



Statistical Signal Processing

Lecture 10

chapter 3: Optimal Filtering

Wiener filtering

- FIR Wiener filtering
 - iterative solution: steepest-descent algorithm

chapter 4: Adaptive Filtering

- LMS algorithm
- Normalized LMS (NLMS) algorithm
- tracking behavior of LMS and RLS
- optimal tracking via Kalman filtering

chapter 5: Sinusoids in Noise



RLS Algorithm

- LS: replace the statistical averages by a time averages:

$$\xi_k(H) = \sum_{i=1}^k (x_i - H^T Y_i)^2 + (H - H_0)^T R_0 (H - H_0) \quad ,$$

where the second term with $R_0 = R_0^T > 0$ allows for a proper initialization of the algorithm (the first term alone has a singular Hessian $(= 2 \sum_{i=1}^k Y_i Y_i^T)$ for $k < N$).

- We can rewrite

$$\begin{aligned} \xi_k(H) &= H^T \left(\sum_{i=1}^k Y_i Y_i^T \right) H - 2H^T \left(\sum_{i=1}^k Y_i x_i \right) + \sum_{i=1}^k x_i^2 + (H - H_0)^T R_0 (H - H_0) \\ &= H^T \left(R_0 + \sum_{i=1}^k Y_i Y_i^T \right) H - 2H^T \left(R_0 H_0 + \sum_{i=1}^k Y_i x_i \right) + \sum_{i=1}^k x_i^2 + H_0^T R_0 H_0 \\ &= H^T R_k H - 2H^T P_k + \sum_{i=1}^k x_i^2 + H_0^T R_0 H_0 \end{aligned}$$

where

$$\begin{aligned} R_k &= R_0 + \sum_{i=1}^k Y_i Y_i^T &= R_{k-1} + Y_k Y_k^T \\ P_k &= R_0 H_0 + \sum_{i=1}^k Y_i x_i &= P_{k-1} + Y_k x_k \quad . \end{aligned}$$



Recursive Least-Squares Algorithm (2)

- By putting the gradient of $\xi_k(H)$ equal to zero and noting that the Hessian $2R_k > 0$, we find that the LS filter H_k that minimizes the LS criterion solves the following normal equations

$$R_k H_k = P_k \quad .$$

To solve this set of equations at each time instant k would take $\mathcal{O}(N^3)$ operations at each time instant. In what follows, we shall derive the Recursive LS algorithm, which allows us, using information obtained at time $k-1$, to obtain H_k with only $\mathcal{O}(N^2)$ operations.

- we can rewrite $P_k = P_{k-1} + Y_k x_k$ as

$$\begin{aligned} R_k H_k &= R_{k-1} H_{k-1} + Y_k x_k \\ &= (R_k - Y_k Y_k^T) H_{k-1} + Y_k x_k \\ &= R_k H_{k-1} + Y_k \epsilon_k^p \end{aligned}$$

where $\epsilon_k^p = x_k - H_{k-1}^T Y_k$ as in the LMS algorithm. This leads immediately to

$$H_k = H_{k-1} + R_k^{-1} Y_k \epsilon_k^p$$

where $R_k^{-1} Y_k$ is called the Kalman gain (the RLS algorithm is a special case of the so-called Kalman filter).



Recursive Least-Squares Algorithm (3)

- Clearly, the RLS algorithm requires the recursive update of R_k^{-1} . This can be obtained using the Matrix Inversion Lemma:

$$\begin{aligned} R_k^{-1} &= (R_{k-1} + Y_k Y_k^T)^{-1} \\ &= R_{k-1}^{-1} - R_{k-1}^{-1} Y_k (1 + Y_k^T R_{k-1}^{-1} Y_k)^{-1} Y_k^T R_{k-1}^{-1} . \end{aligned}$$

This equation allows us to obtain R_k^{-1} from R_{k-1}^{-1} and Y_k using $\mathcal{O}(N^2)$ operations. When multiplying both sides with Y_k to the right, we obtain

$$R_k^{-1} Y_k = R_{k-1}^{-1} Y_k (1 + Y_k^T R_{k-1}^{-1} Y_k)^{-1} .$$

We find for the *a posteriori* error

$$\epsilon_k = x_k - H_k^T Y_k = (1 - Y_k^T R_{k-1}^{-1} Y_k) \epsilon_k^p = (1 + Y_k^T R_{k-1}^{-1} Y_k)^{-1} \epsilon_k^p .$$

- All this can be formulated as the RLS algorithm:

$$\left\{ \begin{array}{l} \epsilon_k^p = x_k - H_{k-1}^T Y_k \\ \epsilon_k = \epsilon_k^p (1 + Y_k^T R_{k-1}^{-1} Y_k)^{-1} \\ H_k = H_{k-1} + R_{k-1}^{-1} Y_k \epsilon_k \\ R_k^{-1} = R_{k-1}^{-1} - R_{k-1}^{-1} Y_k (1 + Y_k^T R_{k-1}^{-1} Y_k)^{-1} Y_k^T R_{k-1}^{-1} . \end{array} \right.$$



Recursive Least-Squares Algorithm (4)

- The initial values for R_k^{-1} and H_k are R_0^{-1} and H_0 . Compared to the LMS algorithm, the scalar stepsize μ gets replaced by a matrix stepsize R_k^{-1} . The RLS algorithm takes $\mathcal{O}(N^2)$ operations while the LMS algorithm takes only $2N$ operations. However, it converges much faster.
- *performance analysis* : with $x_k = H^o{}^T Y_k + \tilde{x}_k$ (and $R_0 = 0$), we get
$$R_k H_k = P_k = \sum_{i=1}^k Y_i x_i = R_k H^o + \sum_{i=1}^k Y_i \tilde{x}_i . \text{ Hence}$$

$$\widetilde{H}_k = H^o - H_k = -R_k^{-1} \sum_{i=1}^k Y_i \tilde{x}_i$$

From this, we obtain

$$C_k \triangleq E \widetilde{H}_k \widetilde{H}_k^T = \sigma_{\tilde{x}}^2 R_k^{-1} .$$

Since R_k^{-1} behaves as $1/k$, we see that C_k converges to zero as $1/k$.

- *Exponential Weighting* In order to be able to track a possibly time-varying $H^o = H_k^o$, one introduces an exponential forgetting factor $\lambda \in (0, 1)$ into the cost function to obtain

$$\xi_k(H) = \sum_{i=1}^k \lambda^{k-i} (x_i - H^T Y_i)^2 + \lambda^k (H - H_0)^T R_0 (H - H_0) .$$

This implies that the past (and in particular the initial conditions H_0, R_0) is forgotten exponentially fast with a window with time constant $1/(1-\lambda)$.



Recursive Least-Squares Algorithm (5)

- Wiener filtering: x_k and y_k are two joint stochastic processes and we're trying to estimate x_k from the y_k using a LMMSE estimator. For an FIR Wiener filter, there are a finite set of coefficients H^o involved in this LMMSE estimator.
RLS approach: replaced statistical averages with temporal averages.

Parameter estimation interpretation

- Assume now our usual model for the *measurements* x_k ,

$$x_k = H^{oT} Y_k + \tilde{x}_k$$

where the \tilde{x}_k are iid with zero mean and variance $\sigma_{\tilde{x}}^2$. Consider here $\{y_k\}$ as a deterministic signal, so the only randomness comes from the $\{\tilde{x}_k\}$. The H^o are the unknown parameters governing the model.

- The analysis of the RLS algorithm is much simpler than that of the LMS algorithm since for each k the RLS solution H_k coincides with the solution of a Least-Squares problem with a closed-form solution: $H_k = R_k^{-1} P_k$.
- Assume now $k \geq N$, $H_0 = 0$, $R_0 = 0$, and that R_k is nonsingular. The performance of the least-squares estimate is simple to analyze and leads to

$$C_k \triangleq E \tilde{H}_k \tilde{H}_k^T = \sigma_{\tilde{x}}^2 R_k^{-1} .$$

- If \tilde{x}_k Gaussian, $\Rightarrow H_k = \text{ML estimate of } H^o \text{ (efficient, } C_k = \text{CRB})$.



Recursive Least-Squares Algorithm (6)

A Bayesian Context - A Priori Information

- Instead of treating the filter coefficients H^o as unknown constant parameters, we could also consider H^o as a stochastic parameter vector about which we have some prior information, possibly from previous adaptive filtering experience. Assume now that, prior to obtaining the measurements x_1, x_2, \dots , we know that H^o has a distribution with mean $E H^o = H_0$ and covariance $E (H^o - H_0) (H^o - H_0)^T = C_0$. So now the randomness in the x_k comes from both the \tilde{x}_k and H^o .
- The problem formulation can now be recognized to be one of a *Bayesian Linear Model*. The AMMSE estimator can be shown to be the filter estimate resulting from the original RLS criterion with $R_0 = \sigma_{\tilde{x}}^2 C_0^{-1}$. $C_k = E \tilde{H}_k \tilde{H}_k^T$ now satisfies

$$C_k^{-1} = \sigma_{\tilde{x}}^{-2} R_k = \sigma_{\tilde{x}}^{-2} \sum_{i=1}^k Y_i Y_i^T + C_0^{-1} .$$

Note that C_k^{-1} is an increasing function of C_0^{-1} and hence C_k is a decreasing function of C_0^{-1} and hence of R_0 .

- So we see that H_0 and R_0 in the LS cost function have the interpretation of the prior mean and the inverse of the prior covariance of H^o . We'll choose R_0 small if we don't have a lot of confidence in our prior guess H_0 (C_0 big). In practice, R_0 is often chosen as $R_0 = \eta I_N$.



Other Adaptive Filtering Algorithms

- Fast RLS algorithms: Fast Transversal Filter (FTF) algorithm ($8N$), Fast Lattice/QR Algorithms ($\mathcal{O}(N)$ complexity)
- LMS with prewhitened input
- block processing/frequency domain LMS
- subband structures
- Fast Newton Transversal Filter (FNTF): replace R^{-1} in RLS by a banded matrix (appropriate for AR processes, hence speech)
- projection algorithms (like NLMS) on an extended subspace of L input vectors (FAP: Fast Affine Projection: complexity $2N + \mathcal{O}(L^2)$ or $2N + \mathcal{O}(L)$)
- Fast Subsampled Updating (FSU) versions of LMS and FTF: introduce some delay to reduce complexity below $\mathcal{O}(N)$
- multistage Wiener filter / polynomial expansion:

$$r_0 R^{-1} = \left[\frac{1}{r_0} R \right]^{-1} = \left[\underbrace{I}_{\text{diagonal}} + \underbrace{\left(\frac{1}{r_0} R - I \right)}_{\text{off-diagonal part}} \right]^{-1} = \sum_{i=0}^{\infty} \left(I - \frac{1}{r_0} R \right)^i = \sum_{i=0}^{\infty} \alpha_i R^i$$

- convergence speed (RLS best) versus tracking speed (FAP best?)



Initial Convergence RLS

- Consider now $H_0 \neq 0$, $R_0 \neq 0$,

$$R_k = R_0 + R_{1:k}, R_{1:k} = Y_{1:k} Y_{1:k}^T, Y_{1:k} = [Y_1 \cdots Y_k], P_k = R_0 H_0 + P_{1:k}$$

- $\widetilde{H}_k = H^o - H_k = H^o - R_k^{-1} P_k = (R_0 + R_{1:k})^{-1} (R_0 \widetilde{H}_0 - \sum_{i=1}^k Y_i \widetilde{x}_i)$

- $C_k = E \widetilde{H}_k \widetilde{H}_k^T$ hence

$$\begin{aligned} C_k &= (R_0 + R_{1:k})^{-1} R_0 \widetilde{H}_0 \widetilde{H}_0^T R_0 (R_0 + R_{1:k})^{-1} + \sigma_{\widetilde{x}}^2 (R_0 + R_{1:k})^{-1} R_{1:k} (R_0 + R_{1:k})^{-1} \\ &= \underbrace{\sigma_{\widetilde{x}}^2 (R_0 + R_{1:k})^{-1}}_{\sim \frac{1}{k}} + \underbrace{(R_0 + R_{1:k})^{-1} (R_0 \widetilde{H}_0 \widetilde{H}_0^T R_0 - \sigma_{\widetilde{x}}^2 R_0) (R_0 + R_{1:k})^{-1}}_{\sim \frac{1}{k^2} \text{ due to initialization}} \end{aligned}$$

- noiseless case: $\sigma_{\widetilde{x}}^2 = 0$

$$C_k = (R_0 + R_{1:k})^{-1} R_0 \widetilde{H}_0 \widetilde{H}_0^T R_0 (R_0 + R_{1:k})^{-1}$$

- initial convergence: $1 \leq k < N$, consider $R_0 = \eta I$

$$C_k = \eta^2 (\eta I + R_{1:k})^{-1} \widetilde{H}_0 \widetilde{H}_0^T (\eta I + R_{1:k})^{-1}$$



Initial Convergence RLS (2)

- *Singular Value Decomposition (SVD)*: $Y_{1:k}$, $N \times k$ assumed full column rank

$$Y_{1:k} = V \Sigma U^T \quad V^T V = I_k, U^{-1} = U^T, \Sigma = \text{diag}\{\sigma_1, \dots, \sigma_k\}$$

$\sigma_1 \geq \sigma_2 \geq \dots \geq \sigma_k \geq 0$ “singular values” full column rank $\leftrightarrow \sigma_k > 0$

- *Moore-Penrose pseudo-inverse*: $Y_{1:k}^+ = U \Sigma^{-1} V^T = (Y_{1:k}^T Y_{1:k})^{-1} Y_{1:k}^T$
- projection on column space: $P_{Y_{1:k}} = Y_{1:k} Y_{1:k}^+ = V V^T$
- $V^+ = V^T$ $P_{Y_{1:k}} = V V^T = V V^+ = P_V$ $P_V^+ = P_V$
- eigendecomposition: $R_{1:k} = Y_{1:k} Y_{1:k}^T = V \Sigma^2 V^T$
- let V^\perp be such that $[V \ V^\perp]$ is orthogonal:

$$[V \ V^\perp][V \ V^\perp]^T = I = V V^T + V^\perp V^{\perp T} = P_V + P_{V^\perp} = P_V + P_V^\perp$$

$P_V^\perp = I - P_V = P_{V^\perp}$, V^\perp spans orthogonal complement of V

- SVD alternatively: $Y_{1:k} = [V \ V^\perp] \begin{bmatrix} \Sigma \\ 0 \end{bmatrix} U^T$, $\begin{bmatrix} \Sigma \\ 0 \end{bmatrix}^+ = [\Sigma^+ \ 0]$, $\sigma^+ = \begin{cases} 1/\sigma & , \sigma > 0 \\ 0 & , \sigma = 0 \end{cases}$
- eigendecomposition projection:

$$P_V = V I V^T = [V \ V^\perp] \begin{bmatrix} I & 0 \\ 0 & 0 \end{bmatrix} [V \ V^\perp]^T \quad \text{eigenvalues are 1 or 0}$$



Initial Convergence RLS (3)

- let $\eta \ll \sigma_k^2$ be small, then

$$\begin{aligned}
 \eta(\eta I + R_{1:k})^{-1} &= \eta(\eta V^\perp V^{\perp T} + \eta V V^T + V \Sigma^2 V^T)^{-1} \approx \eta(\eta V^\perp V^{\perp T} + V \Sigma^2 V^T)^{-1} \\
 &= \eta \left([V \ V^\perp] \begin{bmatrix} \Sigma^2 & 0 \\ 0 & \eta I \end{bmatrix} [V \ V^\perp]^T \right)^{-1} = \eta [V \ V^\perp] \begin{bmatrix} \Sigma^2 & 0 \\ 0 & \eta I \end{bmatrix}^{-1} [V \ V^\perp]^T \\
 &= \eta V \Sigma^{-2} V^T + V^\perp V^{\perp T} = \eta R_{1:k}^+ + \mathbf{P}_{R_{1:k}}^\perp
 \end{aligned}$$

where $R_{1:k}^+ = Y_{1:k} (Y_{1:k}^T Y_{1:k})^{-1} Y_{1:k}^T$ and $\mathbf{P}_{R_{1:k}} = \mathbf{P}_{Y_{1:k}}$

- hence

$$C_k = (\eta R_{1:k}^+ + \mathbf{P}_{R_{1:k}}^\perp) \widetilde{H}_0 \widetilde{H}_0^T (\eta R_{1:k}^+ + \mathbf{P}_{R_{1:k}}^\perp)$$

$\eta R_{1:k}^+ \widetilde{H}_0$ reduced to $\mathcal{O}(\eta)$ in k -dim. subspace, column space of $Y_{1:k}$

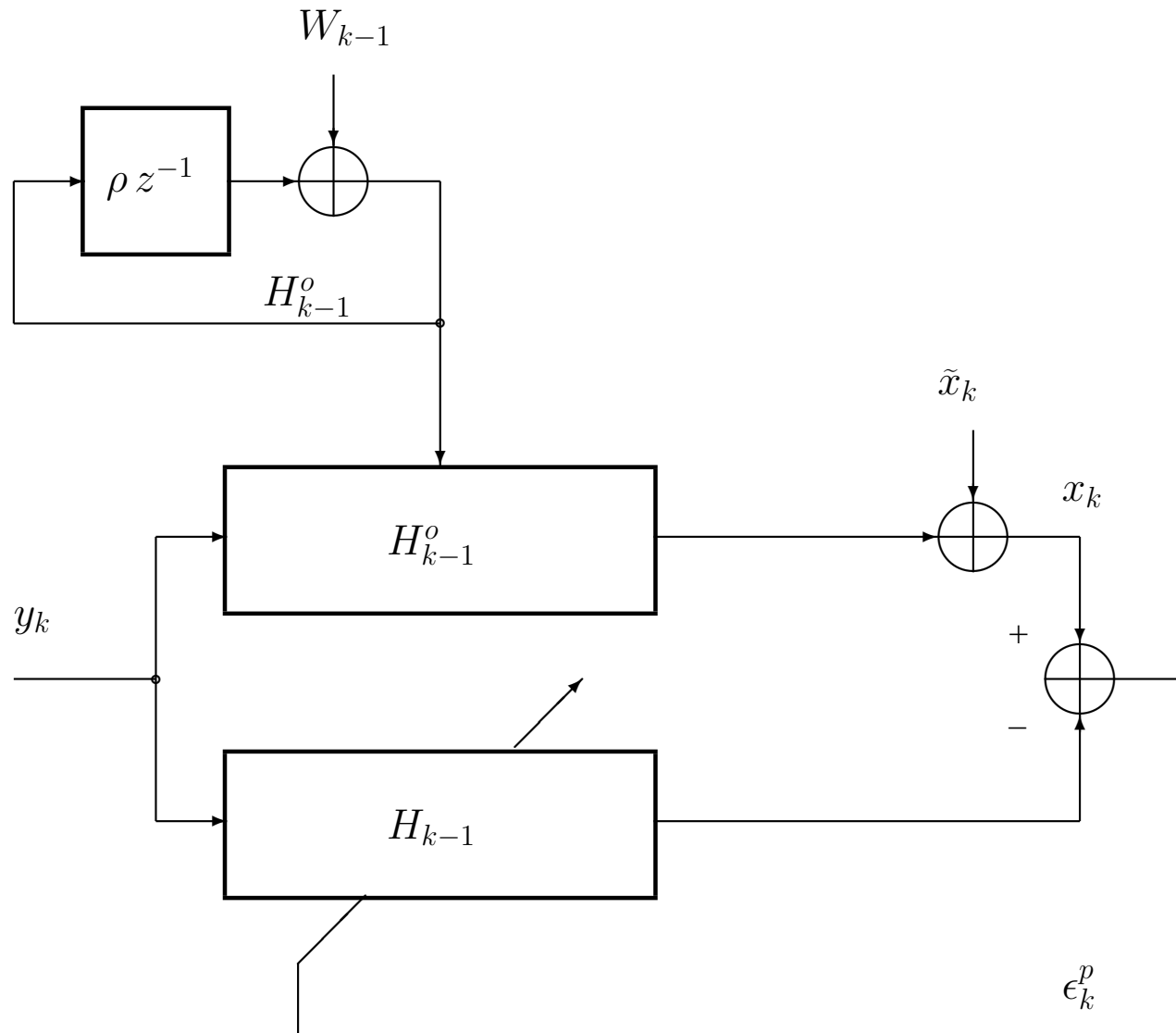
$\mathbf{P}_{R_{1:k}}^\perp \widetilde{H}_0$ unchanged in $(N - k)$ -dim. orthogonal complement

- C_k rank 1 (noiseless case): only the mean of \widetilde{H}_k needs to converge
- RLS: the mean of \widetilde{H}_k has essentially converged (filter estimate unbiased) after $k = N \Rightarrow$ very fast (mean dominates initial convergence in general)

LMS: the mean needs to converge exponentially, dynamics of steepest-descent



Tracking Time-Varying Filters





Time-Varying System Identification Set-Up

- system processes:

$$\begin{aligned} x_k &= Y_k^T H_{k-1}^o + \tilde{x}_k & E \tilde{x}_k \tilde{x}_i &= \xi^o \delta_{ki} \\ H_k^o &= \rho H_{k-1}^o + W_k & E W_k W_i^T &= Q \delta_{ki} \\ \widetilde{H}_k &= H_k^o - H_k & E W_k \tilde{x}_i &= 0 \end{aligned}$$

- time-varying filter modeled as AR(1) process, requires $|\rho| < 1$ for stationarity
→ stationary case of nonstationarity
- adaptive filter a priori error signal:

$$\epsilon_k^p = x_k - Y_k^T H_{k-1} = Y_k^T \widetilde{H}_{k-1} + \tilde{x}_k$$

- learning curve: (independence assumption)

$$\xi_k = E(\epsilon_k^p)^2 = \xi^o + \xi_k^e = \xi^o(1 + \mathcal{M}) , \quad \xi_k^e = \text{tr}\{R_{YY} C_{k-1}\} , \quad C_k = E \widetilde{H}_k \widetilde{H}_k^T$$

- consider $\frac{1}{1-\rho} \gg$ adaptation time constants so that we can take $\rho = 1$ for the analysis



Tracking Analysis LMS

- filter deviation recursion:

$$\begin{aligned}\tilde{H}_k &= \tilde{H}_{k-1} - \mu \epsilon_k^p Y_k + W_k \\ &= (I - \mu Y_k Y_k^T) \tilde{H}_{k-1} - \mu \tilde{x}_k Y_k + W_k \\ &\approx (I - \mu R_{YY}) \tilde{H}_{k-1} - \mu \tilde{x}_k Y_k + W_k\end{aligned}$$

where we introduced the averaging approach in the last step

- filter error correlation matrix recursion:

$$C_k = (I - \mu R_{YY}) C_{k-1} (I - \mu R_{YY}) + \mu^2 \xi^o R_{YY} + Q$$

- the stationary nonstationarity combined with a constant stepsize leads to a steady-state, for which we get (with small μ):

$$R_{YY} C_\infty + C_\infty R_{YY} = \mu \xi^o R_{YY} + \frac{1}{\mu} Q$$

- steady-state misadjustment: $\mathcal{M}_{LMS} = \underbrace{\frac{\mu}{2} \text{tr} R_{YY}}_{\text{estimation noise}} + \underbrace{\frac{1}{2\mu \xi^o} \text{tr} Q}_{\text{lag noise}}$



Tracking Analysis RLS

- filter deviation recursion:

$\lambda < 1$ to allow tracking

$$\begin{aligned}\tilde{H}_k &= \tilde{H}_{k-1} - R_k^{-1} Y_k \epsilon_k^p + W_k \\ &= (I - R_k^{-1} Y_k Y_k^T) \tilde{H}_{k-1} - R_k^{-1} Y_k \tilde{x}_k + W_k\end{aligned}$$

- after averaging in steady-state, assuming small $1 - \lambda$

$$(I - R_k^{-1} Y_k Y_k^T) = \lambda R_k^{-1} R_{k-1} \approx \lambda I, \quad R_k^{-1} \approx (1 - \lambda) R_{YY}^{-1}$$

$$\tilde{H}_k = \lambda \tilde{H}_{k-1} - (1 - \lambda) R_{YY}^{-1} Y_k \tilde{x}_k + W_k \quad (\text{dynamics indep. of } R_{YY})$$

- filter error correlation matrix recursion:

$$C_k = \lambda^2 C_{k-1} + (1 - \lambda)^2 \xi^o R_{YY}^{-1} + Q$$

- which leads to the steady-state value (assuming small $1 - \lambda$)

$$C_\infty = \frac{1 - \lambda}{2} \xi^o R_{YY}^{-1} + \frac{1}{2(1 - \lambda)} Q$$

- steady-state misadj.: $\mathcal{M}_{RLS} = \underbrace{\frac{1 - \lambda}{2} N}_{\text{estimation noise}} + \underbrace{\frac{1}{2(1 - \lambda) \xi^o} \text{tr}\{R_{YY} Q\}}_{\text{lag noise}}$



Tracking Optimization & LMS-RLS Comparison

- stepsize μ , $1 - \lambda$ design result of compromise between:
 estimation noise: finite stepsize prevents convergence, consistency
 lag noise: small stepsize leads to lowpass filtering and to a filter estimate that lags behind the true filter

- LMS: $\mu^{opt} = \sqrt{\frac{\text{tr}Q}{\xi^o \text{tr}R_{YY}}}$, $\mathcal{M}_{LMS}^{opt} = \sqrt{\frac{\text{tr}R_{YY} \text{tr}Q}{\xi^o}}$

- RLS: $\lambda^{opt} = 1 - \sqrt{\frac{\text{tr}\{R_{YY}Q\}}{N \xi^o}}$, $\mathcal{M}_{RLS}^{opt} = \sqrt{\frac{N \text{tr}\{R_{YY}Q\}}{\xi^o}}$

- comparison:

$$\frac{\mathcal{M}_{LMS}^{opt}}{\mathcal{M}_{RLS}^{opt}} = \sqrt{\frac{\text{tr}R_{YY} \text{tr}Q}{N \text{tr}\{R_{YY}Q\}}}$$

$$Q = \begin{cases} q I & : \text{equal performance, at least for small } q \\ q R_{YY} & : \text{LMS is better} \\ q R_{YY}^{-1} & : \text{RLS is better} \end{cases}$$

- faster initial convergence of RLS could be exploited for *jumping* parameters, if window size properly adapted



Optimal Tracking via Kalman Filtering

- *state-space model*: state = AR(1) vector process

$$\begin{aligned} \text{state equation} \quad H_k^o &= A_k H_{k-1}^o + W_k \\ \text{measurement equation} \quad x_k &= Y_k^T H_{k-1}^o + \tilde{x}_k \end{aligned}, \quad E \begin{bmatrix} W_k \\ \tilde{x}_k \end{bmatrix} \begin{bmatrix} W_i \\ \tilde{x}_i \end{bmatrix}^T = \begin{bmatrix} Q_k & P_k^T \\ P_k & \xi_k^o \end{bmatrix} \delta_{ki}$$

time-varying (at the very least due to Y_k), usually $P_k = 0$

state noise: W_k , measurement noise: \tilde{x}_k , state transition matrix A_k

- Kalman filter (KF): estimates recursively in time the *state* H_{k-1}^o on the basis of the *measurements* x_0, \dots, x_k in a LMMSE sense.
Sources of randomness: W_k, \tilde{x}_k . Signal y_k treated as deterministic.
- special case: $H_k^o = H_{k-1}^o \Rightarrow$ Kalman filter \rightarrow RLS algorithm
- RLS with exponential weighting can be interpreted as KF for the case of some non-zero Q_k
- in the time-invariant case (x_k and H_k^o jointly stationary apart from initial conditions), the KF converges to the causal Wiener filter.



Chapter 5: Sinusoids in White Noise

- $x_k = \sum_{i=1}^M A_i \cos(\omega_i k + \phi_i), \quad y_k = x_k + v_k \quad \omega_i = 2\pi f_i$
- the support of $S_{xx}(f) = \sum_{i=1}^M \frac{A_i^2}{4} (\delta(f - f_i) + \delta(f + f_i))$ has measure zero,
 R_{XX} is singular for dimension $> 2M$
- $P(q) x_k = 0, \quad P(z) = \prod_{i=1}^M (1 - 2 \cos \omega_i z^{-1} + z^{-2}), \quad q^{-1} x_k = x_{k-1}$
- Hence, x_k is perfectly predictable from the previous $2M$ samples. $P(z)$ and hence the ω_i can be found by linear prediction: *Prony* method.
(Baron Prony, 18th century)
Normal equations:

$$P R_{XX} = [0 \cdots 0 \sigma^2], \quad \sigma^2 = 0$$

where

$$\begin{aligned} R_{XX} &= E X X^T \\ X &= [x_0 \cdots x_{2M}]^T \\ P &= [P_{2M} \cdots P_1 \ P_0] \\ P_0 &= 1 \\ P_i &= P_{2M-i}, \quad i = 0, \dots, M-1 \end{aligned}$$



Sinusoids in Noise: Signal and Noise Subspaces

- signal structure

$$X_k = \begin{bmatrix} x_0 \\ x_1 \\ \vdots \\ x_k \end{bmatrix} = \begin{bmatrix} 1 & 0 & \cdots & 1 & 0 \\ \cos \omega_1 & \sin \omega_1 & \cdots & \cos \omega_M & \sin \omega_M \\ \vdots & \vdots & \vdots & \vdots & \vdots \\ \cos \omega_1 k & \sin \omega_1 k & \cdots & \cos \omega_M k & \sin \omega_M k \end{bmatrix} \begin{bmatrix} A_1 \cos \phi_1 \\ -A_1 \sin \phi_1 \\ \vdots \\ A_M \cos \phi_M \\ -A_M \sin \phi_M \end{bmatrix} = \mathcal{V} S$$

- one calls

$$\begin{aligned} \text{Range} \{ \mathcal{V} \} &= \text{signal subspace} \\ (\text{Range} \{ \mathcal{V} \})^\perp &= \text{noise subspace} \end{aligned}$$

- covariance structure:

$$Y_k = X_k + V_k = \mathcal{V}_k S + V_k \Rightarrow R_{YY} = \mathcal{V} R_{SS} \mathcal{V}^T + \sigma_v^2 I$$

if angles uniform and uncorrelated:

$$R_{SS} = \frac{1}{2} \text{diag} \{ A_1^2, A_1^2, \dots, A_M^2, A_M^2 \}$$



Sinusoids in Noise: Signal and Noise Subspaces (2)

- Consider the eigendecomposition of R_{YY} ($\lambda_1 \geq \lambda_2 \geq \dots$):

$$R_{YY} = \sum_{i=1}^{2M} \lambda_i V_i V_i^T + \sum_{i=2M+1}^{k+1} \lambda_i V_i V_i^T = V_S \Lambda_S V_S^T + V_N \Lambda_N V_N^T$$

where $\Lambda_N = \sigma_v^2 I_{k+1-2M}$.

- Assuming \mathcal{V} and R_{SS} to have full rank, the sets of eigenvectors V_S and V_N are orthogonal: $V_S^T V_N = 0$, and $\lambda_i > \sigma_v^2$, $i = 1, \dots, 2M$.
- Equivalent descriptions of the signal and noise subspaces:

$$\text{Range}\{V_S\} = \text{Range}\{\mathcal{V}\} \quad , \quad V_N^T \mathcal{V} = 0$$

- Linear prediction in the noisy case: minimize variance subject to norm constraint: *Pisarenko* method
with $k = 2M$: noise subspace dimension = 1

$$\min_{\|P\|=1} P R_{YY} P^T = \min_{\|P\|=1} P R_{XX} P^T + \sigma_v^2 \Rightarrow P R_{XX} = [0 \cdots 0], \quad P^T = V_{2M+1}$$



Sinusoids in Noise: Signal Subspace Fitting

- two equivalent signal subspace descriptions: \mathcal{V} and V_S
- with an estimated covariance matrix, V_S is approximate, so consider

$$\min_{\omega, T} \|\mathcal{V}(\omega) - V_S T\|_F^2 \quad \|A\|_F^2 = \text{tr } AA^T$$

where $\omega = [\omega_1 \cdots \omega_M]$.

- separable problem: orthogonality of LS: $V_S^T(\mathcal{V} - V_S T) = 0 \Rightarrow T = V_S^T \mathcal{V}$
- $\mathcal{V} - V_S T = (I - V_S V_S^T) \mathcal{V} = (I - P_{V_S}) \mathcal{V} = P_{V_S}^\perp \mathcal{V}$
- projection on column space of X : $P_X = X(X^T X)^{-1} X^T$, $P = P^T$, $PP = P$

$$\|P_{V_S}^\perp \mathcal{V}\|_F^2 = \text{tr } \mathcal{V}^T P_{V_S}^\perp \mathcal{V} = \text{tr } \mathcal{V}^T P_{V_N} \mathcal{V} = \|V_N^T \mathcal{V}\|_F^2$$

- hence multi-D optim.

$$= \sum_{i=2M+1}^k \|V_i^T \mathcal{V}\|^2 = \sum_{j=1}^M \sum_{i=2M+1}^k |V_i(\omega_j)|^2$$

- approximate solution: plot as a function of ω and find M largest peaks of

$$\frac{1}{\sum_{i=2M+1}^k |V_i(\omega)|^2} \quad \text{MUSIC!}$$



Sinusoids in Noise: Noise Subspace Parameterization

- $P(q) \cos \omega_i k = 0, P(q) \sin \omega_i k = 0, \Rightarrow \mathcal{G}(P)^T \mathcal{V} = 0$ where

$$\mathcal{G}^T(P) = \begin{bmatrix} P_0 & P_1 & \cdots & P_{2M} & 0 & \cdots & 0 \\ 0 & P_0 & P_1 & \cdots & P_{2M} & \cdots & 0 \\ \vdots & \ddots & \ddots & \ddots & & \ddots & \vdots \\ 0 & \cdots & 0 & P_0 & P_1 & \cdots & P_{2M} \end{bmatrix} \quad \text{Toeplitz, } (k-2M) \times (k+1)$$

- noise subspace fitting:

$$\min_{P, T} \|\mathcal{G}(P) - V_{\mathcal{N}} T\|_F$$

- separable problem $\Rightarrow T = V_{\mathcal{N}}^T \mathcal{G}, \mathcal{G} - V_{\mathcal{N}} T = \mathbf{P}_{V_{\mathcal{N}}}^\perp \mathcal{G}$ and hence

$$\begin{aligned} \|\mathbf{P}_{V_{\mathcal{N}}}^\perp \mathcal{G}\|_F^2 &= \text{tr} \mathcal{G}^T \mathbf{P}_{V_{\mathcal{N}}}^\perp \mathcal{G} = \text{tr} \mathcal{G}^T \mathbf{P}_{V_{\mathcal{S}}} \mathcal{G} = \|V_{\mathcal{S}}^T \mathcal{G}\|_F^2 \\ &= \sum_{i=1}^{2M} \|\mathcal{G}^T V_i\|^2 \end{aligned}$$

Let $\mathcal{G}^T V_i = \mathcal{W}_i P^T$ where $\mathcal{W}_i = \mathcal{W}(V_i)$ is Hankel, then we get (with $P = PJ$)

$$\min_P P \left[\left(\sum_{i=1}^{2M} \mathcal{W}_i^T \mathcal{W}_i \right) + J \left(\sum_{i=1}^{2M} \mathcal{W}_i^T \mathcal{W}_i \right) J \right] P^T$$

subject to $P_0 = 1$ or $\|P\| = 1$.



Sinusoids in Noise: Maximum Likelihood Estimation

- additive noise v_k white and Gaussian \rightarrow likelihood criterion

$$\min_{\omega, S} \|Y - \mathcal{V}(\omega) S\|^2$$

- separable: $\mathcal{V}^T(Y - \mathcal{V} S) = 0 \Rightarrow S = (\mathcal{V}^T \mathcal{V})^{-1} \mathcal{V}^T Y$

$$\Rightarrow \|Y - \mathcal{V} S\|^2 = Y^T \mathbf{P}_{\mathcal{V}}^\perp Y = Y^T \mathbf{P}_{\mathcal{G}(P)} Y = P \mathcal{Y}^T (\mathcal{G}(P)^T \mathcal{G}(P))^{-1} \mathcal{Y} P^T$$

where $\mathcal{G}(P)Y = \mathcal{Y}(Y) P^T$ (commutativity of convolution, \mathcal{Y} Hankel)

- IQML (Iterative Quadratic Maximum Likelihood), iteration n :

$$\min_{P^{(n)}} P^{(n)} \mathcal{Y}^T (\mathcal{G}(P^{(n-1)})^T \mathcal{G}(P^{(n-1)}))^{-1} \mathcal{Y} P^{(n)T}$$

subject to $P_0 = 1$ or $\|P\| = 1$

- with a consistent initialization, only one iteration is required to get a BAN estimate at high SNR

- denoised IQML : $Y^T \mathbf{P}_{\mathcal{G}(P)} Y = \text{tr} \{ \mathbf{P}_{\mathcal{G}(P)} Y Y^T \} \rightarrow \text{tr} \{ \mathbf{P}_{\mathcal{G}(P)} (Y Y^T - \hat{\sigma}_v^2 I) \}$

- Pseudo-QML (PQML): $\frac{\partial}{\partial P} Y^T \mathbf{P}_{\mathcal{G}(P)} Y = 2Q(P)P \rightarrow 2Q(\hat{P})P = \frac{\partial}{\partial P} P^T Q(\hat{P}) P$



Sinusoids in Noise: Adaptive Notch Filtering

- notch filter model

$$P(q)x_k = 0 \Rightarrow P(q)y_k = P(q)v_k \Rightarrow v_k = \frac{P(q)}{P(q/\rho)} y_k \text{ as } \rho \rightarrow 1$$

- notch filter output

$$\epsilon_k = H(q) y_k = H(q)x_k + H(q)v_k, \quad H(q) = \frac{P(q)}{P(q/\rho)}$$

- notch filter $H(z)$: zeros = $e^{\pm j\omega_i}$, poles = $\rho e^{\pm j\omega_i}$,

- notch filter output variance ($\mathbf{H}(f) = H(e^{j2\pi f})$)

$$E \epsilon_k^2 = \int_{-\frac{1}{2}}^{\frac{1}{2}} |\mathbf{H}(f)|^2 S_{xx}(f) df + \int_{-\frac{1}{2}}^{\frac{1}{2}} |\mathbf{H}(f)|^2 S_{vv}(f) df = \sum_{i=1}^M \frac{A_i^2}{2} |\mathbf{H}(f_i)|^2 + \sigma_v^2$$

- notch adaptation by output variance minimization

$$\min_P E \epsilon_k^2$$

adaptively: Recursive Prediction Error Method (RPEM): solve ML recursively