n)\boldsymbol{x}(n)e(n) \\ &\quad + \boldsymbol{\Delta}\boldsymbol{\theta}^T(n)\boldsymbol{x}(n)\boldsymbol{x}^T(n)\boldsymbol{\Delta}\boldsymbol{\theta}(n)\end{aligned}$$

- Using the  $\boldsymbol{\Delta}\boldsymbol{\theta}(n)$  obtained from the LMS algorithm and by minimization of the previous equation with respect to  $\mu(n)$ ,

$$\mu(n) = \left( \frac{1}{2\boldsymbol{x}^T(n)\boldsymbol{x}(n)} \right)$$

### 2.3.5 The Transform-Domain LMS Algorithm

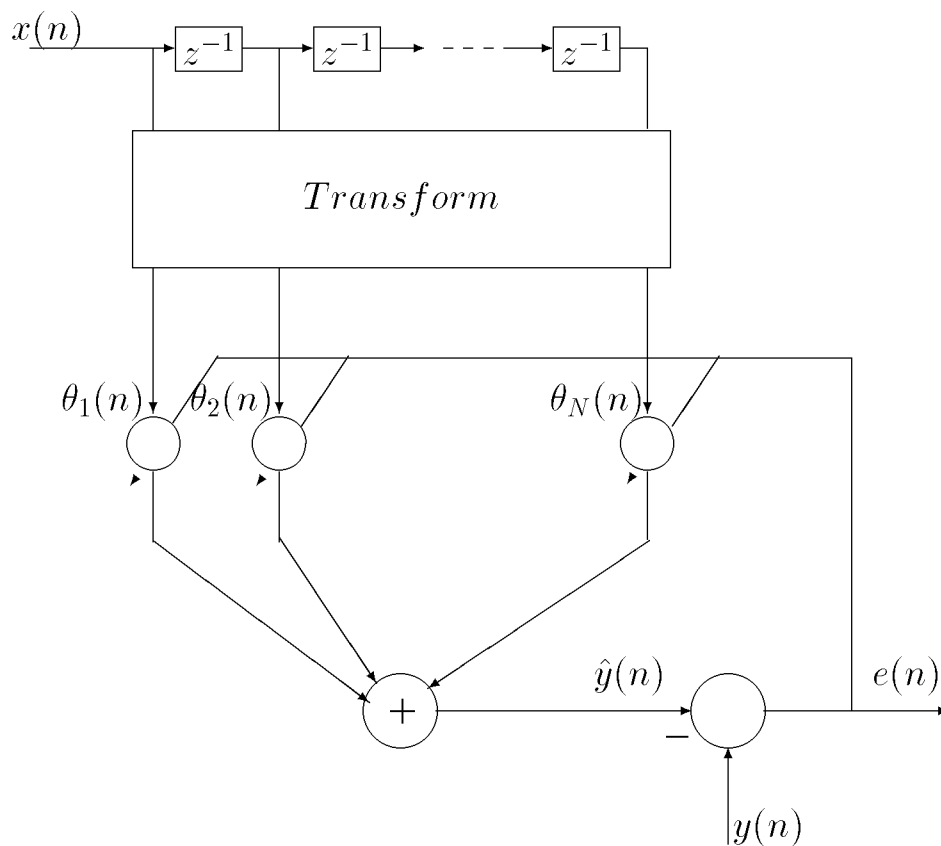


Figure 12: The transform domain adaptive filter

- Convergence speed is related to different principal axes length of the **MSE surface** countours.
- If these countours are circular the optimum situation is at hand. This can be achieved only if the eigenvectors of the  $\mathbf{R}_x$  matrix are known.
- The MSE surface is changed by a coordinate transform,  $\hat{\mathbf{x}}(n) = \mathbf{T}\mathbf{x}(n)$  where  $\mathbf{T}\mathbf{T}^T = \mathbf{I}$ , or

$$\xi = E\{e^2(n)\} = \xi_{min} + \tilde{\boldsymbol{\theta}}^T E\{\hat{\mathbf{x}}(n)\hat{\mathbf{x}}^T(n)\} \tilde{\boldsymbol{\theta}}$$

where  $\tilde{\boldsymbol{\theta}} = \hat{\boldsymbol{\theta}}(n) - \hat{\boldsymbol{\theta}}_o$ . Then

$$\xi - \xi_{min} = \tilde{\boldsymbol{\theta}}^T \mathbf{T}\mathbf{R}_x\mathbf{T}^T \tilde{\boldsymbol{\theta}}$$

that represent a **rotation** of the parameter space related to the direct form FIR filter.

- The intersection of the different MSE contour with the  $i$ -th space parameter coordinates is  $\xi - \xi_{min} = [\mathbf{T}\mathbf{R}_x\mathbf{T}^T]_{ii} \tilde{\theta}_i$ .
- For an hypersphere it is necessary that  $|\tilde{\theta}_i| = |\tilde{\theta}_j|$  for all  $(i, j)$ .
- This conditions can be achieved, at least approximately, using an scaling factor

$$[\mathbf{T}\mathbf{R}_x\mathbf{T}^T]_{ii} \cong E\{\hat{x}_i^2(n)\} = \hat{\sigma}_i^2$$

- The updating equation of the Transform Domain LMS algorithm is the following

$$\hat{\boldsymbol{\theta}}(n+1) = \hat{\boldsymbol{\theta}}(n) + \mu\boldsymbol{\Lambda}^{-1}\hat{\mathbf{x}}(n)e(n) = \hat{\boldsymbol{\theta}}(n) + \mu\boldsymbol{\Lambda}^{-1}\mathbf{T}\mathbf{x}(n)e(n)$$

where  $\boldsymbol{\Lambda} = \text{diag} [\hat{\sigma}_1^2, \dots, \hat{\sigma}_N^2]$  and  $\hat{\sigma}_i^2(n+1) = (1 - \mu_\sigma)\hat{\sigma}_i^2(n) + \mu_\sigma\hat{x}_i^2(n)$ , with  $\mu_\sigma$  a small constant.

- Two suitable transform for this algorithm are the *Discrete Fourier Transform* (complex) and the *Discrete Cosine Transform* (real), given by

$$\hat{x}_i(n) = \sqrt{\frac{2}{N+1}} \sum_{k=0}^N x(k-n) \cos\left(\pi i \frac{(2k+1)}{2(N+1)}\right)$$

### 2.3.6 The Quasi-Newton algorithm

- Higher complexity than the LMS but with fast (initial) convergence speed using an estimate of  $\mathbf{R}_x^{-1}$ .
- A possible algorithm is the following

$$\boldsymbol{\theta}(n+1) = \boldsymbol{\theta}(n) + \mu \mathbf{P}(n+1) \mathbf{x}(n) e(n) \quad (11)$$

where

$$\mathbf{P}(n+1) = \left( \frac{1}{1-\mu} \right) \left( \mathbf{P}(n) - \frac{\mathbf{P}(n) \mathbf{x}(n) \mathbf{x}^T(n) \mathbf{P}(n)}{\frac{1-\mu}{\mu} + \mathbf{x}^T(n) \mathbf{P}(n) \mathbf{x}(n)} \right) \quad (12)$$

- $\mathbf{P}(n+1)$  represents an estimate of  $\mathbf{R}_x^{-1}$  at time  $n+1$ , in this case using the *matrix inversion lemma*. This algorithm is called *Quasi-Newton*.

## 2.4 Other algorithms

### 2.4.1 The RLS algorithm

Assuming a linear regressor model:

$$y(n) = \sum_{k=1}^N \theta_k^o x(n-k) + \nu(n)$$

The RLS algorithm estimates the  $\boldsymbol{\theta}_o$  parameters by minimizing

$$V_N(\boldsymbol{\theta}) = \frac{1}{N} \sum_{n=1}^N e^2(n)$$

where  $e(n) = y(n) - \boldsymbol{\theta}^T(n) \mathbf{x}_N(n)$ . The well known recursive solution of this problem is

$$\boldsymbol{\theta}(n) = \boldsymbol{\theta}(n-1) + \boldsymbol{\kappa}_N(n) (y(n) - \boldsymbol{\theta}^T(n-1) \mathbf{x}_N(n))$$

where

$$\begin{aligned} \boldsymbol{\kappa}_N(n) &= \mathbf{R}_{N-1}^{-1}(n) \mathbf{x}_N(n) \quad n \geq N \\ \mathbf{R}_{N-1}(n) &= \mathbf{R}_{N-1}(n-1) + \mathbf{x}_N(n) \mathbf{x}_N^T(n) \end{aligned}$$

### 2.4.2 The fast RLS algorithm

- The fast RLS will be derived by close relationship with the *conjugate direction algorithm* and the forward and backward prediction filters.
- The choice of two particular conjugate directions is essential for the present derivation of fast RLS algorithm. These conjugate directions are related to the **forward and backward prediction filter coefficients** as discussed below.
- The fast RLS algorithm is related to the Kalman gain updating (in time)  $\boldsymbol{\kappa}_N(n-1) \rightarrow \boldsymbol{\kappa}_N(n)$ .
- This updating can be seen as composed of time update and order update.
  1.  $\boldsymbol{\kappa}_N(n-1) \rightarrow \boldsymbol{\kappa}_{N+1}(n)$ ,
  2.  $\boldsymbol{\kappa}_{N+1}(n) \rightarrow \boldsymbol{\kappa}_N(n)$ .
- Due to the shifted structure of the regressor  $\boldsymbol{x}_N(n)$ ,

$$\boldsymbol{x}_{N+1}(n) = \begin{bmatrix} x(n) \\ \boldsymbol{x}_N(n-1) \end{bmatrix} = \begin{bmatrix} \boldsymbol{x}_N(n) \\ x(n-N) \end{bmatrix}$$

where  $\boldsymbol{x}_{N+1,1:N}(n) = \boldsymbol{x}_N(n-1)$  and  $\boldsymbol{x}_{N+1,0:N-1}(n) = \boldsymbol{x}_N(n)$ .

- Then, assuming that  $\boldsymbol{x}_N(n) = 0$  for  $n \leq 0$ ,

$$\boldsymbol{R}_{N-1}(n-1) = \boldsymbol{R}_{1:N}(n) \quad \boldsymbol{R}_{N-1}(n) = \boldsymbol{R}_{0:N-1}(n)$$

where  $\boldsymbol{R}_{1:N}(n)$  and  $\boldsymbol{R}_{0:N-1}(n)$  are the lower right and upper left corner of  $\boldsymbol{R}_N(n)$ , respectively.

- Kalman gain at times  $n - 1$  and  $n$  can be written

$$\begin{aligned}\boldsymbol{\kappa}_N(n-1) &= \mathbf{R}_{1:N}^{-1}(n-1)\mathbf{x}_{N+1,1:N}(n) \\ \boldsymbol{\kappa}_N(n) &= \mathbf{R}_{0:N-1}^{-1}(n)\mathbf{x}_{N+1,0:N-1}(n) \\ \boldsymbol{\kappa}_{N+1}(n) &= \mathbf{R}_N^{-1}(n)\mathbf{x}_{N+1}(n)\end{aligned}$$

- Following the first step (time update) above, the problem can be stated has: given  $\boldsymbol{\kappa}_N(n-1)$  and  $\boldsymbol{\kappa}_N(n)$ , find  $\boldsymbol{\kappa}_{N+1}(n)$  as the solution to the  $N + 1$ -dimensional problem

$$\underset{\mathbf{z} = \boldsymbol{\kappa}_{N+1}(n)}{\text{Minimize}} \quad \left( \frac{1}{2} \mathbf{z}^T \mathbf{R}_N(n) \mathbf{z} - \mathbf{x}_{N+1}^T(n) \mathbf{z} \right)$$

- This can be achieved with a conjugate direction algorithm with

$$\begin{aligned}\mathbf{d}_N &= [1 \ \mathbf{a}_N^T(n)]^T \\ \mathbf{d}_N^T \nabla(n) &= [1 \ \mathbf{a}_N^T(n)] \mathbf{x}_{N+1}(n) = e_N^f(n) \\ \mathbf{d}_N^T \mathbf{R}_N(n) \mathbf{d}_N &= \xi_N^f(n)\end{aligned}$$

where  $\mathbf{a}_N(n)$  are the coefficients of the forward prediction filter,  $\xi_N^f(n)$  is an estimate of the *least square forward prediction error* and  $e_N^f(n)$  is the *aposteriori forward prediction error*.

- Then

$$\boldsymbol{\kappa}_{N+1}(n) = \begin{bmatrix} 0 \\ \boldsymbol{\kappa}_N(n-1) \end{bmatrix} + \begin{bmatrix} 1 \\ \mathbf{a}_N(n) \end{bmatrix} \left( \xi_N^f(n) \right)^{-1} e_N^f(n)$$



- For the second step (order update), i.e.,  $\boldsymbol{\kappa}_{N+1}(n) \rightarrow \boldsymbol{\kappa}_N(n)$ , the problem can be stated as: given  $\boldsymbol{\kappa}_N(n-1)$  and  $\boldsymbol{\kappa}_N(n)$ , find  $\boldsymbol{\kappa}_{N+1}(n)$  as the solution to the  $N+1$ -dimensional problem

$$\underset{\mathbf{z} = \boldsymbol{\kappa}_{N+1}(n)}{\text{Minimize}} \quad \left( \frac{1}{2} \mathbf{z}^T \mathbf{R}_N(n) \mathbf{z} - \mathbf{x}_{N+1}^T(n) \mathbf{z} \right)$$

- This can be achieved using a conjugate direction algorithm with

$$\begin{aligned} \mathbf{d}_N &= [\mathbf{b}_N^T(n) \ 1]^T \\ \mathbf{d}_N^T \nabla(n) &= [\mathbf{b}_N^T(n) \ 1] \mathbf{x}_{N+1}(n) = e_N^b(n) \\ \mathbf{d}_N^T \mathbf{R}_N(n) \mathbf{d}_N &= \xi_N^b(n) \end{aligned}$$

where  $\mathbf{b}_N(n)$  are the coefficients of the backward prediction filter,  $\xi_N^b(n)$  is an estimate of the *least square backward prediction error* and  $e_N^b(n)$  is the *aposteriori backward prediction error*.

- Using this results,

$$\boldsymbol{\kappa}_{N+1}(n) = \begin{bmatrix} \boldsymbol{\kappa}_N(n) \\ 0 \end{bmatrix} + \begin{bmatrix} \mathbf{b}_N(n) \\ 1 \end{bmatrix} (\xi_N^b(n))^{-1} e_N^b(n)$$

- Since the required solution is the Kalman time update,  $\boldsymbol{\kappa}_N(n)$  is obtained as a function of  $\boldsymbol{\kappa}_{N+1}(n)$  from the previous equation.
- The complete fast RLS algorithm requires 2 CD algorithms for the time update of the Kalman gain and 2 CD algorithms to obtain:
  - a) the time update of the prediction filter coefficients and
  - b) the parameter updates  $\boldsymbol{\theta}(n)$ .

### 2.4.3 QR decomposition based RLS algorithm

- If the standard RLS algorithm

$$\boldsymbol{\theta}(n+1) = \boldsymbol{\theta}(n) + \left[ \sum_{k=0}^n \lambda^{n-k} \mathbf{x}(k) \mathbf{x}^T(k) \right]^{-1} \mathbf{x}(n) e(n)$$

where  $e(n) = y(n) - \boldsymbol{\theta}^T(n) \mathbf{x}(n)$  ( $\mathbf{x}(n) = \mathbf{x}_N(n)$ ) and  $0 \ll \lambda \leq 1$  is the forgetting factor, is rewritten as

$$\begin{bmatrix} e(n) \\ 0 \\ \vdots \\ 0 \end{bmatrix} = \begin{bmatrix} \mathbf{x}^T(n) \\ \lambda^{1/2} \mathbf{x}^T(n-1) \\ \vdots \\ \lambda^{1/2} \mathbf{x}^T(0) \end{bmatrix} \tilde{\boldsymbol{\theta}}(n) \quad (13)$$

where  $\tilde{\boldsymbol{\theta}}(n) = \boldsymbol{\theta}(n+1) - \boldsymbol{\theta}(n)$ .

- Then  $e(n) \mathbf{u}_1 - \mathbf{X}(n) \tilde{\boldsymbol{\theta}}(n)$ , where  $\mathbf{u}_1$  is the unit vector with a "1" in the first position, and

$$\mathbf{X}(n) = \begin{bmatrix} \mathbf{x}^T(n) \\ \lambda^{1/2} \mathbf{X}(n-1) \end{bmatrix}$$

- If an  $n \times n$  (with  $n \geq N+1$ ) orthogonal matrix  $\mathbf{Q}(n-1)$  is known at time  $n-1$  such that

$$\mathbf{Q}(n-1) \mathbf{X}(n-1) = \begin{bmatrix} \circ \\ \mathbf{R}(n-1) \end{bmatrix}$$

where  $\mathbf{R}(n-1)$  is an upper triangular matrix of dimension  $N \times N$  (dimension of  $\boldsymbol{\theta}(n)$ ).

- Then

$$\begin{bmatrix} 1 & \\ & \mathbf{Q}(n-1) \end{bmatrix} (e(n)\mathbf{u}_1 - \mathbf{X}(n)\tilde{\boldsymbol{\theta}}(n)) = \begin{bmatrix} e(n) \\ 0 \\ \vdots \\ 0 \end{bmatrix} - \begin{bmatrix} \mathbf{x}^T(n) \\ \circ \\ \lambda^{1/2}\mathbf{R}(n-1) \end{bmatrix}$$

- If  $\mathbf{R}(n-1)$  is known, the triangularization at time  $n$  can be completed by introducing zeros into the locations occupied by the most recent vector  $\mathbf{x}(n)$ .
- This is achieved by an  $n \times n$  orthogonal matrix  $\hat{\mathbf{Q}}(n)$

$$\hat{\mathbf{Q}}(n) \begin{bmatrix} e(n) \\ 0 \\ \vdots \\ 0 \end{bmatrix} - \hat{\mathbf{Q}}(n) \begin{bmatrix} \mathbf{x}^T(n) \\ \circ \\ \lambda^{1/2}\mathbf{R}(n-1) \end{bmatrix} \tilde{\boldsymbol{\theta}}(n) = e(n)\hat{\mathbf{q}}_1(n) - \begin{bmatrix} \circ \\ \mathbf{R}(n) \end{bmatrix} \tilde{\boldsymbol{\theta}}(n) \quad (14)$$

where  $\hat{\mathbf{q}}_1(n)$  is the first column of  $\hat{\mathbf{Q}}(n)$ .

- (14) can be performed using *Givens rotations*, such that

$$\hat{\mathbf{Q}}(n) = \hat{\mathbf{Q}}_N \dots \hat{\mathbf{Q}}_1$$

with

$$\hat{\mathbf{Q}}_k = \begin{bmatrix} \cos \varphi_k & & & -\sin \varphi_k & & \\ & \mathbf{I}_{n+k-N-1} & & & & \\ \sin \varphi_k & & & \cos \varphi_k & & \\ & & & & & \mathbf{I}_{N-k} \end{bmatrix}$$

- The proper selection of the rotation angles  $\{\varphi_k\}$  will annihilate the elements of  $\mathbf{x}^T(n)$  appearing in (14).

- The term  $\hat{\mathbf{q}}_1(n)$  in (14) in closed form is

$$\hat{\mathbf{q}}_1(n) = \begin{bmatrix} \prod_{k=1}^N \cos \varphi_k \\ \mathbf{0} \\ \mathbf{g} \end{bmatrix}$$

where  $\mathbf{g} = [g_1, \dots, g_N]^T$ ,  $g_k = \sin \varphi_k \prod_{i=1}^{k-1} \cos \varphi_i$ .

- The parameter update  $\tilde{\boldsymbol{\theta}}(n)$  is then solved from

$$e(n)\mathbf{g} = \mathbf{R}(n)\tilde{\boldsymbol{\theta}}(n)$$

using back substitution.

- An useful scaled algorithm can be obtained considering the  $QR$  decomposition of  $\mathbf{X}(n)$

$$\mathbf{Q}(n)\mathbf{X}(n) = \begin{bmatrix} \mathbf{0} \\ \mathbf{R}(n) \end{bmatrix}$$

- Because  $\mathbf{Q}(n)$  is orthogonal, we have

$$\mathbf{R}^T(n)\mathbf{R}(n) = \mathbf{X}^T(n)\mathbf{X}(n) = \sum_{k=0}^n \lambda^{n-k} \mathbf{x}(k)\mathbf{x}^T(k)$$

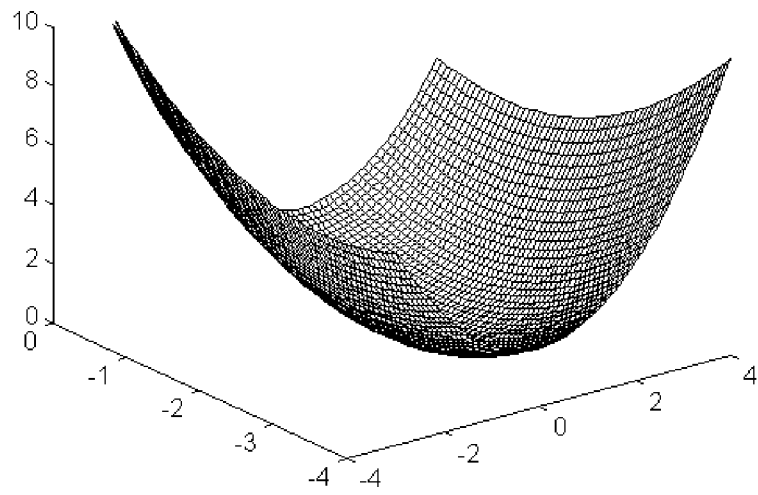
- If  $\mathbf{x}(n)$  is stationary,

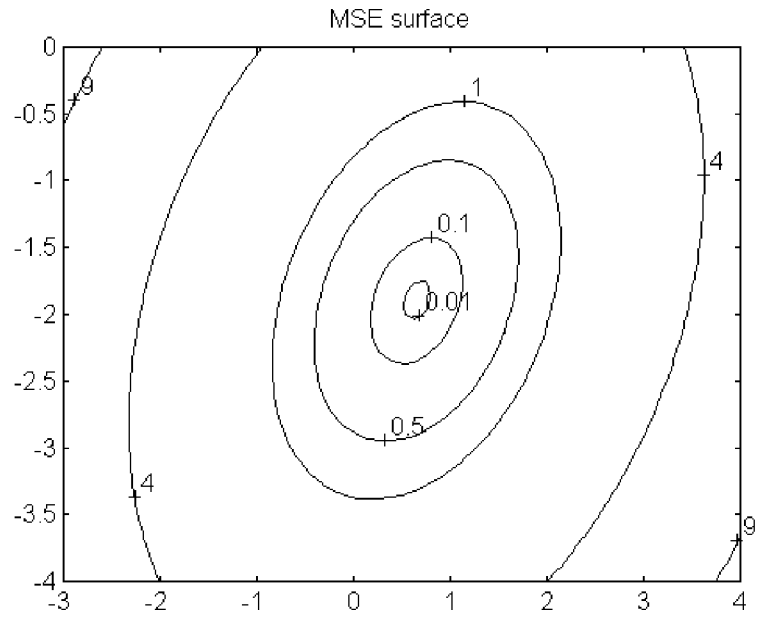
$$n \xrightarrow{\lim} \infty E\{\mathbf{R}^T(n)\mathbf{R}(n)\} = \frac{E\{\mathbf{x}(n)\mathbf{x}^T(n)\}}{1 - \lambda}$$

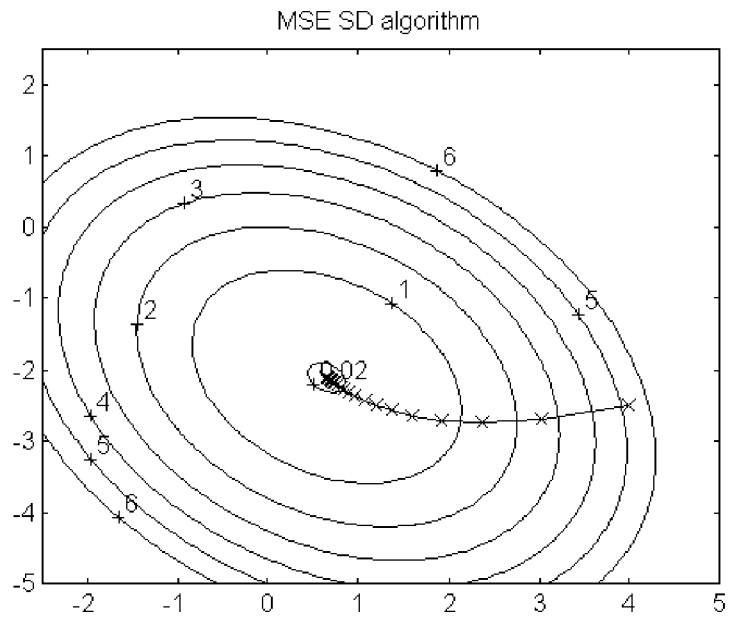
- Then for  $\lambda \rightarrow 1$ , the elements of  $\mathbf{R}(n)$  can become large.
- Overflow in  $\mathbf{R}(n)$  can be avoided considering in (13) that if  $\{\mathbf{x}(k)\}_{k=0}^n$  and  $e(n)$  are similarly scaled, the LS is left unchanged.
- From (15) an appropriate choice is  $\sqrt{1 - \lambda}$ .

## Examples

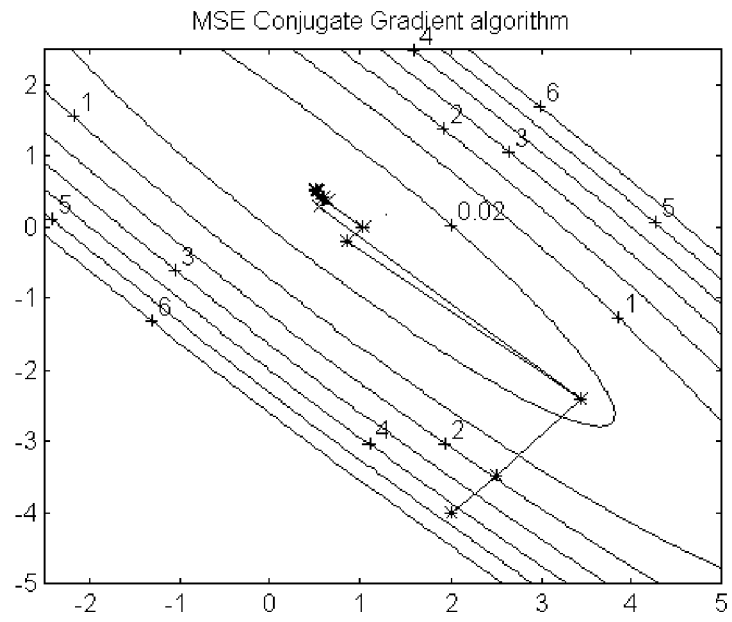
- An (adaptive) signal-cancelling application, with two taps!
- $x(n) = \sin w_0 n + \nu(n)$  with  $E\{\nu^2(n)\} = r$ ,
- $y(n) = 2 \cos w_0 n$ ,
- $\hat{y}(n) = \theta_0 x(n) + \theta_1 x(n-1)$ ,
- $e(n) = y(n) - \hat{y}(n)$
- $\mathbf{R}_x = \frac{1}{2} \begin{bmatrix} 1 + 2r & \cos w_0 \\ \cos w_0 & 1 + 2r \end{bmatrix}$  and  $\mathbf{p} = 2 \begin{bmatrix} 0 \\ -\sin w_0 \end{bmatrix}$ .
- $\boldsymbol{\theta}^* = \mathbf{R}^{-1} \mathbf{p}$ .
- $E\{e^2(n)\} = (\frac{1}{2} + r)(\theta_0^2 + \theta_1^2) + \theta_0 \theta_1 \cos w_0 + 2\theta_1 \sin w_0 + 2$

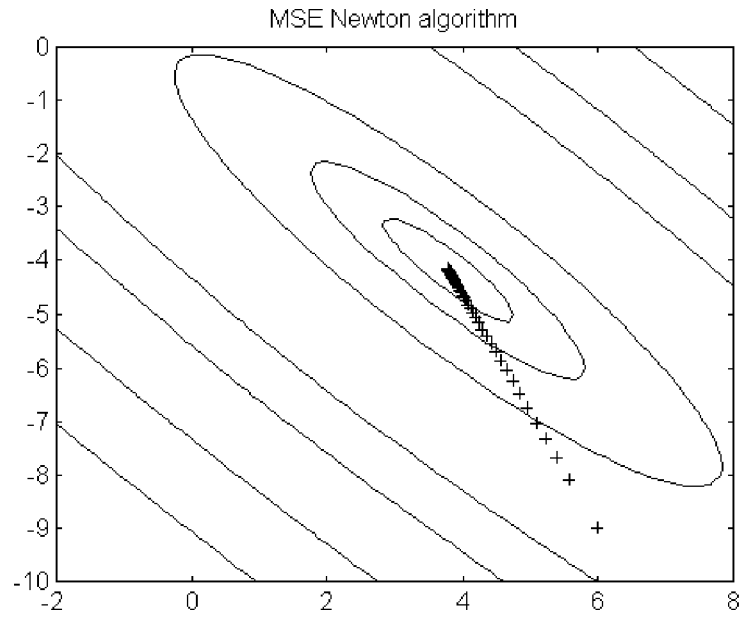


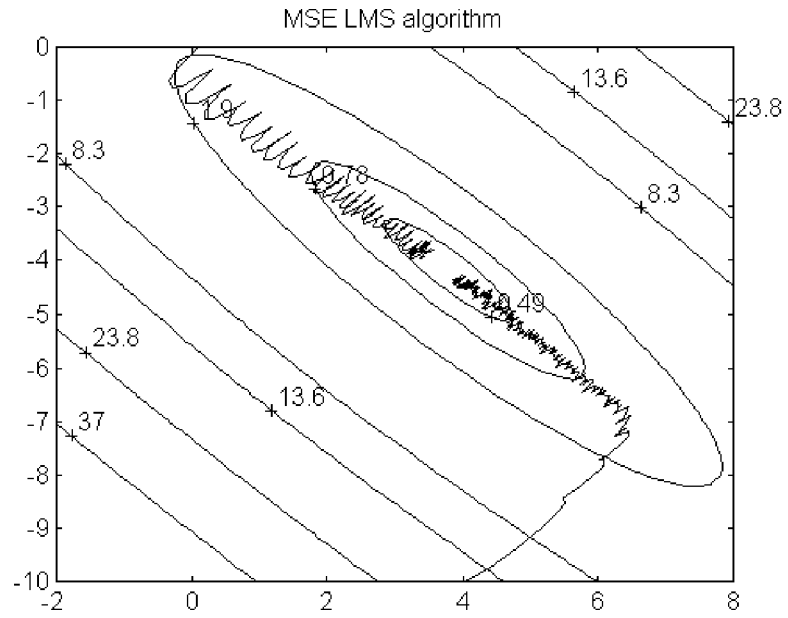


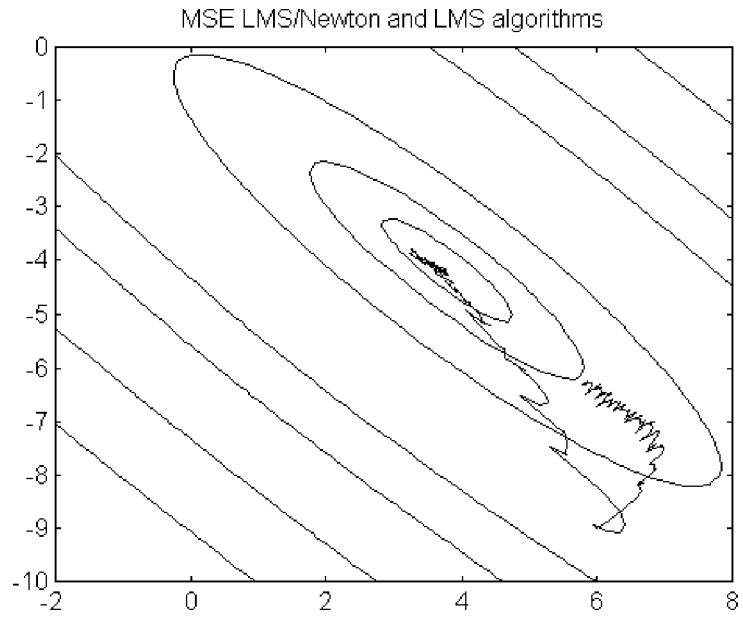












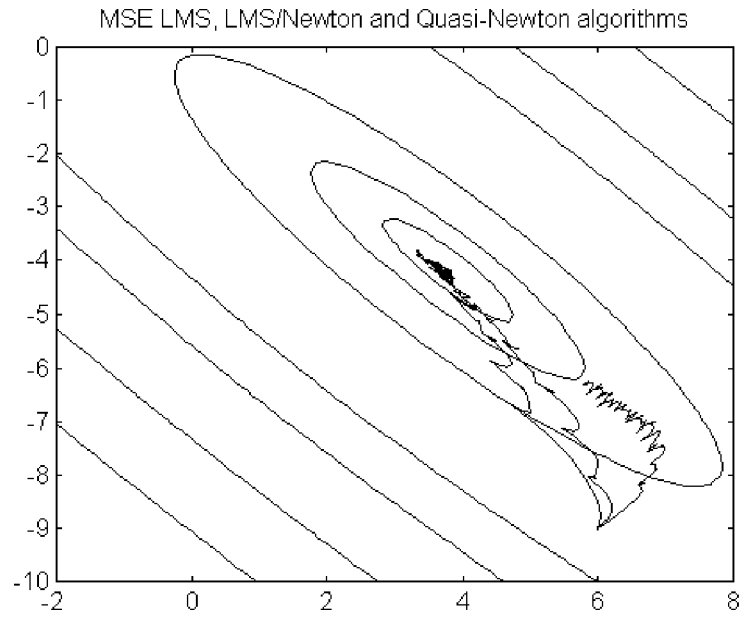


Figure 21:

