# Chapter 8 Digital Filter Structures

**Zhiliang Liu** 

Zhiliang Liu@uestc.edu.cn

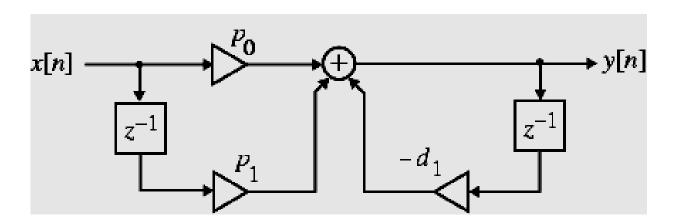
4/25/2019

#### Advantages of block diagram representation

- (1) Easy to write down the <u>computational</u> algorithm by inspection
- (2) Easy to analyze the block diagram to determine the explicit <u>relation between the</u> <u>output and input</u>

- (3) Easy to manipulate a block diagram to derive other "equivalent" block diagrams yielding different computational algorithms
- (4) Easy to determine the <u>hardware</u>
   <u>requirements</u>
- (5) Easier to develop block diagram representations from the <u>transfer function</u> directly

- For the implementation of an LTI digital filter, the input-output relationship must be described by a valid <u>computational algorithm</u>
- To illustrate what we mean by a computational algorithm, consider the causal first-order LTI digital filter shown below



• The filter is described by the difference equation

$$y[n] = -d_1y[n-1] + p_0x[n] + p_1x[n-1]$$

• Using the above equation we can compute y[n] for  $n \ge 0$  knowing the initial condition y[n-1] and the input x[n] for  $n \ge -1$ 

$$y[0] = -d_1y[-1] + p_0x[0] + p_1x[-1]$$

$$y[1] = -d_1y[0] + p_0x[1] + p_1x[0]$$

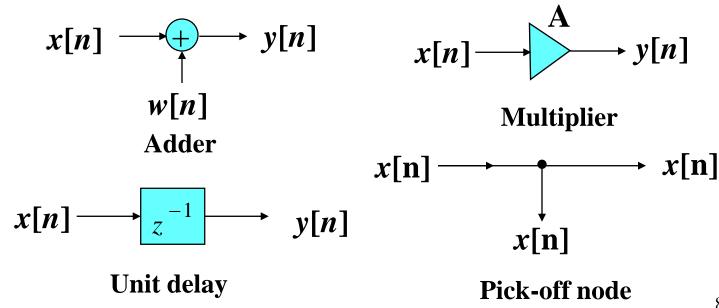
$$y[2] = -d_1y[1] + p_0x[2] + p_1x[1]$$
....

• We can continue this calculation for any value of the time index *n* we desire

- Each step of the calculation requires a knowledge of the previously calculated value of the output sample (delayed value of the output), the present value of the input sample, and the previous value of the input sample (delayed value of the input)
- As a result, the first-order difference equation can be interpreted as a valid computational algorithm

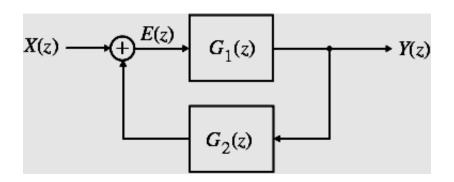
## § 8.1.1 Basic Building Blocks

• The computational algorithm of an LTI digital filter can be conveniently represented in block diagram form using the basic building blocks shown below



- Carried out by writing down the expressions for the output signals of each adder as a sum of its input signals, and developing a set of equations relating the filter input and output signals in terms of all internal signals
- Eliminating the unwanted internal variables then results in the expression for the output signal as a function of the input signal and the filter parameters that are the multiplier coefficients

**Example** - Consider the single-loop feedback structure shown below



The output E(z) of the adder is

$$E(z) = X(z) + G_2(z)Y(z)$$

But from the figure

$$Y(z) = G_1(z)E(z)$$

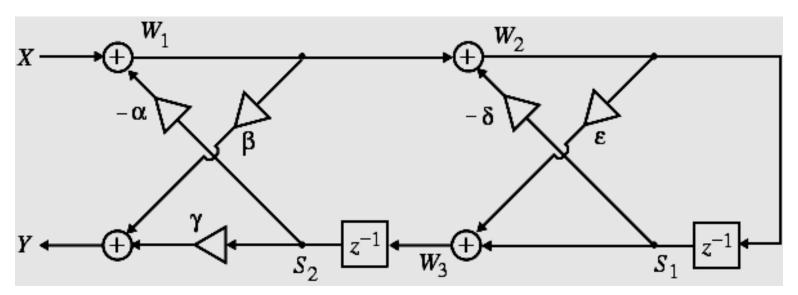
• Eliminating E(z) from the previous two equations we arrive at

$$[1-G_1(z)G_2(z)]Y(z) = G_1(z)X(z)$$

which leads to

$$H(z) = \frac{Y(z)}{X(z)} = \frac{G_1(z)}{1 - G_1(z)G_2(z)}$$

• <u>Example</u> - Analyze the cascaded lattice structure shown below where the z-dependence of signal variables are not shown for brevity



 The output signals of the four adders are given by

$$W_1 = X - \alpha S_2 \qquad W_2 = W_1 - \delta S_1$$

$$W_3 = S_1 - \varepsilon W_2 \qquad Y = \beta W_1 - \gamma S_2$$

From the figure we observe

$$S_2 = z^{-1} W_3$$
  $S_1 = z^{-1} W_2$ 

• Substituting the last two relations in the first four equations we get

$$W_1 = X - \alpha z^{-1} W_3$$
  $W_2 = W_1 - \delta z^{-1} W_2$   
 $W_3 = z^{-1} W_2 + \varepsilon W_2$   $Y = \beta W_1 + \gamma z^{-1} W_3$ 

• From the second equation we get  $W_2 = W_1/(1+\delta z^{-1})$  and from the third equation we get  $W_3 = (\varepsilon + z^{-1})W_2$ 

Combining the last two equations we get

$$W_3 = \frac{\varepsilon + z^{-1}}{1 + \delta z^{-1}} W_1$$

Substituting the above equation in

$$W_1 = X - \alpha z^{-1}W_3, \qquad Y = \beta W_1 + \gamma z^{-1}W_3$$

we finally arrive at

$$H(z) = \frac{Y}{X} = \frac{\beta + (\beta \delta + \gamma \varepsilon) z^{-1} + \gamma z^{-2}}{1 + (\delta + \alpha \varepsilon) z^{-1} + \alpha z^{-2}}$$

#### *Note*:

**<u>Delay-free loop</u>**: feedback loops without any delay elements (physically impossible).

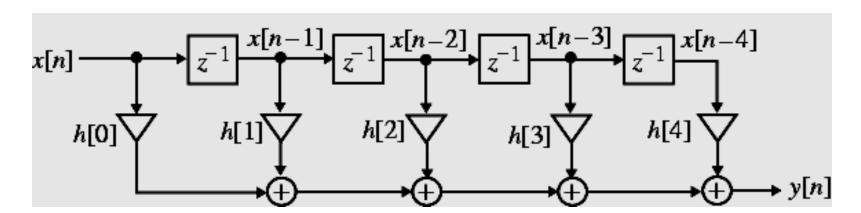
<u>canonic structure</u>: The number of delays is equal to the order of the difference equation.

- Two digital filter structures are defined to be equivalent if they have the same transfer function
- We describe next a number of methods for the generation of equivalent structures
- However, a fairly simple way to generate an equivalent structure from a given realization is via the *transpose operation*

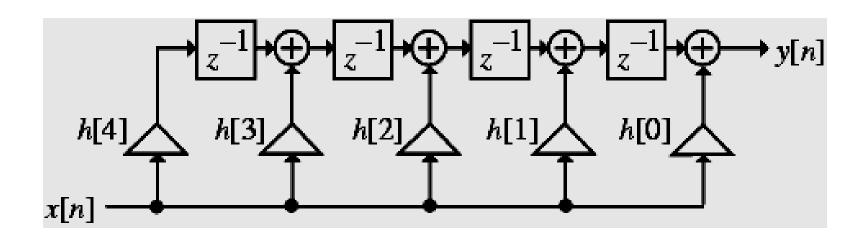
#### Transpose Operation

- (1) Reverse all paths
- (2) Replace pick-off nodes by adders, and vice versa
- (3) Interchange the input and output nodes

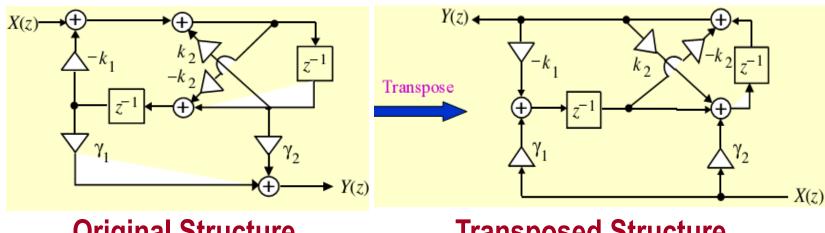
#### **Example:** Original structure



#### **Transposed structure**



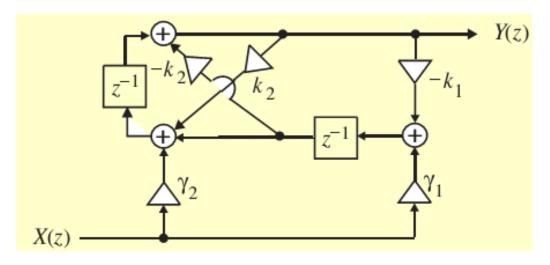
• **Example:** The transpose operation is illustrated below



**Original Structure** 

**Transposed Structure** 

Redrawn transposed structure is shown below



All other methods for developing equivalent structures are based on <u>a specific algorithm for each structure</u>

- There are literally an infinite number of equivalent structures realizing the same transfer function
- It is thus impossible to develop all equivalent realizations
- In this course we restrict our attention to a discussion of some commonly used structures

- Under infinite precision arithmetic any given realization of a digital filter behaves identically to any other equivalent structure
- However, in practice, due to the <u>finite word-length limitations</u>, a specific realization behaves totally differently from its other equivalent realizations

- Hence, it is important to choose a structure that has the <u>least quantization effects</u> when implemented using finite precision arithmetic
- One way to arrive at such a structure is to determine a large number of equivalent structures, <u>analyze</u> the finite wordlength effects in each case, <u>and select</u> the one showing the least effects

- In certain cases, it is possible to develop a structure that by construction has the least quantization effects
- We defer the review of these structures after a discussion of the analysis of quantization effects
- Here, we review some simple realizations that in many applications are quite adequate

## 8.3 Basic FIR Digital Filter Structures

• A causal FIR filter of order N is characterized by a transfer function H(z) given by

$$H(z) = \sum_{n=0}^{N} h[n]z^{-n}$$

which is a polynomial in  $z^{-1}$ 

• In the time-domain the input-output relation of the above FIR filter is given by

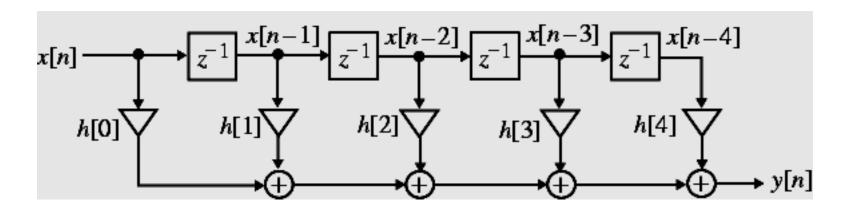
$$y[n] = \sum_{k=0}^{N} h[k]x[n-k]$$

## 8.3.1 Direct Form FIR Digital Filter Structures

- An FIR filter of order N is characterized by N+1 coefficients and, in general, require N+1 multipliers and N two-input adders
- Structures in which the multiplier coefficients are precisely the coefficients of the transfer function are called <u>direct form structures</u>

### 8.3 Basic FIR Digital FilterStructures

 A direct form realization --- convolution sum description as indicated below for N = 4



### 8.3 Basic FIR Digital FilterStructures

An analysis of this structure yields

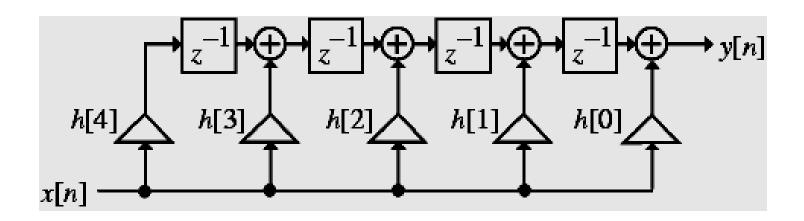
$$y[n] = h[0]x[n] + h[1]x[n-1] + h[2]x[n-2] + h[3]x[n-3] + h[4]x[n-4]$$

which is precisely of the form of the convolution sum description

 The direct form structure shown on the previous slide is also known as a <u>tapped</u> <u>delay line</u> or a <u>transversal filter</u>

## 8.3 Basic FIR Digital FilterStructures

• The transpose of the direct form structure shown earlier is indicated below



## 8.3.2 Cascade Form FIR Digital Filter Structures

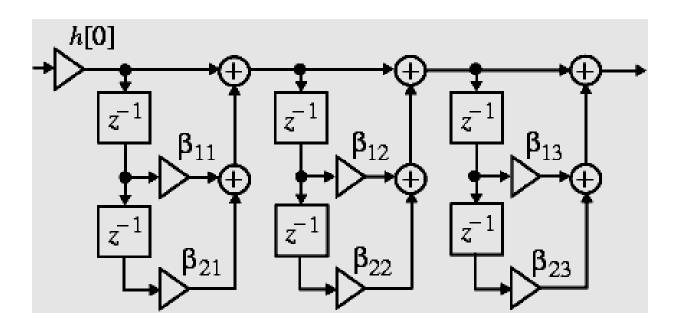
- A higher-order FIR transfer function can also be realized as a cascade of second-order FIR sections and possibly a first-order section
- To this end we express H(z) as

$$H(z) = h[0] \prod_{k=1}^{K} (1 + \beta_{1k} z^{-1} + \beta_{2k} z^{-2})$$

where K = N/2 if N is even, and K = (N+1)/2 if N is odd, with  $\beta_{2K} = 0$ 

## 8.3.2 Cascade Form FIR Digital Filter Structures

• A cascade realization for N = 6 is shown below



Each second-order section in the above structure can also be realized in the direct form

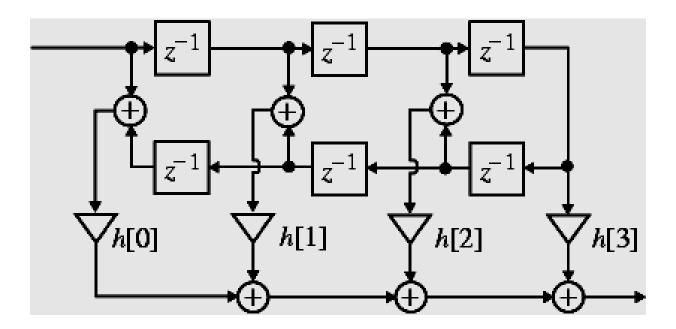
- The symmetry (or anti-symmetry) property of a linear-phase FIR filter can be exploited to reduce the number of multipliers into almost half of that in the direct form implementations
- Consider a length-7 Type 1 FIR transfer function with a symmetric impulse response

$$H(z) = h[0] + h[1]z^{-1} + h[2]z^{-2} + h[3]z^{-3}$$
$$+ h[2]z^{-4} + h[1]z^{-5} + h[0]z^{-6}$$

#### • Rewriting H(z) in the form

$$H(z) = h[0](1+z^{-6}) + h[1](z^{-1}+z^{-5}) + h[2](z^{-2}+z^{-4}) + h[3]z^{-3}$$

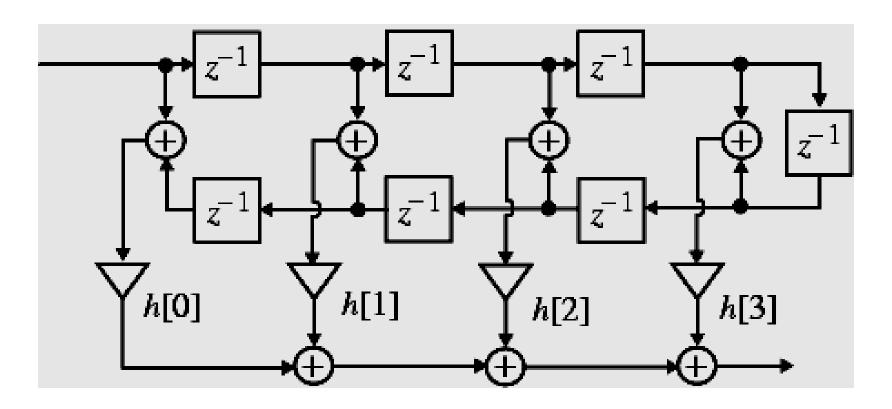
#### we obtain the realization shown below



- A similar decomposition can be applied to a Type 2 FIR transfer function
- For example, a length-8 Type 2 FIR transfer function can be expressed as

$$H(z) = h[0](1+z^{-7}) + h[1](z^{-1}+z^{-6})$$
$$+ h[2](z^{-2}+z^{-5}) + h[3](z^{-3}+z^{-4})$$

• The corresponding realization is shown on the next slide



 Note: The Type 1 linear-phase structure for a length-7 FIR filter requires 4 multipliers, whereas a direct form realization requires 7 multipliers

#### 8.3.3 Linear-Phase FIR Structures

- Note: The Type 2 linear-phase structure for a length-8 FIR filter requires 4 multipliers, whereas a direct form realization requires 8 multipliers
- Similar savings occurs in the realization of Type 3 and Type 4 linear-phase FIR filters with antisymmetric impulse responses

#### 8.4 Basic IIR Digital Filter Structures

- The causal IIR digital filters we are concerned with in this course are characterized by a real rational transfer function of z<sup>-1</sup> or, equivalently by a constant coefficient difference equation
- From the difference equation representation, it can be seen that the realization of the causal IIR digital filters requires some form of <a href="feedback">feedback</a>

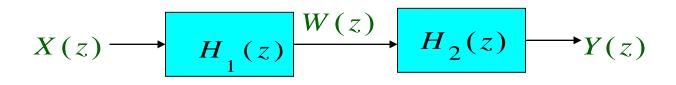
#### 8.4 Basic IIR Digital Filter Structures

- An N-th order IIR digital transfer function is characterized by 2N+1 unique coefficients, and in general, requires 2N+1 multipliers and 2N two-input adders for implementation
- Direct form IIR filters: Filter structures in which the multiplier coefficients are precisely the coefficients of the transfer function

• Consider for simplicity a 3rd-order IIR filter with a transfer function

$$H(z) = \frac{P(z)}{D(z)} = \frac{p_0 + p_1 z^{-1} + p_2 z^{-2} + p_3 z^{-3}}{1 + d_1 z^{-1} + d_2 z^{-2} + d_3 z^{-3}}$$

• We can implement H(z) as a cascade of two filter sections as shown on the next slide



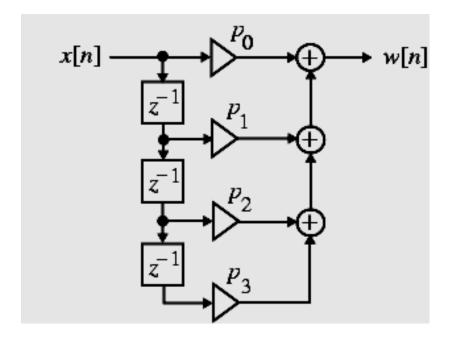
$$H_1(z) = \frac{W(z)}{X(z)} = P(z) = p_0 + p_1 z^{-1} + p_2 z^{-2} + p_3 z^{-3}$$

$$H_{2}(z) = \frac{Y(z)}{W(z)} = \frac{1}{D(z)} = \frac{1}{1 + d_{1}z^{-1} + d_{2}z^{-2} + d_{3}z^{-3}}$$

• The time-domain representation of  $H_1(z)$  is given by

$$w[n] = p_0 x[n] + p_1 x[n-1] + p_2 x[n-2] + p_3 x[n-3]$$

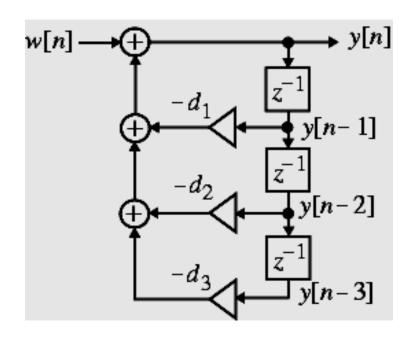
• The filter section  $H_1(z)$  can be seen to be an FIR filter and can be realized as shown right



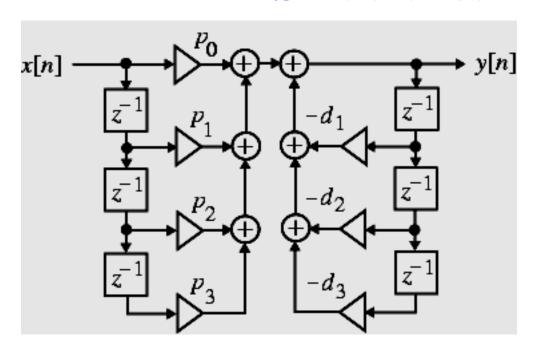
• The time-domain representation of  $H_2(z)$  is given by

$$y[n] = w[n] - d_1 y[n-1] - d_2 y[n-2] - d_3 y[n-3]$$

•Realization of follows from the above equation and is shown on the right



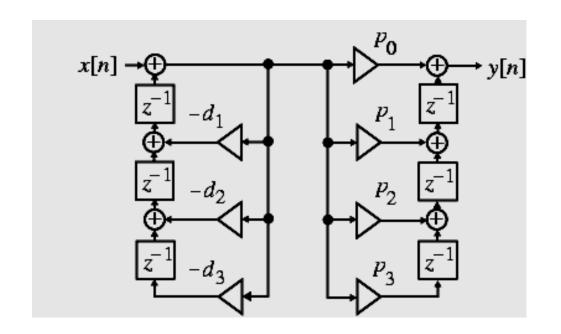
• A cascade of the two structures realizing  $H_1(z)$  and  $H_2(z)$  leads to the realization of H(z) shown below and is known as the <u>Direct Form I</u> structure



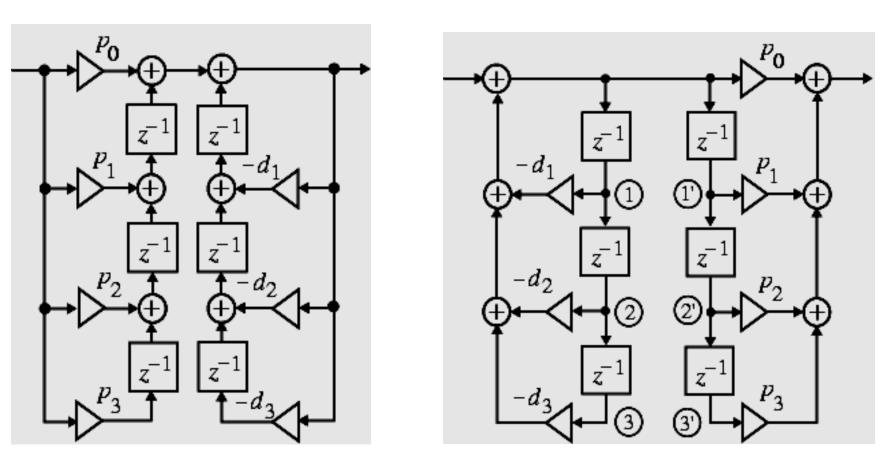
**Direct Form I** 

 Note: The direct form I structure is <u>noncanonic</u> as it employs 6 delays to realize a 3rd-order transfer function

•A transpose of the direct form I structure is shown on the following figure and is called the direct form It structure

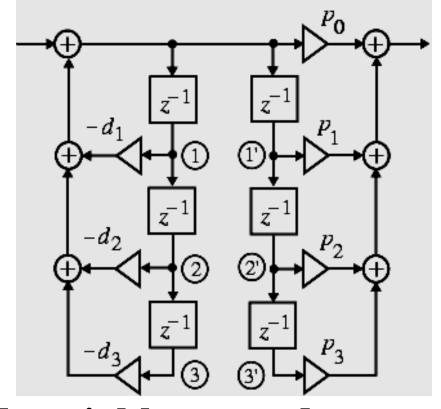


direct form It



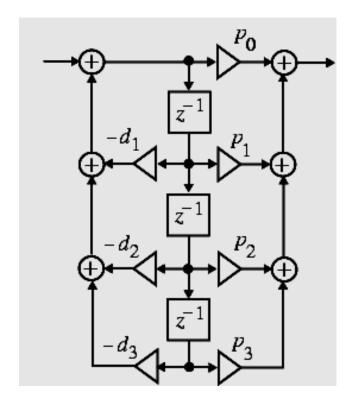
Other noncanonic direct form structures

Observe in the direct form structure shown right, the signal variable at nodes (1) and (1') are the same, and hence the two top delays can be shared

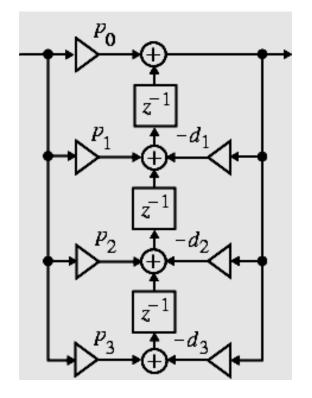


• Likewise, the signal variables at nodes and (2) are the same, permitting the (2) sharing of the middle two delays

Sharing of all delays reduces the total number of delays to 3 resulting in a canonic realization shown in following



**Direct form II** 



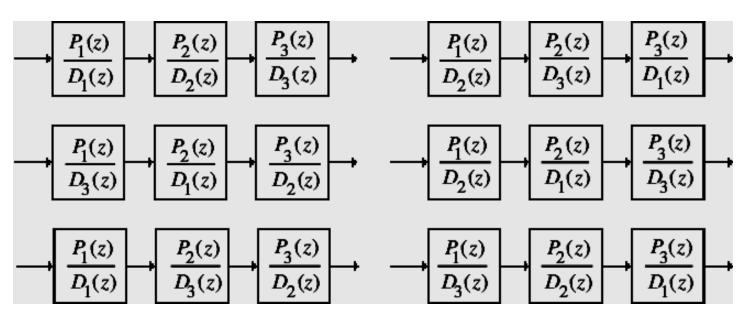
**Direct form IIt** 

• It is noted that, Direct form realizations of an N-th order IIR transfer function should be evident from above 3-order structure.

- By expressing the numerator and the denominator polynomials of the transfer function as a product of polynomials of lower degree, a digital filter can be realized as a cascade of low-order filter sections
- Consider, for example, H(z) = P(z)/D(z) expressed as

$$H(z) = \frac{P(z)}{D(z)} = \frac{P_1(z)P_2(z)P_3(z)}{D_1(z)D_2(z)D_3(z)}$$

 Examples of cascade realizations obtained by different pole-zero pairings are shown below



• There are altogether a total of 36 different cascade realizations of

$$H(z) = \frac{P_1(z)P_2(z)P_2(z)}{D_1(z)D_2(z)D_3(z)}$$

based on pole-zero-pairings and ordering.

Due to finite word-length effects, each such cascade realization behaves differently from others.

• Usually, the polynomials are factored into a product of 1st-order and 2nd-order polynomials:

$$H(z) = p_0 \prod_{k} \left( \frac{1 + \beta_{1k} z^{-1} + \beta_{2k} z^{-2}}{1 + \alpha_{1k} z^{-1} + \alpha_{2k} z^{-2}} \right)$$

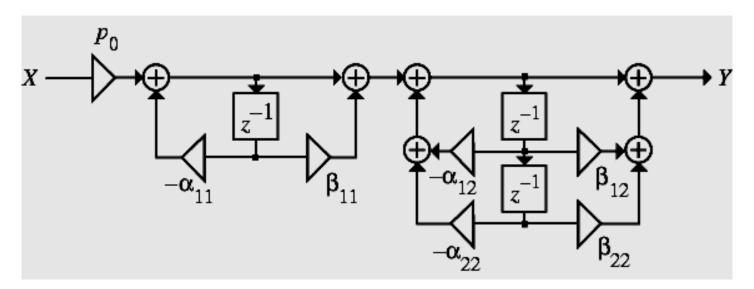
In the above, for a first-order factor

$$\alpha_{2k} = \beta_{2k} = 0$$

Consider the 3rd-order transfer function

$$H(z) = p_0 \left( \frac{1 + \beta_{11} z^{-1}}{1 + \alpha_{11} z^{-1}} \right) \left( \frac{1 + \beta_{12} z^{-1} + \beta_{22} z^{-2}}{1 + \alpha_{12} z^{-1} + \alpha_{22} z^{-2}} \right) (8.28)$$

One possible realization is shown below

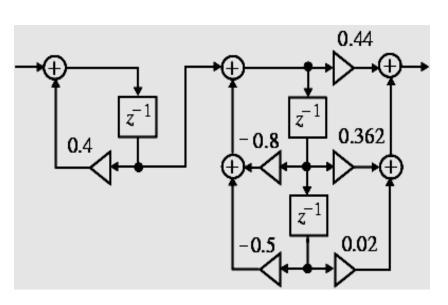


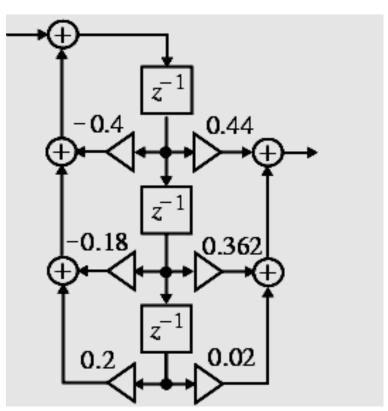
• <u>Example</u> - Direct form II and cascade form realizations of

$$H(z) = \frac{0.44 z^{-1} + 0.362 z^{-2} + 0.02 z^{-3}}{1 + 0.4 z^{-1} + 0.18 z^{-2} - 0.2 z^{-3}}$$

$$= \left(\frac{0.44 + 0.362 z^{-1} + 0.02 z^{-2}}{1 + 0.8 z^{-1} + 0.5 z^{-2}}\right) \left(\frac{z^{-1}}{1 - 0.4 z^{-1}}\right)$$
(8.29)

are shown on the next slide





**Cascade form** 

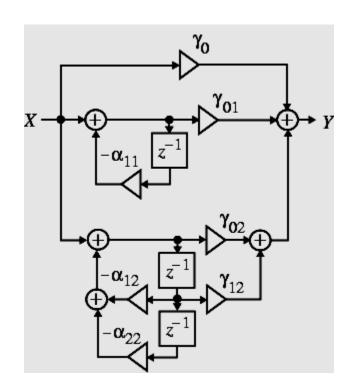
**Direct form II** 

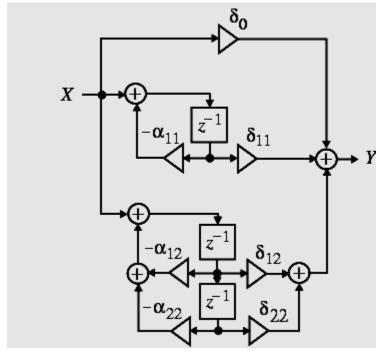
- A partial-fraction expansion of the transfer function in  $z^{-1}$  leads to the parallel form I structure
- Assuming simple poles, the transfer function H(z) can be expressed as

$$H(z) = \gamma_0 + \sum_{k} \left( \frac{\gamma_{0k} + \gamma_{1k} z^{-1}}{1 + \alpha_{1k} z^{-1} + \alpha_{2k} z^{-2}} \right)$$
 (8.30)

$$H(z) = \delta_0 + \sum_{k} \left[ \frac{\delta_{0k} z^{-1} + \delta_{1k} z^{-2}}{1 + \alpha_{1k} z^{-1} + \alpha_{2k} z^{-2}} \right]$$
(8.31)

• The two basic parallel realizations of a 3rdorder IIR transfer function are shown below





**Parallel form I** 

**Parallel form II** 

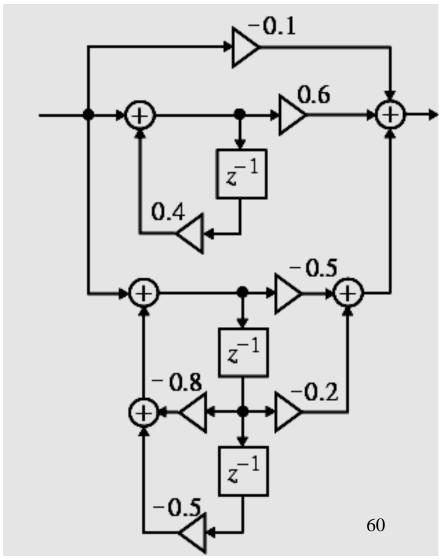
• Example - A partial-fraction expansion of

$$H(z) = \frac{0.44 z^{-1} + 0.362 z^{-2} + 0.02 z^{-3}}{1 + 0.4 z^{-1} + 0.18 z^{-2} - 0.2 z^{-3}}$$

#### in $z^{-1}$ yields

$$H(z) = -0.1 + \frac{0.6}{1 - 0.4z^{-1}} + \frac{-0.5 - 0.2z^{-1}}{1 + 0.8z^{-1} + 0.5z^{-2}}$$

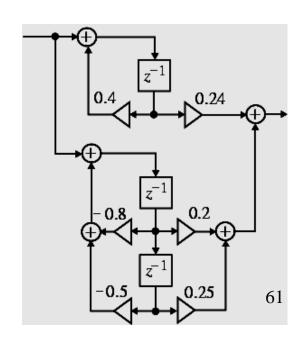
 The corresponding parallel form I realization is shown right



• Likewise, a partial-fraction expansion of H(z) in z yields

$$H(z) = \frac{0.24 z^{-1}}{1 - 0.4 z^{-1}} + \frac{0.2 z^{-1} + 0.25 z^{-1}}{1 + 0.8 z^{-1} + 0.5 z^{-2}}$$

•The corresponding parallel form II realization is shown on the right



#### 8.5 Realization Using MATLAB

- The cascade form requires the factorization of the transfer function which can be developed using the M-file zp2sos
- The statement sos = zp2sos(z, p, k) generates a
  matrix sos containing the coefficients of each 2ndorder section of the equivalent transfer function
  H(z) determined from its pole-zero form

#### •sos is an $L \times 6$ matrix of the form

$$sos = \begin{bmatrix} p_{01} & p_{11} & p_{21} & d_{01} & d_{11} & d_{21} \\ p_{02} & p_{12} & p_{22} & d_{02} & d_{12} & d_{22} \\ \vdots & \vdots & \vdots & \vdots & \vdots & \vdots \\ p_{0L} & p_{1L} & p_{2L} & d_{0L} & d_{1L} & d_{2L} \end{bmatrix}$$

whose i-th row contains the coefficients  $\{p_{il}\}$  and  $\{d_{il}\}$ , of the the numerator and denominator polynomials of the i-th 2nd-order section

• L denotes the number of sections

 The form of the overall transfer function is given by

$$H(z) = \prod_{i=1}^{L} H_{i}(z) = \prod_{i=1}^{L} \frac{p_{0i} + p_{1i}z^{-1} + p_{2i}z^{-2}}{d_{0i} + d_{1i}z^{-1} + d_{2i}z^{-2}}$$

#### Program 8\_1 can be used to factorize an FIR and an IIR transfer function

- Parallel forms I and II can be developed using the functions residuez and residue, respectively
- Program 8\_2 uses these two functions



#### Thanks!

Any questions?