



DSP - Fisika UI



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A finite-duration impulse response filter has a system function of the form

$$H(z) = b_o + b_1 z^{-1} + \dots + b_{M-1} z^{1-M} = \sum_{n=0}^{M-1} b_n z^{-n}$$

• The impulse response h(n) is

$$h(n) = \begin{cases} b_n, & 0 \le n \le M - 1 \\ 0, & \text{else} \end{cases}$$

The difference equation representation is

$$y(n) = b_o x(n) + b_1 x(n-1) + \dots + b_{M-1} x(n-M+1)$$

• The order of the filter is M-1, and the *length* of the filter (which is equal to the number of coefficients) is M

- The FIR filter structures are
  - a linear convolution of finite support
  - always stable
  - relatively simple compared to IIR structures
- FIR filters can be designed to have a linear-phase response, which is desirable in some applications.

1. Direct form: 
$$y(n) = b_o x(n) + b_1 x(n-1) + ... + b_{M-1} x(n-M+1)$$

**2. Cascade form:** 
$$H(z) = b_o + b_1 z^{-1} + ... + b_{M-1} z^{1-M} = \sum_{n=0}^{M-1} b_n z^{-n}$$

this form the system function H(z) is factored into 2nd-order factors, which are then implemented in a cascade connection.

#### 3. Linear-phase form:

its impulse response exhibits certain symmetry conditions to reduce multiplications by about half.

#### 4. Frequency-sampling form:

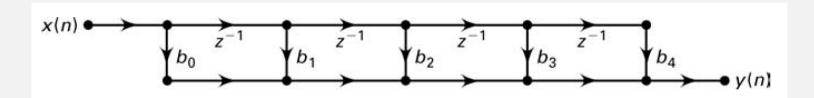
This structure is based on the DFT of the impulse response h(n) and leads to a parallel structure

suitable for a design technique based on the sampling of frequency response  $H(e^{j\omega})$ .

## **DIRECT FORM**

- The difference equation is implemented as a tapped delay line since there are no feedback paths.
- Let *M* = 5 (i.e., a 4th-order FIR filter)

$$y(n) = b_o x(n) + b_1 x(n-1) + b_2 x(n-2) + b_3 x(n-3) + b_4 x(n-4)$$



## **DIRECT FORM**

- The direct form FIR structure is described by the row vector b containing the {bn} coefficients.
- The structure is implemented by the filter function, in which the vector a is set to the scalar value 1

## **CASCADE FORM**

• The system function H(z) is converted into products of 2nd-order sections with real coefficients

$$H(z) = b_o + b_1 z^{-1} + \dots + b_{M-1} z^{-M+1}$$

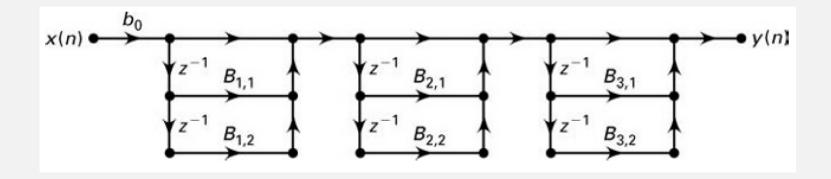
$$= b_o \left( 1 + \frac{b_1}{b_o} z^{-1} + \dots + \frac{b_{M-1}}{b_o} z^{-M+1} \right)$$

$$= b_o \prod_{k=1}^K \left( 1 + B_{k,1} z^{-1} + B_{k,2} z^{-2} \right)$$

• K is equal to M/2, and  $B_{k,1}$  and  $B_{k,2}$  are real numbers representing the coefficients of 2nd-order sections.

# **CASCADE FORM**

• For M = 7 the cascade form is shown as



### **CASCADE FORM**

- Use our dir2cas function by setting the denominator vector a equal to 1
- Use cas2dir to obtain the direct form from the cascade form.

- For frequency-selective filters (e.g., lowpass filters) it is generally desirable to have a phase response that is a linear function of frequency
- where  $\beta = 0$  or  $\pm \pi/2$  and  $\alpha$  is a constant.

$$\angle H(e^{j\omega}) = \beta - \alpha\omega, \quad -\pi < \omega \le \pi$$

- The linear-phase condition imposes the following symmetry conditions on the impulse response h(n)
  - a symmetric impulse response

$$h(n) = h(M-1-n); \quad \beta = 0, \alpha = \frac{M-1}{2}, \quad 0 \le n \le M-1$$

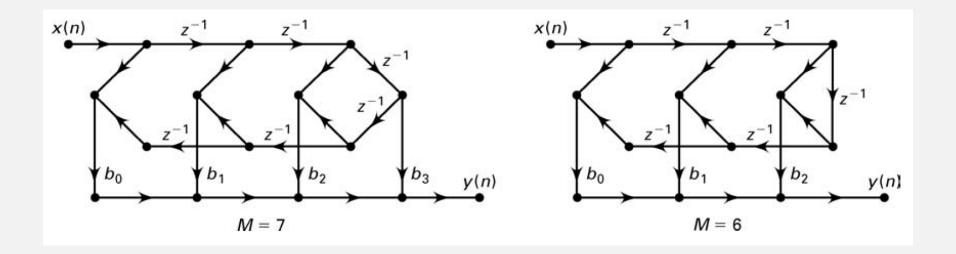
an antisymmetric impulse response

$$h(n) = -h(M-1-n); \quad \beta = \pm \pi/2, \alpha = \frac{M-1}{2}, \quad 0 \le n \le M-1$$

 Consider the difference equation with a symmetric impulse response in a symmetric impulse response

$$y(n) = b_o x(n) + b_1 x(n-1) + \dots + b_1 x(n-M+2) + b_o x(n-M+1)$$
  
=  $b_o [x(n) + x(n-M+1)] + b_1 [x(n-1) + x(n-M+2)] + \dots$ 

• The block diagram implementation of the previous difference equation for both odd and even *M*.



- The linear-phase structure is essentially a direct form drawn differently to save on multiplications.
- Hence in a MATLAB representation of the linear-phase structure is equivalent to the direct form.

An FIR filter is given by the system function

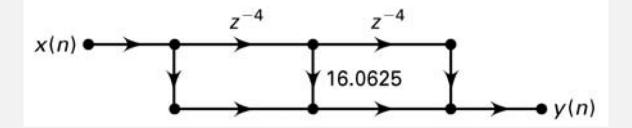
$$H(z) = 1 + 16\frac{1}{16}z^{-4} + z^{-8}$$

• Determine and draw the direct, linear-phase, and cascade form structures

$$H(z) = 1 + 16\frac{1}{16}z^{-4} + z^{-8}$$

• **Direct form:** The difference equation is given by

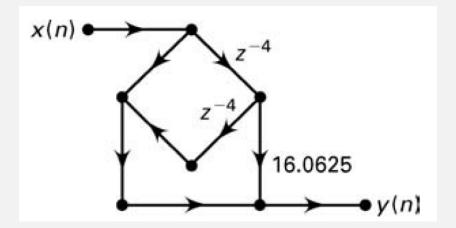
$$y(n) = x(n) + 16.0625x(n - 4) + x(n - 8)$$



$$H(z) = 1 + 16\frac{1}{16}z^{-4} + z^{-8}$$

• Linear-phase form: The difference equation can be written in the form

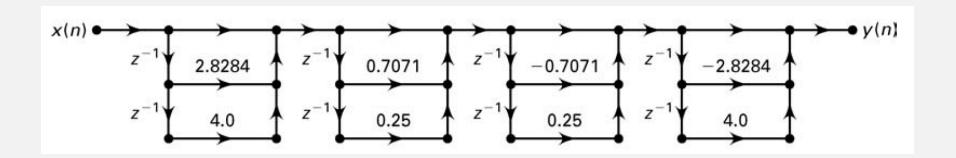
$$y(n) = [x(n) + x(n - 8)] + 16.0625x(n - 4)$$

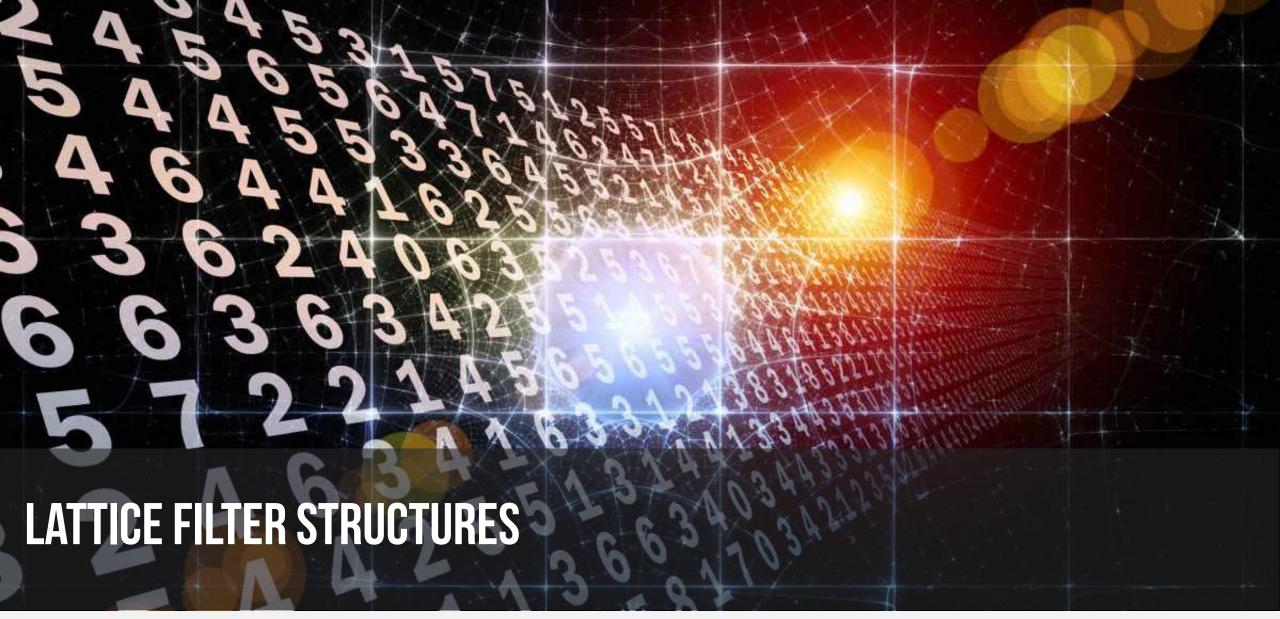


$$H(z) = 1 + 16\frac{1}{16}z^{-4} + z^{-8}$$

#### Cascade form:

>> b=[1,0,0,0,16+1/16,0,0,0,1]; [b0,B,A] = dir2cas(b,1)





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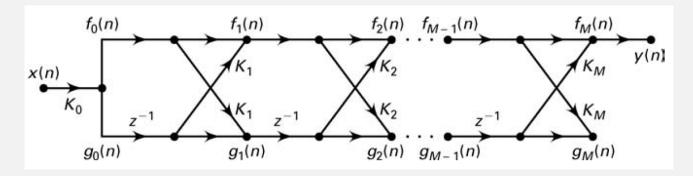


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# LATTICE FILTER STRUCTURES

- Extensively used in digital speech processing and in the implementation of adaptive filters
- The all-zero lattice is the FIR filter representation of the lattice filter, while the lattice ladder is the IIR filter representation.

• An FIR filter of length M (or order M-1) has a lattice structure with M-1 stages



 Each stage of the filter has an input and output that are related by the orderrecursive equations

$$f_m(n) = f_{m-1}(n) + K_m g_{m-1}(n-1), \quad m = 1, 2, ..., M-1$$

$$g_m(n) = K_m f_{m-1}(n) + g_{m-1}(n-1), \quad m = 1, 2, ..., M-1$$

• The parameters  $K_m$ , m = 1, 2, ..., M - 1, called the *reflection coefficients*, are the lattice filter coefficients

## **CONVERSION**

The direct form of the FIR filter

$$H(z) = \sum_{m=0}^{M-1} b_m z^{-m} = b_o \left( 1 + \sum_{m=1}^{M-1} \frac{b_m}{b_o} z^{-m} \right)$$

• Computing the lattice filter coefficients  $\{K_m\}$ 

$$K_{o} = b_{o}$$

$$K_{M-1} = \alpha_{M-1} (M-1)$$

$$J_{m}(z) = z^{-m} A_{m}(z^{-1}), \quad m = M-1,...,1$$

$$A_{m-1}(z) = \frac{A_{m}(z) - K_{m} J_{m}(z)}{1 - K_{m}^{2}}, \quad m = M-1,...,1$$

$$K_{m} = \alpha_{m}(m), \quad m = M-2,...,1$$

```
function [K] = dir2latc(b)
% FIR Direct form to All-Zero Lattice form Conversion
% [K] = dir2latc(b)
% K = Lattice filter coefficients (reflection coefficients)
% b = FIR direct form coefficients (impulse response)
%
M = length(b); K = zeros(1,M); b1 = b(1);
if b1 == 0
      error('b(1) is equal to zero')
end
K(1) = b1; A = b/b1;
for m=M:-1:2
      K(m) = A(m); J = fliplr(A);
      A = (A-K(m)*J)/(1-K(m)*K(m)); A = A(1:m-1);
end
```

```
function [y] = latcfilt(K,x)
% LATTICE form realization of FIR filters
% y = latcfilt(K,x)
% y = output sequence
% K = LATTICE filter (reflection) coefficient array
% x = input sequence
%
Nx = length(x)-1; x = K(1)*x;
M = length(K)-1; K = K(2:M+1); fg = [x; [0 x(1:Nx)]];
for m = 1:M
      fg = [1,K(m);K(m),1]*fg;
      fg(2,:) = [0 fg(2,1:Nx)];
end
y = fg(1,:);
```

```
function [b] = latc2dir(K)
% All-Zero Lattice form to FIR Direct form Conversion
% [b] = latc2dir(K)
% b = FIR direct form coefficients (impulse response)
% K = Lattice filter coefficients (reflection coefficients)
%
M = length(K); J = 1; A = 1;
for m=2:1:M
A = [A,0] + conv([0,K(m)],J); J = fliplr(A);
end
b=A*K(1);
```

An FIR filter is given by the difference equation

$$y(n) = 2x(n) + \frac{13}{12}x(n-1) + \frac{5}{4}x(n-2) + \frac{2}{3}x(n-3)$$

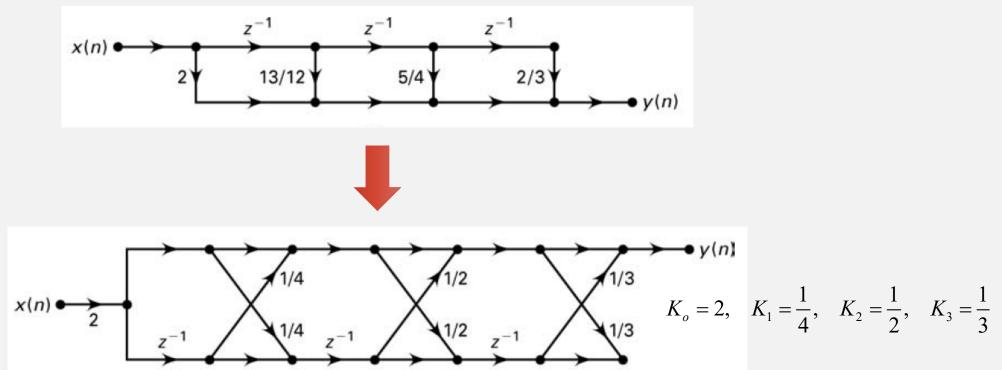
Determine its lattice form

$$y(n) = 2x(n) + \frac{13}{12}x(n-1) + \frac{5}{4}x(n-2) + \frac{2}{3}x(n-3)$$

>> b=[2, 13/12, 5/4, 2/3]; K=dir2latc(b)

K =

2.0000 0.2500 0.5000 0.3333



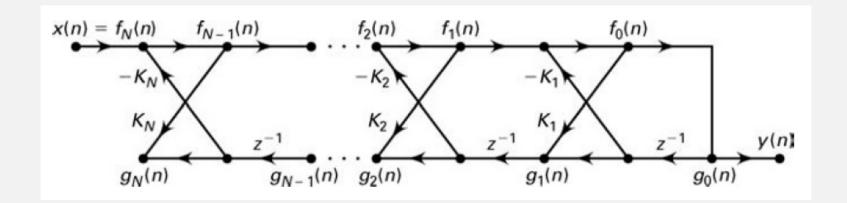
$$y(n) = 2x(n) + \frac{13}{12}x(n-1) + \frac{5}{4}x(n-2) + \frac{2}{3}x(n-3)$$

## **ALL-POLE LATTICE FILTERS**

- A lattice structure for an IIR filter is restricted to an all-pole system function
- It can be developed from an FIR lattice structure

$$H(z) = \frac{1}{1 + \sum_{m=1}^{N} a_N(m) z^{-m}}$$

An inverse system to the FIR lattice (All-pole lattice filter)



## **CONVERSION**

The direct form of the FIR filter

$$H(z) = \sum_{m=0}^{M-1} b_m z^{-m} = b_o \left( 1 + \sum_{m=1}^{M-1} \frac{b_m}{b_o} z^{-m} \right)$$

• Computing the lattice filter coefficients  $\{K_m\}$ 

$$K_{o} = b_{o}$$

$$K_{M-1} = \alpha_{M-1} (M-1)$$

$$J_{m}(z) = z^{-m} A_{m}(z^{-1}), \quad m = M-1,...,1$$

$$A_{m-1}(z) = \frac{A_{m}(z) - K_{m} J_{m}(z)}{1 - K_{m}^{2}}, \quad m = M-1,...,1$$

$$K_{m} = \alpha_{m}(m), \quad m = M-2,...,1$$

### **ALL-POLE LATTICE FILTER**

- Use the dir2latc function to compute lattice coefficients
- Use the latc2dir function to convert the lattice {Km} coefficients into the direct form {aN(m)}

Consider an all-pole IIR filter given by

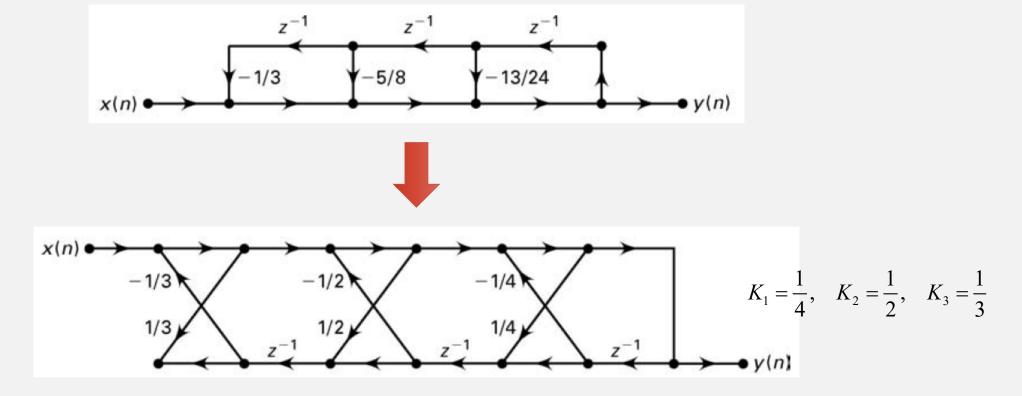
$$H(z) = \frac{1}{1 + \frac{13}{24}z^{-1} + \frac{5}{8}z^{-2} + \frac{1}{3}z^{-3}}$$

Determine its lattice structure

$$H(z) = \frac{1}{1 + \frac{13}{24}z^{-1} + \frac{5}{8}z^{-2} + \frac{1}{3}z^{-3}}$$

K =

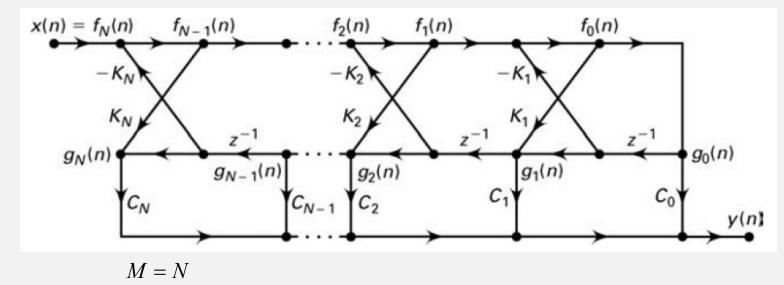
1.0000 0.2500 0.5000 0.3333



 A general IIR filter containing both poles and zeros can be realized as a lattice-type structure by using an all-pole lattice as the basic building block.

$$H(z) = \frac{\sum_{k=0}^{M} b_{M}(k) z^{-k}}{1 + \sum_{k=1}^{N} a_{N}(k) z^{-k}} = \frac{B_{M}(z)}{A_{N}(z)}$$

The lattice-ladder structure



The pole-zero IIR filter output is given by

$$y(n) = \sum_{m=0}^{M} C_m g_m(n)$$

• where  $\{C_m\}$  are called the *ladder coefficients* that determine the zeros of the system function H(z).

```
function [K,C] = dir2ladr(b,a)
% IIR Direct form to pole-zero Lattice/Ladder form Conversion
% [K,C] = dir2ladr(b,a)
% K = Lattice coefficients (reflection coefficients), [K1,...,KN]
% C = Ladder Coefficients, [C0,...,CN]
% b = Numerator polynomial coefficients (deg <= Num deg)
% a = Denominator polynomial coefficients
a1 = a(1); a = a/a1; b = b/a1; M = length(b); N = length(a);
if M > N
       error(' *** length of b must be <= length of a ***')
end
b = [b, zeros(1,N-M)]; K = zeros(1,N-1); A = zeros(N-1,N-1); C = b;
for m = N-1:-1:1
       A(m,1:m) = -a(2:m+1)*C(m+1);
       K(m) = a(m+1); J = fliplr(a);
        a = (a-K(m)*J)/(1-K(m)*K(m)); a = a(1:m);
       C(m) = b(m) + sum(diag(A(m:N-1,1:N-m)));
end
```

```
function [b,a] = ladr2dir(K,C)
% Lattice/Ladder form to IIR Direct form Conversion
% [b,a] = ladr2dir(K,C)
% b = numerator polynomial coefficients
% a = denominator polymonial coefficients
% K = Lattice coefficients (reflection coefficients)
% C = Ladder coefficients
N = length(K); M = length(C); C = [C, zeros(1,N-M+1)];
J = 1; a = 1; A = zeros(N,N);
for m=1:1:N
      a = [a,0]+conv([0,K(m)],J); A(m,1:m) = -a(2:m+1); J = fliplr(a);
end
b(N+1) = C(N+1);
for m = N:-1:1
      A(m,1:m) = A(m,1:m)*C(m+1); b(m) = C(m) - sum(diag(A(m:N,1:N-m+1)));
end
```

Convert the following pole-zero IIR filter into a lattice-ladder structure.

$$H(z) = \frac{1 + 2z^{-1} + 2z^{-2} + z^{-3}}{1 + \frac{13}{24}z^{-1} + \frac{5}{8}z^{-2} + \frac{1}{3}z^{-3}}$$

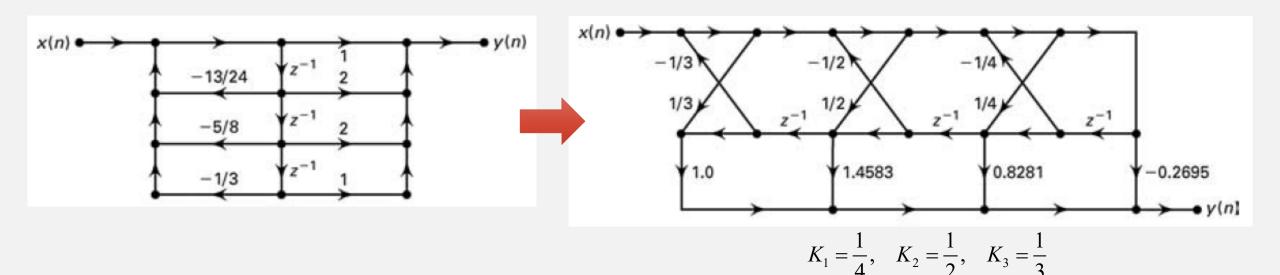
$$H(z) = \frac{1 + 2z^{-1} + 2z^{-2} + z^{-3}}{1 + \frac{13}{24}z^{-1} + \frac{5}{8}z^{-2} + \frac{1}{3}z^{-3}}$$

 $C_0 = -0.2695$ ,  $C_1 = 0.8281$ ,  $C_2 = 1.4583$ ,  $C_3 = 1$ 

0.2500 0.5000 0.3333

C =

-0.2695 0.8281 1.4583 1.0000



$$H(z) = \frac{1 + 2z^{-1} + 2z^{-2} + z^{-3}}{1 + \frac{13}{24}z^{-1} + \frac{5}{8}z^{-2} + \frac{1}{3}z^{-3}}$$

```
>> [x,n]=impseq(0,0,7) format long hdirect = filter(b,a,x)
hdirect =
Columns 1 through 4
1.00000000000000 1.4583333333333 0.5850694444444 -0.56170428240741
Columns 5 through 8
-0.54752302758488 0.45261700163162 0.28426911049255 -0.25435705167494
>> hladder = ladrfilt(K,C,x)
hladder =
Columns 1 through 4
1.00000000000000 1.4583333333333 0.5850694444444 -0.56170428240741
Columns 5 through 8
-0.54752302758488 0.45261700163162 0.28426911049255 -0.25435705167494
```



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