

T-61.3040 Statistical Signal Modeling

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Today's Topics (11.10)

- Signal Modeling (about time...)
- Least Squares Method
- Padé Approximation
- Prony's Method
- Modeling for finite sequences



Signal Modeling

- Why model signals?
- Think about $x(n) = A\sin(n\omega_0 + \phi)$, the Harmonic process
- A bit less data in transfering (A, ω_0, ϕ) and reconstructing at destination than sending all values
- Not obvious which is best: tradeoff between accuracy and efficiency (for general case, where x(n) has no exact expression)

Signal Modeling

- Also for prediction: if you know x(0), ..., x(N-1) and want x(N): modeling x(n) with $n \in [0, N-1]$ gives possibility to predict (how to do it accurately is the problem...)
- Two steps in modeling:
 - Choose an "appropriate" parametric form for the model
 - Find the model parameters that best fit the data records you have



Signal Modeling

- How to do this? Usually, stick to models with well-known properties, such as ARMA type, sum of weighted sinusoids...
- Otherwise, finding model parameters might be intractable (remember you have to optimize a cost function to find them)
- Cost/Error of the modelisation: $e'(n) = x(n) \hat{x}(n)$
- Remember another type of cost?



Also called "Direct Method"

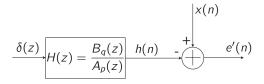


Figure: Block Diagram of the Direct Method

■ Consider modeling of a deterministic signal x(n) as the unit sample response of a LSI filter with rational system function of the form

$$H(z) = \frac{B_q(z)}{A_p(z)} = \frac{\sum_{k=0}^{q} b_q(k) z^{-k}}{1 + \sum_{k=1}^{p} a_p(k) z^{-k}}$$

Assume also that x(n) = 0 for n < 0 (on loss of generality, just shift the signal) and h(n) causal



Criterion to be minimized for this method is based on squares of error:

$$\varepsilon_{LS} = \sum_{n=0}^{\infty} |e'(n)|^2$$

Minimizing the LS cost with such a system function means

$$\begin{cases} \frac{\partial \epsilon_{LS}}{\partial a_p^*(k)} = 0, & \forall k \in [1, p] \\ \frac{\partial \epsilon_{LS}}{\partial b_q^*(k)} = 0 & \forall k \in [0, q] \end{cases}$$



Writing the expression of the LS error in the frequency domain, we get the set of equations (k = 1, ..., p) in the first, k = 0, ..., q in the second)

$$\begin{cases} \frac{\partial \epsilon_{LS}}{\partial a_{p}^{*}(k)} = \frac{1}{2\pi} \int_{-\pi}^{\pi} \left[X(e^{j\omega}) - \frac{B_{p}(e^{j\omega})}{A_{q}(e^{j\omega})} \right] \frac{B_{q}^{*}(e^{j\omega})}{\left[A_{p}^{*}(e^{j\omega}) \right]^{2}} e^{jk\omega} d\omega = 0 \\ \frac{\partial \epsilon_{LS}}{\partial b_{q}^{*}(k)} = \frac{1}{2\pi} \int_{-\pi}^{\pi} \left[X(e^{j\omega}) - \frac{B_{p}(e^{j\omega})}{A_{q}(e^{j\omega})} \right] \frac{e^{jk\omega}}{A_{p}^{*}(e^{j\omega})} d\omega = 0 \end{cases}$$

■ And we have ourselves p + q + 1 nonlinear equations



- Usually intractable, or needs iterative methods: Newton's method, steepest descent...
- We will use indirect methods instead of trying to solve the direct one



- In this case we will end up with a set of linear equations
- With same assumptions on x(n) and the system function (p poles and q zeros, same form as before, x(n) modeled as a unit response to a LSI causal filter)
- We have p + q + 1 degrees of freedom on the filter design
- So, we should be able to make the filter fit exactly the signal for p + q + 1 values out of n, right?



An example:

■ With a causal first order all pole filter such as

$$H(z) = \frac{b(0)}{1 + a(1)z^{-1}}$$

■ The unit sample response of the filter is then

$$h(n) = b(0) [-a(1)]^n u(n)$$



■ So if we want h(n) = x(n) for n = 0, 1, we need

$$\begin{cases} x(0) &= b(0) \\ x(1) &= -b(0)a(1) \end{cases}$$

■ Meaning, if $x(0) \neq 0$, that the model is

$$H(z) = \frac{x^2(0)}{x(0) - x(1)z^{-1}}$$

Another example, higher degree:

Have

$$H(z) = \frac{b(0) + b(1)z^{-1}}{1 + a(1)z^{-1}}$$

Leads to

$$\begin{cases} x(0) &= b(0) \\ x(1) &= -b(0)a(1) + b(1) \\ x(2) &= b(0)a^{2}(1) - b(1)a(1) \end{cases}$$



And finally, solving by substitution

$$\begin{cases} a(1) &= -\frac{x(2)}{x(1)} \\ b(0) &= x(0) \\ b(1) &= x(1) - x(0) \frac{x(2)}{x(1)} \end{cases}$$

assuming $x(1) \neq 0$

■ Did I lie? These last equations are not exactly linear...



- Padé Approximation aims at getting the right coefficients while reformulating the problem to only have linear equations to solve
- Start by expressing the system function

$$H(z) = \frac{B_q(z)}{A_p(z)} = \frac{\sum_{k=0}^{q} b_q(k) z^{-k}}{1 + \sum_{k=1}^{p} a_p(k) z^{-k}}$$

as

$$H(z)A_p(z) = B_q(z)$$



And express it in the time domain as the convolution

$$h(n) + \sum_{k=1}^{p} a_p(k)h(n-k) = b_q(n)$$

with h(n) = 0 for n < 0 and $b_q(n) = 0$ for n < 0 and n > q

Now we want x(n) = h(n) over [0, p + q], so we write the equations

$$x(n) + \sum_{k=1}^{p} a_p(k)x(n-k) = \begin{cases} b_q(n) &, n \in \llbracket 0, q \rrbracket \\ 0 &, n \in \llbracket q+1, q+p \rrbracket \end{cases}$$



Which will be handier in matrix form (why? later...)

$$\begin{bmatrix} x(0) & 0 & \cdots & 0 \\ x(1) & x(0) & \cdots & 0 \\ x(2) & x(1) & \cdots & 0 \\ \vdots & \vdots & \cdots & \vdots \\ \hline x(q+1) & x(q) & \cdots & x(q-p+1) \\ \vdots & \vdots & \ddots & \vdots \\ x(q+p) & x(q+p-1) & \cdots & x(q) \end{bmatrix} \begin{bmatrix} 1 \\ a_p(1) \\ a_p(2) \\ \vdots \\ a_p(p) \end{bmatrix} = \begin{bmatrix} b_q(0) \\ b_q(1) \\ b_q(2) \\ \vdots \\ b_q(q) \\ 0 \\ \vdots \\ 0 \end{bmatrix}$$



To solve this, we first solve the "lower part" of this system to get the $a_p(k)$:

$$\begin{bmatrix} x(q+1) & x(q) & \cdots & x(q-p+1) \\ x(q+2) & x(q+1) & \cdots & x(q-p+2) \\ \vdots & \vdots & \ddots & \vdots \\ x(q+p) & x(q+p-1) & \cdots & x(q) \end{bmatrix} \begin{bmatrix} 1 \\ a_p(1) \\ \vdots \\ a_p(p) \end{bmatrix} = \begin{bmatrix} 0 \\ 0 \\ 0 \end{bmatrix}$$

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Which can be expanded and rewritten as

$$\begin{bmatrix} x(q) & x(q-1) & \cdots & x(q-p+1) \\ x(q+1) & x(q) & \cdots & x(q-p+2) \\ \vdots & \vdots & \ddots & \vdots \\ x(q+p-1) & x(q+p-2) & \cdots & x(q) \end{bmatrix} \begin{bmatrix} a_p(1) \\ a_p(2) \\ \vdots \\ a_p(p) \end{bmatrix} = - \begin{bmatrix} x(q+1) \\ x(q+2) \\ \vdots \\ x(q+p) \end{bmatrix}$$

which we express as $\mathbf{X}_q \mathbf{a}_p = -\mathbf{x}_{q+1}$



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And \mathbf{X}_q is a nonsymmetric Toeplitz matrix, which means that depending on how \mathbf{X}_q is structured, we have solutions or not:

- **X**_q not singular: Then \mathbf{X}_q^{-1} exists and the solution is uniquely determined by $\mathbf{a}_p = -\mathbf{X}_q^{-1}\mathbf{x}_{q+1}$
- X_q singular and a solution exists: The solution is not unique. Might want to choose the one with smallest amount of non-zero values
- **X**_q singular and no solutions: We assumed in the model formulation, that $a_p(0) = 1$ (or was non-zero and could be normalized to be 1). Must be false, so we have in fact $a_p(0) = 0$, and then $\mathbf{X}_q \mathbf{a}_p = 0$ which has a nonzero solution



- And now that we have the \mathbf{a}_p , let us get the \mathbf{b}_q
- Taking the "upper part" of the system, we have

$$\begin{bmatrix} x(0) & 0 & \cdots & 0 \\ x(1) & x(0) & \cdots & 0 \\ \vdots & \vdots & \cdots & \vdots \\ x(q) & x(q-1) & \cdots & x(q-p) \end{bmatrix} \begin{bmatrix} 1 \\ a_p(1) \\ \vdots \\ a_p(p) \end{bmatrix} = \begin{bmatrix} b_q(0) \\ b_q(1) \\ \vdots \\ b_q(q) \end{bmatrix}$$

which in matrix form is this time $\mathbf{X}_0\mathbf{a}_p=\mathbf{b}_q$

lacksquare And you just have to do a matrix multiplication to get the $lackbox{b}_q$



Now an example:

■ Take a signal of which you have values

$$\mathbf{x} = [1, 1.5, 0.75, 0.375, 0.1875, 0.0938]^T$$

- Say we'd like two filters:
 - 1. A second order all pole filter
 - 2. A model with just one pole and one zero



Let's start with the first one: Second order all pole

We have to solve:

$$\begin{bmatrix} x(0) & 0 & 0 \\ x(1) & x(0) & 0 \\ x(2) & x(1) & x(0) \end{bmatrix} \begin{bmatrix} 1 \\ a(1) \\ a(2) \end{bmatrix} = \begin{bmatrix} b(0) \\ 0 \\ 0 \end{bmatrix}$$

■ Take the last two equations and substitute:

$$\begin{bmatrix} 1 & 0 \\ 1.5 & 1 \end{bmatrix} \begin{bmatrix} a(1) \\ a(2) \end{bmatrix} = - \begin{bmatrix} 1.5 \\ 0.75 \end{bmatrix}$$

which gives a(1) = -1.5 and a(2) = 1.5



And I think finding the b(0) should not be difficult

■ So we get

$$H(z) = \frac{1}{1 - 1.5z^{-1} + 1.5z^{-2}}$$

And the approximation given by the model is

$$\hat{\mathbf{x}} = [1, 1.5, 0.75, -1.125, -2.8125, -2.5312]^T$$

And we wanted

$$\mathbf{x} = [1, 1.5, 0.75, 0.375, 0.1875, 0.0938]^T$$

but only enforced on the first three values, due to filter design



Second one: One pole, one zero

We have

$$H(z) = \frac{b(0) + b(1)z^{-1}}{1 + a(1)z^{-1}}$$

So the Padé equations are

$$\begin{bmatrix} x(0) & 0 \\ x(1) & x(0) \\ x(2) & x(1) \end{bmatrix} \begin{bmatrix} 1 \\ a(1) \end{bmatrix} = \begin{bmatrix} b(0) \\ b(1) \\ 0 \end{bmatrix}$$

So
$$a(1) = -\frac{x(2)}{x(1)} = -0.5$$



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And for the **b** coefficients, we solve

$$\left[\begin{array}{c}b(0)\\b(1)\end{array}\right]=\left[\begin{array}{cc}1&0\\1.5&1\end{array}\right]\left[\begin{array}{c}1\\-0.5\end{array}\right]=\begin{array}{c}1\\1\end{array}$$

And the model is then

$$H(z) = \frac{1 + z^{-1}}{1 - 0.5z^{-1}}$$

- Which matches the required values (I'll let you check that by writing the inverse transform of the model)
- It was meant to happen here, for x was designed as the unit sample response of a LSI filter



Padé Approximation: Conclusion

In conclusion on Padé's approximation:

- A model created using the Padé's approximation will always produce an exact fit over the data in the interval [0, p+q] if \mathbf{X}_q is not singular
- Since we do not consider the data outside of that interval, likely we don't have an accurate model outside that interval
- Padé's approximation forces the model to fit over a limited range of values, with no other constraints: model might be unstable (no guarantees)

- Relax the constraint to match exactly the values on the interval $[\![0,p+q]\!]$, in order to better behave over the whole range
- Will also have a set of linear equations to solve to have filter coefficients
- We still want to minimize the error

$$e'(n) = x(n) - h(n)$$

where h(n) is the unit sample response of the filter that is supposed to match x(n)

■ Which can also be written $E'(z) = X(z) - \frac{B_q(z)}{A_0(z)}$



Now if we use the idea from the Padé approximation, and multiply this expression by $A_p(z)$, we have another error, linear with the filter coefficients

$$E(z) = A_p(z)E'(z) = A_p(z)X(z) - B_q(z)$$

Going back to the time domain, we have

$$e(n) = a_p(n) * x(n) - b_q(n) = \hat{b}_q(n) - b_q(n)$$



And since $b_q(n) = 0, \forall n > q$, we can write the error for all values of n as

$$e(n) = \begin{cases} x(n) + \sum_{l=1}^{p} a_{p}(l)x(n-l) - b_{q}(n) &, n = 0, \dots, q \\ x(n) + \sum_{l=1}^{p} a_{p}(l)x(n-l) &, n > q \end{cases}$$



- Now, with the Padé approximation idea, we would set e(n) = 0 for $n = 0, \dots, p + q$
- With Prony's, we first find the a coefficients that minimize the least squares error

$$\varepsilon_{p,q} = \sum_{n=q+1}^{\infty} |e(n)|^2 = \sum_{n=q+1}^{\infty} \left| x(n) + \sum_{l=1}^{p} a_p(l) x(n-l) \right|^2$$



Using partial derivatives w.r.t. $a_p^*(k)$, we get the minimum of $\varepsilon_{p,q}$ as the solution of

$$\sum_{n=q+1}^{\infty} e(n)x^{*}(n-k) = 0, k = 1, \dots, p$$



Substituting the previous expression of e(n) for all the n in this equation, we have finally

$$\sum_{l=1}^{p} a_{p}(l) r_{x}(k, l) = -r_{x}(k, 0), k = 1, \dots, p$$

with
$$r_x(k,l) = \sum_{n=q+1}^{\infty} x(n-l)x^*(n-k)$$
 which is similar to an autocorrelation sequence



Now, the previous equation is a set of equations, that can be written in matrix form as

$$\begin{bmatrix} r_{x}(1,1) & r_{x}(1,2) & \cdots & r_{x}(1,p) \\ r_{x}(2,1) & r_{x}(2,2) & \cdots & r_{x}(2,p) \\ \vdots & \vdots & \ddots & \vdots \\ r_{x}(p,1) & r_{x}(p,2) & \cdots & r_{x}(p,p) \end{bmatrix} \begin{bmatrix} a_{p}(1) \\ a_{p}(2) \\ \vdots \\ a_{p}(p) \end{bmatrix} = - \begin{bmatrix} r_{x}(1,0) \\ r_{x}(2,0) \\ \vdots \\ r_{x}(p,0) \end{bmatrix}$$

or
$$\mathbf{R}_{x}\mathbf{a}_{p}=-\mathbf{r}_{x}$$



Now, with an infinite column length matrix \mathbf{X}_q defined as

$$\mathbf{X}_q = \left[egin{array}{cccc} x(q) & x(q-1) & \cdots & x(q-p+1) \ x(q+1) & x(q) & \cdots & x(q-p+2) \ dots & dots & \ddots & dots \end{array}
ight]$$

- We can express $\mathbf{R}_x = \mathbf{X}_q^H \mathbf{X}_q$ and $\mathbf{r}_x = \mathbf{X}_q^H \mathbf{x}_{q+1}$ with $\mathbf{x}_{q+1} = [x(q+1), x(q+2), \ldots]^T$ (also infinite length)
- And then the Prony equations become

$$\left(\mathbf{X}_{q}^{H}\mathbf{X}_{q}\right)\mathbf{a}_{p}=-\mathbf{X}_{q}^{H}\mathbf{x}_{q+1}$$



 \blacksquare Meaning that if \mathbf{R}_{\times} is nonsingular, we have the coefficients

$$\mathbf{a}_p = -\left(\mathbf{X}_q^H \mathbf{X}_q\right)^{-1} \mathbf{X}_q^H \mathbf{x}_{q+1}$$

Injecting these coefficients in the error, we have the minimum $\epsilon_{p,q}$ of the original error $\varepsilon_{p,q}$ as

$$\epsilon_{p,q} = r_{x}(0,0) + \sum_{k=1}^{p} a_{p}(k) r_{x}(0,k)$$



Now, to find the **b** coefficients, we end up with the same as for Padé's approximation: setting the error to zero and solving

$$e(n) = a_p(n) * x(n) - b_q(n) = 0$$

So, the **b** coefficients are obtained by the set of equations

$$b_q(n) = x(n) + \sum_{k=1}^{p} a_p(k)x(n-k)$$

■ Which is straightforward to obtain



An example:

Let's have a signal x(n) consisting of a pulse of length N, as

$$x(n) = \begin{cases} 1 & , n = 0, \dots, N-1 \\ 0 & \text{otherwise} \end{cases}$$

And we'll model it as the unit sample response of a LSI filter with one pole and one zero:

$$H(z) = \frac{b(0) + b(1)z^{-1}}{1 + a(1)z^{-1}}$$



■ So, with p = 1, the Prony equations in a_p are

$$a(1)r_x(1,1) = -r_x(1,0)$$

■ And we can calculate $r_x(1,1)$ and $r_x(1,0)$ easily:

$$r_X(1,1)$$
 = $\sum_{n=2}^{\infty} x^2(n-1) = N-1$
 $r_X(1,0)$ = $\sum_{n=2}^{\infty} x(n)x(n-1) = N-2$

■ So the denominator of H(z), A(z), becomes

$$A(z) = 1 - \frac{N-2}{N-1}z^{-1}$$

 \blacksquare And for the b_q coefficients, we have the equations

$$b(0) = x(0) = 1$$

 $b(1) = x(1) + a(1)x(0) = 1 - \frac{N-2}{N-1} = \frac{1}{N-1}$



So finally

$$H(z) = \frac{1 + \frac{1}{N-1}z^{-1}}{1 - \frac{N-2}{N-1}z^{-1}}$$

And the minimum squared error can be computed as

$$\epsilon_{1,1} = r_{x}(0,0) + a(1)r_{x}(0,1)$$

■ Since $r_x(0,0) = N - 2$, we have $\epsilon_{1,1} = \frac{N-2}{N-1}$

Further Reading on this subject

- Shank's method: Extend the LS constraint of Padé over the entire series of records
- All pole using Shank's method
- Please have a read on the course webpage "Additional Reading" section (not mandatory): Link to PDF excerpt on Shank's Method



Finite Data records

- Prony's method assumes x(n) known for all n
- What if we just have a window of values on x(n)?
 - Autocorrelation Method: "window" our process x(n) by a rectangular window which sets x(n) to zero outside of the records we have
 - Covariance Method: minimize an error that does not depend on the values of x(n) outside of the records we have



Suppose we have a signal x(n) (eventually complex) known only on the interval [0, N] and that we want to approximate x(n) by an all-pole model

$$H(z) = \frac{b(0)}{1 + \sum_{k=1}^{p} a_{p}(k)z^{-k}}$$



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With Prony's method, we minimize

$$\varepsilon_p = \sum_{n=0}^{\infty} |e(n)|^2$$

with

$$e(n) = x(n) + \sum_{k=1}^{p} a_{p}(k)x(n-k)$$

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- But if we don't know x(n) outside of [0, N], we can't evaluate e(n) for n < p or n > N
- So we create new signal, $\tilde{x}(n)$ based on x(n), by applying a rectangular window:

$$\tilde{x}(n) = x(n)w(n)$$

with

$$w(n) = \begin{cases} 1 & , n = 0, \dots, N \\ 0 & , \text{otherwise} \end{cases}$$



- Then we can use Prony's method to find an all-pole model for $\tilde{x}(n)$
- The Prony equations are identical, except that the autocorrelation is that of $\tilde{x}(n)$ instead of x(n), i.e. replace the usual $r_x(k)$ by $r_{\tilde{x}}(k)$:

$$r_{\tilde{x}}(k) = \sum_{n=0}^{\infty} \tilde{x}(n)\tilde{x}^*(n-k) = \sum_{n=k}^{N} x(n)x^*(n-k), k = 0, \dots, p$$



- Covariance Method
- What if you don't want to force the signal to be zero outside for solving?
- Covariance method usually gives more accurate models than autocovariance one, but lose the Toeplitz structure of the equations (and hence the fast ways to solve)



- Again with an all-pole model
- Different error ε_p^C to minimize, that only uses the known values of e(n) with no assumptions on the other ones (if x(n) is known for [0, N], then e(n) can be known for [p, N])

$$\varepsilon_p^C = \sum_{n=p}^N |e(n)|^2$$

■ Finding the all-pole coefficients minimizing this error is known as the Covariance Method



It leads to the same set of normal equations as for Prony's method, called here the covariance normal equations

$$\begin{bmatrix} r_{x}(1,1) & r_{x}(1,2) & \cdots & r_{x}(1,p) \\ r_{x}(2,1) & r_{x}(2,2) & \cdots & r_{x}(2,p) \\ \vdots & \vdots & \vdots & \vdots \\ r_{x}(p,1) & r_{x}(p,2) & \cdots & r_{x}(p,p) \end{bmatrix} \begin{bmatrix} a_{p}(1) \\ a_{p}(2) \\ \vdots \\ a_{p}(3) \end{bmatrix} = - \begin{bmatrix} r_{x}(1,0) \\ r_{x}(2,0) \\ \vdots \\ r_{x}(p,0) \end{bmatrix}$$

with the autocorrelation sequence $r_x(k, l)$ expressed as

$$r_{x}(k,l) = \sum_{n=p}^{N} x(n-l)x^{*}(n-k)$$



■ And the minimum covariance modeling error ϵ_p^C is

$$\epsilon_p^C = r_x(0,0) + \sum_{k=1}^p a_p(k) r_x(0,k)$$

 Covariance normal equations are not Toeplitz, so no easy/fast solving in the general case (fast solving algorithms exist, but complicated)



Summary

- Least Squares Method
- Padé Approximation
- Prony's Method
- Autocorrelation Method
- Covariance Method



Next time

- More examples for AR, MA, ARMA models
- Power Spectrum estimation
- A look at the Levinson-Durbin recursion to solve Toeplitz equations
- A look at lattice filters

