Rational Minimax Filtering

Arthur J. Krener Wei Kang

ajkrener@nps.edu wkang@nps.edu

Research supported in part by AFOSR and NSF

Dedicated to our Esteemed Colleague Eduardo Sontag on the occasion of his 60^{th} birthday

We assume that the dynamics and measurement processes can be modeled by a linear system

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measurements

$$egin{array}{rcl} y(t) &=& Cx(t) \ \dot{y}(t) &=& CAx(t) \ \ddot{y}(t) &=& CA^2x(t) \ & & & \cdot \end{array}$$

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What is standard white Gaussian noise? It is the formal derivative of a standard Weiner process and is mathematically characterized by the following properties.

If $f(t) \in L^2([t_1, t_2], I\!\!R^m)$ then the random variable

$$X \;\;=\;\; \int_{t_1}^{t_2} f'(t) w(t) \;\, dt$$

is Gaussian with zero mean and variance

$$\mathrm{E}(X^2) \;\; = \;\; \int_{t_1}^{t_2} \|f(t)\|^2 \; dt$$

Why white Gaussian noise? There are several possible answers.

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There are generalizations and extensions to handle the following.

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We assume that the filter for $x_i(t)$ is a weighted sum of the past observations. The estimate is

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We wish to choose the weighing pattern $k(s) \in \mathbb{R}^{1 \times p}$ to minimize $E(\tilde{x}_i(t))^2$ where $\tilde{x}_i(t) = x_i(t) - \hat{x}_i(t)$.

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$$egin{array}{rcl} \dot{h}&=&hA+kC\ h(0)&=&-e^i \end{array}$$

where e^i is the i^{th} unit row vector.

$$\begin{aligned} \hat{x}_{i}(t) &= \int_{0}^{\infty} k(s)y(t-s) \ ds \\ &= \int_{0}^{\infty} k(s)Cx(t-s) + k(s)Dw(t-s) \ ds \\ &= \int_{0}^{\infty} \left(\dot{h}(s) - h(s)A\right)x(t-s) + k(s)Dw(t-s) \ ds \\ &= \left[h(s)x(t-s)\right]_{0}^{\infty} + \int_{0}^{\infty} h(s)Bv(t-s) + k(s)Dw(t-s) \ ds \end{aligned}$$

We assume that $h(\infty)=0$ so

$$\tilde{x}_i(t) = -\int_0^\infty h(s)Bv(t-s) + k(s)Dw(t-s) ds$$
$$E(\tilde{x}_i(t))^2 = \int_0^\infty h(s)BB'h'(s) + k(s)DD'k'(s) ds$$

Linear Quadratic Regulator

So we have the optimal control problem of minimizing by choice of k(s)

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We assume that the minimum is a quadratic form in h^0

$$h^0 P(h^0)' = \min_k \int_0^\infty h(s) BB' h'(s) + k(s) DD' k'(s) \ ds$$

$$h^0 P(h^0)' = \min_k \int_0^\infty h B B' h' + k D D' k' ds$$

$$h^0 P(h^0)' = \min_k \int_0^\infty hBB'h' + kDD'k' \, ds$$

$$[h(s)Ph'(s)]_0^\infty = \int_0^\infty \frac{d}{ds} h(s)Ph'(s) ds h^0 P(h^0)' = -\int_0^\infty (hA + kC) Ph' + hP (hA + kC)' ds$$

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Subtracting

$$0 = \min_{k} \int_{0}^{\infty} [h,k] \begin{bmatrix} AP + PA' + BB' & PC' \\ CP & DD' \end{bmatrix} [h,k]' ds$$

lf

$$0 = AP + PA' + BB' - PC'(DD')^{-1}CP G = PC'(DD')^{-1}$$

then the above reduces to a perfect square

$$0 = \min_{k} \int_0^\infty (k + hG) DD'(k + hG)' \, ds$$

so the optimal k = -hG .

To filter all states at once we let $H(s) \in I\!\!R^{n \times n}$ satisfy

$$\dot{H} = H(A - GC)$$

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and K(s) = H(s)G then

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This derivation is easily extended to discrete time, time varying and/or finite horizon linear systems.

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$$\min_k \max_{|u| \le 1} E_w(\tilde{x}(t))^2$$

Johansen and Berkovitz-Pollard Problem Given a k(s) define h(s) by

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 $\label{eq:constraint} Johansen \ and \ Berkovitz-Pollard \ Problem \\ \textbf{Then}$

$$E_w(ilde{x}(t))^2 \;\; = \;\; \left(\int_0^\infty h(s) u(s) \;\, ds
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and we have a differential game.

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Clearly for a given k(s), h(s) , the maximizing u(s) are

$$u(s) = \pm \operatorname{sign}(h(s))$$

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The differential game reduces to a non standard optimal control of choosing k(s), h(s) to minimize this quantity subject to

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The Euler Lagrange equation for this problem is

$$h^{(4)} = -\gamma \operatorname{sign}(h)$$

where

$$\gamma ~=~ \int_0^\infty |h(s)|~ds$$

Consider the related differential equation

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Johansen and Berkovitz-Pollard Problem Consider the related differential equation

$$\phi^{(4)}~=~- ext{sign}(\phi)$$

Two one parameter groups act on the space of solutions of this equation.

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We look for a self similar solution that has consecutive simple zeros at s = 0, s = 1 and satisfies for $s \in [0, \alpha]$

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$$\phi(s+1) = -\alpha^4 \phi(s/\alpha)$$

On $s \in [0,1]$

$$\phi(s) = c_1 s + c_2 s^2 / 2 + c_3 s^3 / 6 + c_4 s^4 / 24$$

where $c_4 = -\operatorname{sign}(c_1) \neq 0$

Johansen and Berkovitz-Pollard Problem Matching $\phi(s)$ and its first three derivatives at $s = 1^{\pm}$ we obtain

$$\left[egin{array}{c} 0 \ 0 \ 0 \ 0 \ 0 \end{array}
ight] \;=\; \left[egin{array}{c} 1 & 1/2! & 1/3! & 1/4! \ 1+lpha^3 & 1 & 1/2! & 1/3! \ 0 & 1+lpha^2 & 1 & 1/2! \ 0 & 0 & 1+lpha & 1 \end{array}
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so the determinant of this matrix must be zero.

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The first and third roots yield self similar solutions to $\phi^{(4)} = -\operatorname{sign}(\phi)$ while the second root yields a periodic solution to $\phi^{(4)} = \operatorname{sign}(\phi)$.

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Then

$$h(s) ~=~ \gamma eta^4 \phi(s/eta)$$

where β is chosen so that

$$1 \hspace{.1in} = \hspace{.1in} \int_0^\infty |eta^4 \phi(s/eta)| \hspace{.1in} ds$$

Then γ is chosen so that

$$h(0) = -1$$

For $s \in [0,\beta]$

- $\begin{array}{lll} h(s) &=& -s + 0.872575492926169 s^2 0.253795996951782 s^3 \\ &+ 0.024616157365051 s^4 \end{array}$
- $\begin{aligned} k(s) &= 1.745150985852338 1.522775981710693s \\ &+ 0.295393888380611s^2 \end{aligned}$

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And what about a general linear system?

Linear Time Invariant Minimax Filtering

Plant:

$$egin{array}{lll} \dot{x} = Ax + Bu, & \|u\|_{\infty} \leq 1 \ y = Cx + Dw, & w \ extsf{WGN} \ z = Lx, & z \in I\!\!R \end{array}$$

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Linear Filter:

$$\hat{z} ~=~ \int_0^\infty k(s) y(t-s) ~ds$$

Goal:

$$\min_k \max_{\|u\|_{\infty} \leq 1} \mathsf{E}_w(\tilde{x}_i)^2$$

Linear Time Invariant Minimax Filtering

Given a k(s) define h(s) as before

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After integration by parts

$$\begin{split} \tilde{z}(t) &= \int_0^\infty h(s) Bu(t-s) + k(s) Dw(t-s) \ ds \\ \mathsf{E}_w(\tilde{z}(t))^2 &= \left(\int_0^\infty h(s) Bu(t-s) \ ds \right)^2 \\ &+ \int_0^\infty k(s) DD'k'(s) \ ds \\ \max_{\|\|u\|_\infty \le 1} \mathsf{E}_w(\tilde{z}(t))^2 &= \left(\int_0^\infty \|h(s)B\|_1 \ ds \right)^2 \\ &+ \int_0^\infty k(s) DD'k'(s) \ ds \end{split}$$

Non Standard Optimal Control Problem

Minimize

$$\left(\int_0^\infty \|h(s)B\|_1 \ ds\right)^2 + \int_0^\infty k(s)DD'k'(s) \ ds$$

subject to

$$\dot{h} = hA + kC$$

 $h(0) = -L$

• State
$$h(s) \in {I\!\!R}^{1 imes n}$$
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This optimization problem is too complicated for the Euler-Lagrange approach so we apply the Pontryagin Maximum Principle instead.

Add an extra state coordinate

 $\dot{h}_{n+1} = \|hB\|_1$

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Control Hamiltonian

$$\mathcal{H} = hA\xi + kC\xi + \|hB\|_1\zeta + kDD'k$$

Adjoint Dynamics

$$\dot{\xi} = -\left(rac{\partial \mathcal{H}}{\partial h}
ight)' = -A\xi - B\left(\operatorname{sign}(hB)
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 $\dot{\zeta} = -\left(rac{\partial \mathcal{H}}{\partial h_{n+1}}
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Maximize the Hamiltonian with respect to the control

$$\begin{array}{rcl} 0 & = & \displaystyle \frac{\partial \mathcal{H}}{\partial k} = C\xi + 2DD'k' \\ \\ k & = & \displaystyle -\frac{\xi'C'(DD')^{-1}}{2} \end{array}$$

and plug into the dynamics.

Hamiltonian Dynamics and Transversality Conditions

$$\dot{h} = hA - \frac{\xi' C' (DD')^{-1} C}{2}$$

$$\dot{h}_{n+1} = \|hB\|_1$$

$$\dot{\xi} = -A\xi - B (\operatorname{sign}(hB))' \zeta$$

$$\dot{\zeta} = -2\|hB\|_1$$

$$h(0) = -L$$

 $h_{n+1}(0) = 0$
 $\xi(\infty) = 0$
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This is usually too complicated to solve explicitly and even if we could the resulting filter would probably be infinite dimensional.

Therefore we restrict the optimization to weighing patterns k(s) that are the impulse responses of finite dimensional linear systems.

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In other words we restrict to k(s) whose Laplace transforms are rational.

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$$k(s) \;\; = \;\; \sum_{i=1}^N \gamma_i e^{\lambda_i s}$$

This guarantees that the resulting filter is finite dimensional, it can be realized by a finite dimensional time invariant linear system.

$$egin{array}{rcl} \hat{z}(t)&=&\int_{0}^{\infty}k(s)y(t-s)\;ds\ k(s)&=&\sum_{i=1}^{N}\gamma_{i}e^{\lambda_{i}s} \end{array}$$

is realized by

$$\dot{\xi} = egin{bmatrix} \lambda_1 & 0 \ & \ddots & \ 0 & \lambda_N \end{bmatrix} \xi + egin{bmatrix} 1 & 0 \ & \ddots & \ 0 & 1 \end{bmatrix} y$$
 $\hat{z}(t) = egin{bmatrix} \gamma_1 & \ldots & \gamma_N \end{bmatrix} \xi$

If we look for a filter the same size as the original system N=n, A,B is a controllable pair and all the eigenvalues of A are in the closed right half plane then the filter takes the form

$$k(s) = -h(s)G$$

for some G .

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In other words we are finding the linear feedback that

$$\min_{G} \left(\int_0^\infty \|h(s)B\|_1 \ ds \right)^2 + \int_0^\infty k(s)DD'k'(s) \ ds$$

subject to

$$\dot{h} = hA + kC$$

 $h(0) = -L$
 $k(s) = -h(s)G$

One virtue of this approach is that the resulting filter is realized by the linear system

$$\dot{\xi} = (A - GC)\xi + Gy = A\xi + G(y - C\xi)$$

 $\hat{z} = L\xi$

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Notice that there may be a different gain G and different filter for each linear functional of the state z = Lx.

This suggests the following approach. Use numerical routines to minimize the optimal control problem with and without the restriction that k(s) = h(s)G. If the optimal cost of the former is close enough to that of the latter, accept the filter. If not expand the class of rational filters that are considered.

Single Integrator

We tried this approach on some model problems.

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$$A = 0 \qquad B = 1$$

C = 1 D = 1

$$z = x$$

Optimal Cost	Suboptimal Rational Cost	Ratio
1.1006	1.1906	1.0818

We were able to compute the optimal infinite dimensional filter explicitly. The subantimal filter was computed using a numerical

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Double Integrator

$$A = \left[egin{array}{cc} 0 & 1 \ 0 & 0 \end{array}
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(JBP Problem)

 $z = x_1$

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1.7452	1.7880	1.0245

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Optimal Cost	Suboptimal Rational Cost	Ratio
1.7452	1.7880	1.0245

 $z = x_2$

Optimal Cost	Suboptimal Rational Cost	Ratio
2.1269	2.2733	1.0688

Again we were able to compute the optimal infinite dimensional filters explicitly. The suboptimal filters were computed using a numerical optimization routine.

Triple Integrator

Estimate x_1

$$A = \begin{bmatrix} 0 & 1 & 0 \\ 0 & 0 & 1 \\ 0 & 0 & 0 \end{bmatrix} \qquad B = \begin{bmatrix} 0 \\ 0 \\ 1 \end{bmatrix}$$
$$C = \begin{bmatrix} 1 & 0 & 0 \end{bmatrix} \qquad D = \begin{bmatrix} 1 \end{bmatrix}$$

 $z = x_1$

Approx. Optimal Cost	Suboptimal Rational Cost	Ratio
2.4074	2.4282	1.009

We computed the optimal filter and the suboptimal filter using numerical optimization routines.

Quadruple Integrator

Estimate x_1

$$A = \begin{bmatrix} 0 & 1 & 0 & 0 \\ 0 & 0 & 1 & 0 \\ 0 & 0 & 0 & 1 \\ 0 & 0 & 0 & 0 \end{bmatrix} \qquad B = \begin{bmatrix} 0 \\ 0 \\ 0 \\ 1 \end{bmatrix}$$
$$C = \begin{bmatrix} 1 & 0 & 0 & 0 \end{bmatrix} \qquad D = \begin{bmatrix} 1 \end{bmatrix}$$

 $z = x_1$

Approx. Optimal Cost	Suboptimal Rational Cost	Ratio
3.0722	3.0901	1.006

We computed the optimal filter and the suboptimal filter using numerical optimization routines.

Harmonic Oscillator

Estimate x_1

$$A = \left[egin{array}{ccc} 0 & -1 \ 1 & 0 \end{array}
ight] \qquad B = \left[egin{array}{ccc} 0 \ 1 \end{array}
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ight] \qquad D = \left[egin{array}{cccc} 1 \end{array}
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$$z = x_1$$

Approx. Optimal Cost	Suboptimal Cost	Ratio
1.26	1.3536	1.07

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- Happy Birthday Eduardo