

Chapter-4: Discrete Filter Structures

□ FIR Filter, Structures for FIR Filter:

1. Introduction:

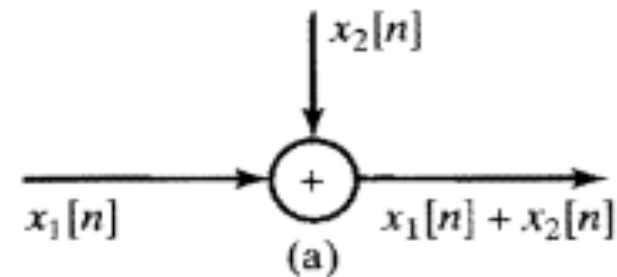
- For the *design of digital filters*, the *system function $H(z)$* or the *impulse response $h[n]$* must be specified. Then the *digital filter structure* can be implemented in *hardware/software* form by its *difference equation* obtained directly from $H(z)$ or $h[n]$.
- To implement the specified *difference equation* of the system, the required basic operations are *addition, delay and multiplication by a constant*.
- The *structures are derived* on the basis of *computational complexity, ease of implementation of finite word length effect* etc.

2. Block Diagram Representation:

- When the system function $H(z)$ or the impulse response $h[n]$ is specified then the digital filters can be implemented or realized using **block diagram**. The following are the **basic elements required for the implementation**.

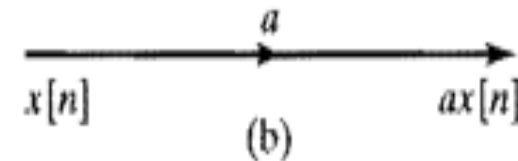
a. An adder:

- It performs the addition of two signals.



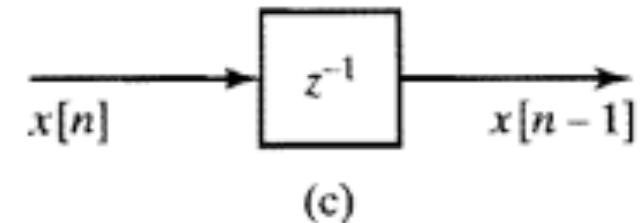
b. A Constant Multiplier:

- It multiplies a signal with a constant ' a '.



c. A Unit Delay Element:

- It delays the input sequence by one element.



3. Advantages of Representing the Digital System (i.e. Filters) in Block Diagram Form:

- i. The computation algorithm can be easily written just by inspection.
- ii. The hardware requirement can be easily determined.
- iii. The relationship between input and output can be easily determined.

4. Canonic and Non-canonic Structures:

- If the ***number of delays*** in the structure or realization block diagram is ***equal*** to the ***order of the difference equation or the order of the system function*** of a digital filter, then the structure is canonic otherwise non-canonic.

4. Structures for FIR Systems:

- A **causal FIR system** can be described by the **difference equation**

$$y[n] = \sum_{k=0}^{M-1} b_k x[n - k] \quad \text{.....i}$$

Or, equivalently, by the **system function**

$$H(z) = \sum_{k=0}^{M-1} b_k z^{-k} \quad \text{.....ii}$$

- If we replace b_k by $h[k]$, we obtain

$$y[n] = \sum_{k=0}^{M-1} h[k] x[n - k] \quad \text{.....iii}$$

(convolution sum)

Or,
$$H(z) = \sum_{k=0}^{M-1} h[k] z^{-k} \quad \text{.....iv}$$

- Therefore, we can write

$$h[n] = \begin{cases} b_n, & 0 \leq n \leq M - 1 \\ 0, & \text{Otherwise} \end{cases} \quad \text{.....v}$$

- **Note that FIR filter is called all-zero filter (or comb filter).**

➤ There are several methods for implementation of FIR system. They are:

- a. Direct form structures
- b. Cascade form structures
- c. Frequency sampling structures
- d. Lattice structures
- e. Linear phase structures

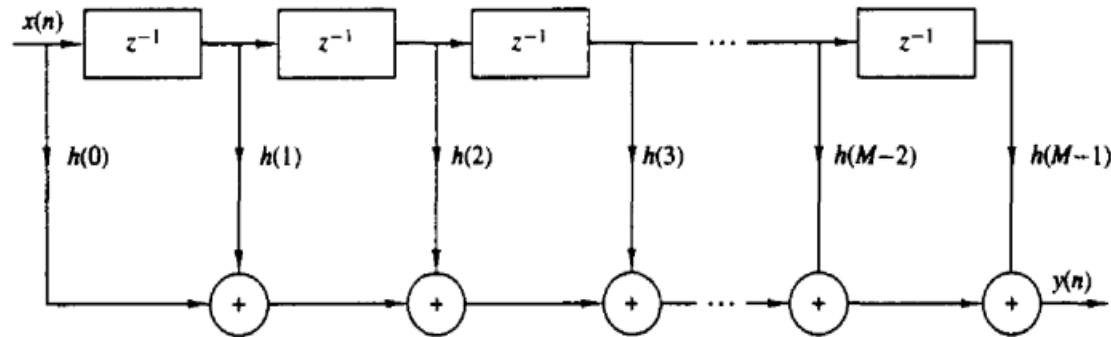
a. Direct Form Structures:

➤ The direct-form realization follows immediately from the non-recursive difference equation or, equivalently, by the convolution summation

$$y[n] = \sum_{k=0}^{M-1} h[k] x[n - k] \quad \text{.....i}$$

$$\text{Or, } y[n] = h[0] x[n] + h[1] x[n - 1] + \cdots + h[M - 1] x[n - (M - 1)] \quad \text{.....ii}$$

➤ The direct form structure can be realized as shown in figure below:



- Because of the ***chain of delay elements across the top of the diagram***, this structure is also referred to as a ***tapped delay line structure or a transversal filter structure***.

b. Cascade Form Structures:

- The ***cascade form structure*** of FIR system is obtained by ***factoring the polynomial system function*** as:

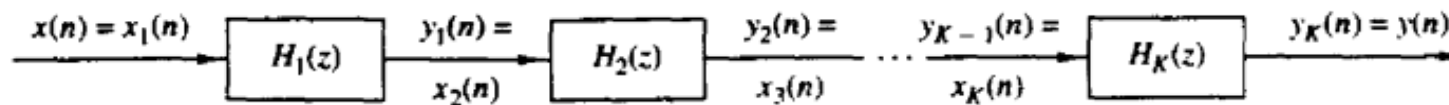
$$H(z) = \prod_{k=1}^K H_k(z)$$

where

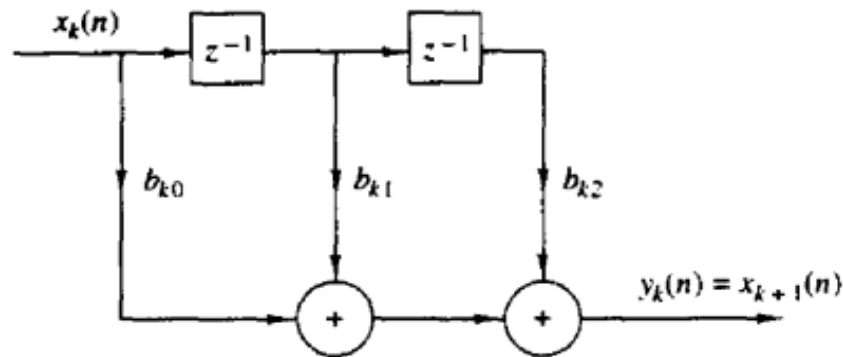
$$H_k(z) = b_{k0} + b_{k1}z^{-1} + b_{k2}z^{-2} \quad k = 1, 2, \dots, K$$

where, $K = \text{integer part of } \frac{M+1}{2}.$

- If ***M is odd*** then $K = \frac{M+1}{2}.$
- If ***M is even*** then $K = \frac{M}{2}$ with $b_{k2} = 0.$



(a)



(b)

Fig: Cascade realization of an FIR system

Example 5.5: Obtain the direct-form and cascade-form realizations for the transfer function of an FIR system given by

$$(a) \quad H(z) = (1 + 2z^{-1} - z^{-2})(1 + z^{-1} - z^{-2})$$

$$(b) \quad H(z) = 1 + \frac{5}{2}z^{-1} + 2z^{-2} + 2z^{-3}$$

Solution:

$$(a) \quad H(z) = (1 + 2z^{-1} + z^{-2})(1 + z^{-1} - z^{-2})$$

(i) Direct-form realization:

→ For this, we have

$$\begin{aligned} H(z) &= (1 + 2z^{-1} - z^{-2})(1 + z^{-1} - z^{-2}) \\ &= 1 + z^{-1} - z^{-2} + 2z^{-1} + 2z^{-2} - 2z^{-3} - z^{-2} - z^{-3} + z^{-4} \end{aligned}$$

$$\therefore H(z) = 1 + 3z^{-1} - 3z^{-3} + z^{-4}$$

The direct-form realization is shown in Fig 5.22.

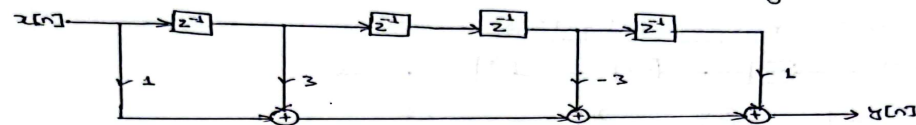


Fig 5.22: Direct-form realization

(ii) cascade-form realization:

→ For this, we have

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

$$H(z) = H_1(z) H_2(z)$$

$$\text{where, } H_1(z) = (1 + 2z^{-1} - z^{-2})$$

$$\text{and } H_2(z) = (1 + z^{-1} - z^{-2})$$

The cascade-form realization is shown in Fig 5.23.

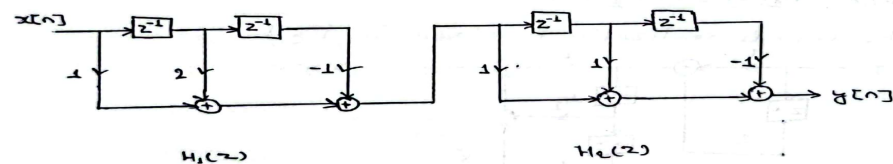


Fig 5.23: cascade-form realization

Notes:

Fig 5.23 can be redrawn for cascade-form realization as shown in Fig 5.24 below.

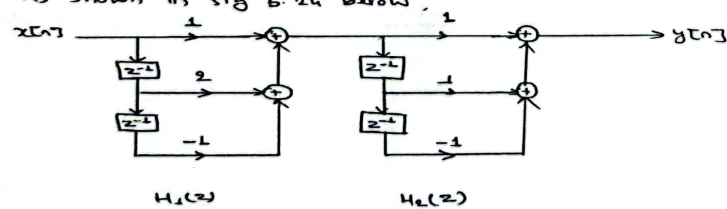


Fig 5.24

(24)

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

(i) Direct-form realization:

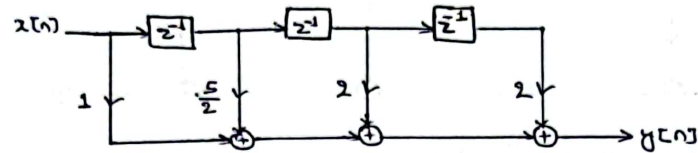


Fig 5.25: Direct-form realization

(ii) Cascade-form realization:

⇒ For this, we have

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

$$\text{or } H(z) = (1 + 2z^{-1})(1 + \frac{1}{2}z^{-1} + z^{-2}) = H_1(z) H_2(z)$$

$$\therefore H_1(z) = (1 + 2z^{-1})$$

$$\text{and } H_2(z) = (1 + \frac{1}{2}z^{-1} + z^{-2})$$

The cascade-realization is shown in fig

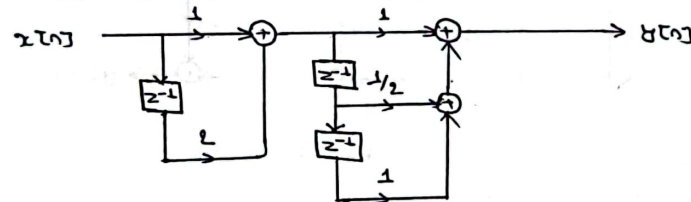


Fig 5.26; cascade-form realization

c. Frequency-Sampling Structures:

- The ***frequency-sampling realization*** is an alternative structure for an FIR filter in which the ***parameters that characterize the filter are the values of the desired frequency response instead of the impulse response $h(n)$.***

- To ***derive the frequency sampling structure***, we specify ***the desired frequency response at a set of equally spaced frequencies***, namely

$$\omega_k = \frac{2\pi}{M}(k + \alpha) \quad \begin{array}{l} k = 0, 1, \dots, \frac{M-1}{2} \text{ for } \mathbf{M} \text{ odd.} \\ k = 0, 1, \dots, \frac{M}{2} - 1 \text{ for } \mathbf{M} \text{ even.} \\ \alpha = 0 \text{ or } \frac{1}{2}. \end{array}$$

and ***solve for the $h[n]$.***

- The ***frequency response*** is

$$\mathbf{H}(e^{j\omega}) = \mathbf{H}(\omega) = \sum_{n=0}^{M-1} h[n] e^{-j\omega n}$$

and values of $\mathbf{H}(e^{j\omega})$ at $\omega_k = \frac{2\pi}{M}(k + \alpha)$ are

$$\begin{aligned}
 H(k + \alpha) &= H\left(\frac{2\pi}{M}(k + \alpha)\right) \\
 &= \sum_{n=0}^{M-1} h(n) e^{-j2\pi(k+\alpha)n/M} \quad k = 0, 1, \dots, M-1
 \end{aligned}$$

- The set of values $\frac{2\pi}{M}(k + \alpha)$ are called the **frequency samples of $H(e^{j\omega})$** . In the case, where $\alpha = 0$, $\{H(k)\}$ corresponds to the **M -point DFT of $\{h[n]\}$** .
- The **impulse response $h[n]$** of above equation is

$$h(n) = \frac{1}{M} \sum_{k=0}^{M-1} H(k + \alpha) e^{j2\pi(k+\alpha)n/M} \quad n = 0, 1, \dots, M-1$$

When $\alpha = 0$, above equation is simply the **IDFT of $H(k)$** .

- Now, if we use above equation to substitute for $h[n]$ in the z-transform $H(z)$, we have

$$\begin{aligned}
H(z) &= \sum_{n=0}^{M-1} h(n)z^{-n} \\
&= \sum_{n=0}^{M-1} \left[\frac{1}{M} \sum_{k=0}^{M-1} H(k+\alpha) e^{j2\pi(k+\alpha)n/M} \right] z^{-n} \\
H(z) &= \sum_{k=0}^{M-1} H(k+\alpha) \left[\frac{1}{M} \sum_{n=0}^{M-1} (e^{j2\pi(k+\alpha)/M} z^{-1})^n \right] \\
&= \frac{1 - z^{-M} e^{j2\pi\alpha}}{M} \sum_{k=0}^{M-1} \frac{H(k+\alpha)}{1 - e^{j2\pi(k+\alpha)/M} z^{-1}}
\end{aligned}$$

- Thus, the **system function** $H(z)$ is characterized by the set of frequency samples $H(k+\alpha)$ instead of $\{h[n]\}$.
- We view this FIR filter realization as a **cascade of two filters** [i.e., $H(z) = H_1(z) \cdot H_1(z)$]

$$H_1(z) = \frac{1}{M} (1 - z^{-M} e^{j2\pi\alpha}) \quad (\text{all zero filter or comb filter})$$

Its **zeros** are located at equally spaced points on the unit circle at

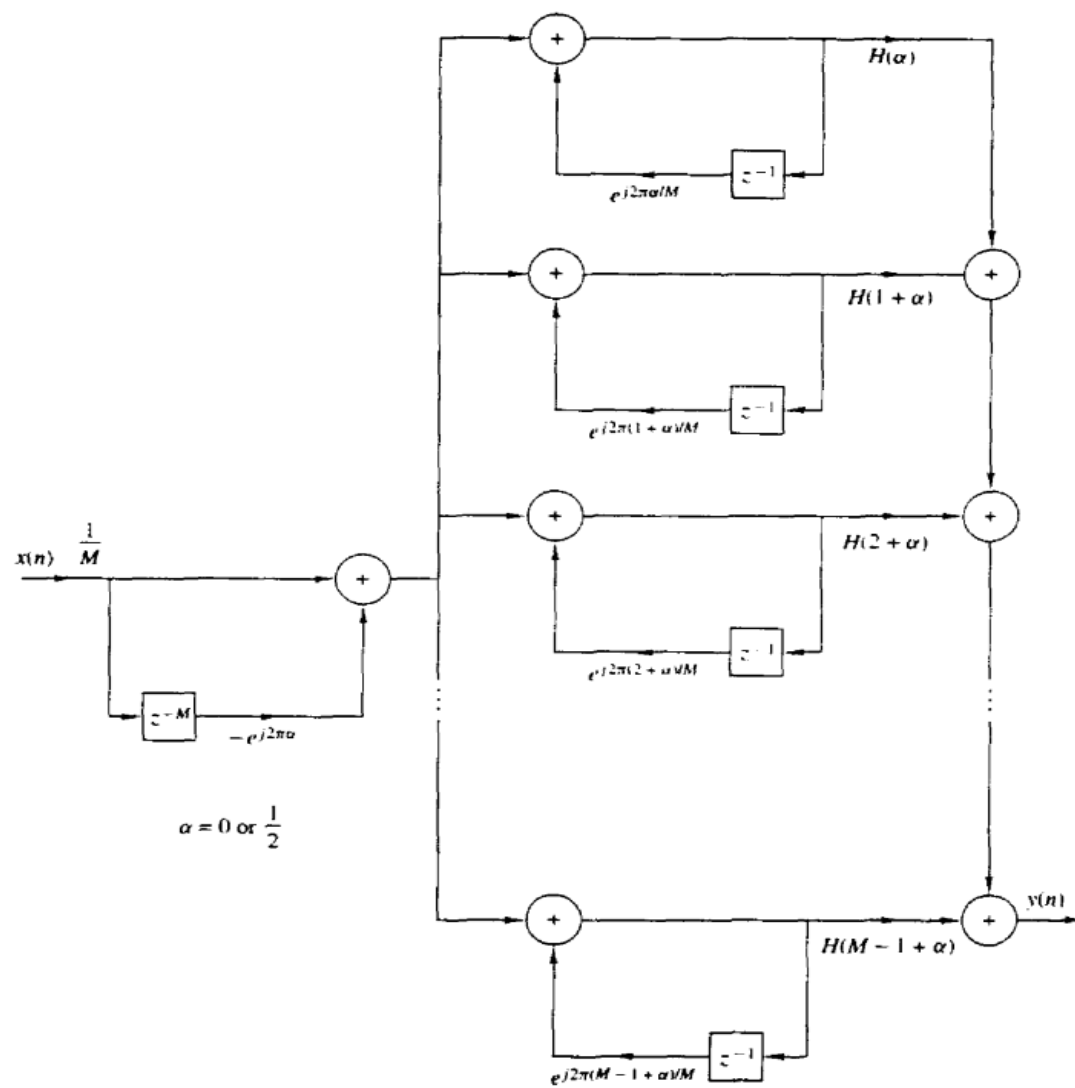
$$z_k = e^{j2\pi(k+\alpha)/M}, \quad k = 0, 1, \dots, M-1$$

and
$$H_2(z) = \sum_{k=0}^{M-1} \frac{H(k+\alpha)}{1 - e^{j2\pi(k+\alpha)/M} z^{-1}}$$

consists **parallel banks of single-pole filters** with resonant frequencies

$$p_k = e^{j2\pi(k+\alpha)/M}, \quad k = 0, 1, \dots, M-1$$

➤ The cascade realization is shown in figure below:



➤ For **linear phase**, $H(k) = H^*(M - k)$ for $\alpha = 0$, and

$$H\left(k + \frac{1}{2}\right) = H^*\left(M - k - \frac{1}{2}\right) \quad \text{for } \alpha = \frac{1}{2}$$

As a result, **a pair of single pole filters can be combined to form a single two-pole filter** with ($\alpha = 0$).

$$H_2(z) = \frac{H(0)}{1 - z^{-1}} + \sum_{k=1}^{(M-1)/2} \frac{A(k) + B(k)z^{-1}}{1 - 2\cos(2\pi k/M)z^{-1} + z^{-2}}$$

$$H_2(z) = \frac{H(0)}{1 - z^{-1}} + \frac{H(M/2)}{1 + z^{-1}} + \sum_{k=1}^{(M/2)-1} \frac{A(k) + B(k)z^{-1}}{1 - 2\cos(2\pi k/M)z^{-1} + z^{-2}}$$

where, by definition

$$A(k) = H(k) + H(M - k)$$

$$B(k) = H(k)e^{-j2\pi k/M} + H(M - k)e^{j2\pi k/M}$$

Similar expressions can be obtained for $\alpha = \frac{1}{2}$.

d. Lattice Structures:

- **Lattice filters** are used extensively in **digital speech processing** and in the implementation of **adaptive filters**. Let an FIR filter with system function

$$H(z) = A_m(z), \quad m = 0, 1, \dots, M-1$$
$$H(z) = A_m(z) = 1 + \sum_{k=1}^m a_m(k)z^{-k}, \quad m \geq 1 \quad \text{.....1}$$

with $A_0(z) = 1$, and impulse response

$$h[n] = \begin{cases} 1, & n = 0 \\ a_m(n), & n = 1, 2, \dots, m \end{cases}$$

OR,

$$Y(z) = X(z) \left[1 + \sum_{k=1}^m a_m(k)z^{-k} \right]$$
$$Y(z) = X(z) + X(z) \sum_{k=1}^m a_m(k)z^{-k}$$

- Taking inverse z-transform, we get

$$y[n] = x[n] + \sum_{k=1}^m a_m(k)x[n-k] \quad \text{.....2}$$

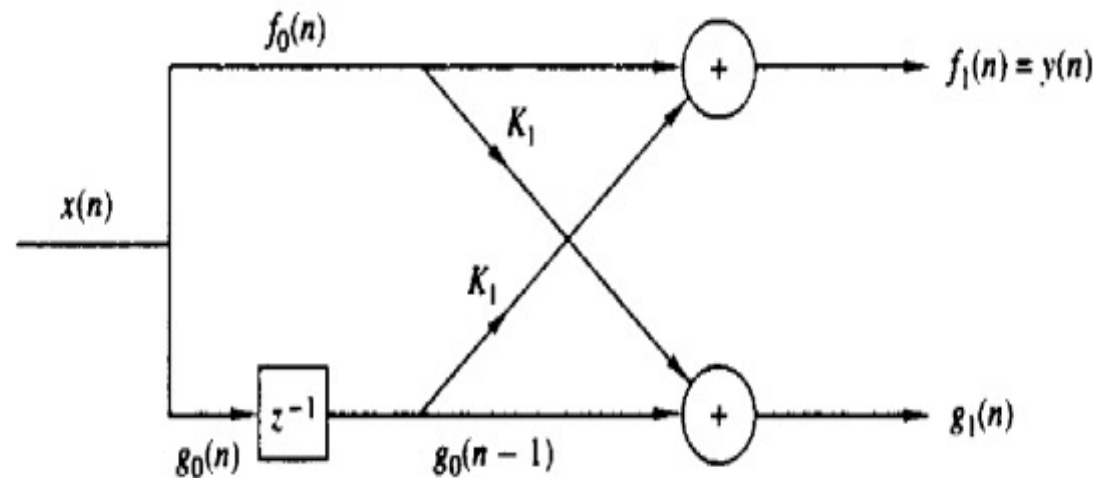
(*m is the degree of polynomial*)

- For $m = 1$:

- Equation 2 reduces to

$$y[n] = x[n] + a_1(1)x[n-1] \quad \text{.....3}$$

This *single stage lattice filter* can be realized as shown in figure below:



➤ From above figure, we have

$$x[n] = f_0[n] = g_0[n] \quad \text{.....4}$$

$$y[n] = f_1[n] = f_0[n] + k_1 g_0[n-1] = x[n] + k_1 x[n-1] \quad \text{.....5}$$

$$g_1[n] = k_1 f_0[n] + g_0[n-1] = k_1 x[n] + x[n-1] \quad \text{.....6}$$

- Comparing equations 3 and 4, we get

$$a_1(0) = 1, \quad a_1(1) = k_1 \quad \text{.....7}$$

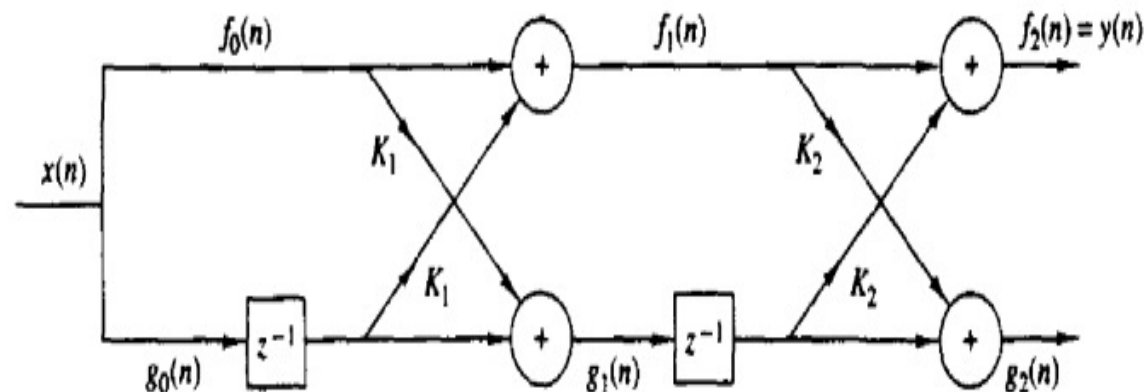
- Where, k_1 is **lattice coefficient**. Similarly, $a_1(0)$ and $a_1(1)$ are **direct form coefficients**.

- For $m = 2$:

- Equation 2 reduces to

$$y[n] = x[n] + a_2(1)x[n - 1] + a_2(2)x[n - 2] \quad \text{.....8}$$

the **two-stage lattice structure** for this is shown in figure below:



➤ From above figure, we have

$$\mathbf{y[n] = f_2[n] = f_1[n] + k_2 g_1[n - 1]}$$

But,
$$f_1[n] = f_0[n] + k_1 g_0[n - 1] = x[n] + k_1 x[n - 1]$$

and,
$$g_1[n - 1] = k_1 f_0[n - 1] + g_0[n - 2] = k_1 x[n - 1] + x[n - 2]$$

therefore,
$$y[n] = f_2[n] = x[n] + k_1 x[n - 1] + k_2 \{k_1 x[n - 1] + x[n - 2]\}$$

$$\mathbf{y[n] = f_2[n] = x[n] + k_1(1 + k_2)x[n - 1] + k_2x[n - 2] \quad \text{.....9}}$$

➤ From equations 8 and 9, we get

$$\mathbf{a_2(0) = 1, \quad a_2(1) = k_1(1 + k_2), \quad a_2(2) = k_2}$$

➤ Also, from above figure

$$\mathbf{g_2[n] = k_2 f_1[n] + g_1[n - 1]}$$

➤ Therefore,

$$g_2[n] = k_2 x[n] + k_1(1 + k_2)x[n - 1] + x[n - 2] \quad \text{.....10}$$

Note that two sets of filter coefficients in $f[n]$ and $g[n]$ are in reverse order.

○ For $m = M - 1$:

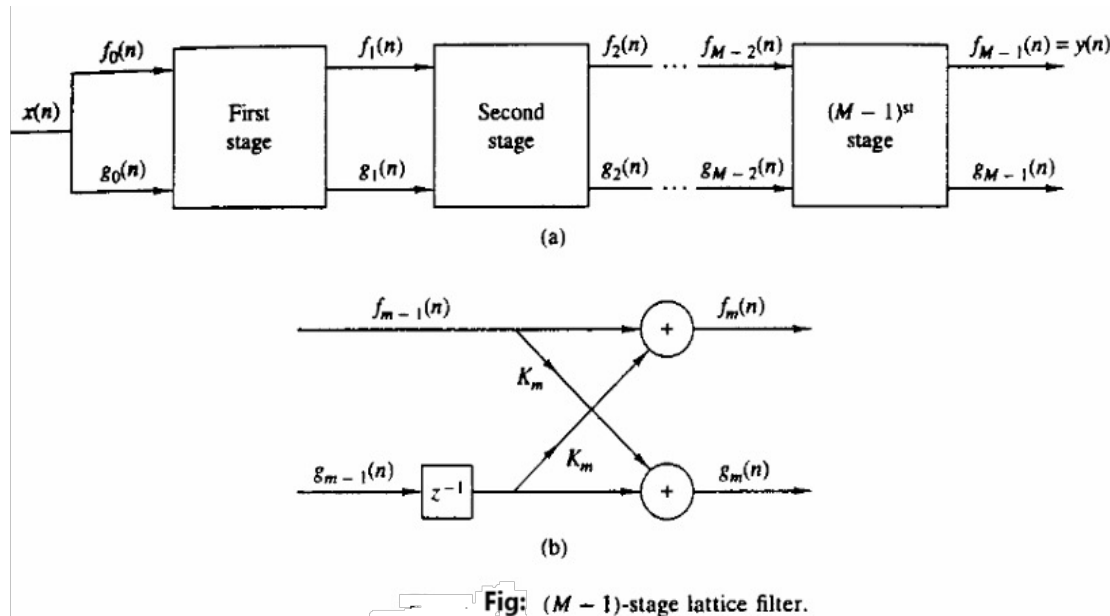
➤ For M-1 stage filter, we have

$$\begin{aligned} f_0(n) &= g_0(n) = x(n) \\ f_m(n) &= f_{m-1}(n) + K_m g_{m-1}(n-1) \quad m = 1, 2, \dots, M-1 \end{aligned} \quad \text{.....11}$$

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

➤ The output of M-1 stage is

$$y[n] = f_{M-1}[n] \quad \text{.....12}$$



- **Conversion of Lattice Coefficients to Direct Form Coefficients:**

➤ In general,

$$a_m(0) = 1$$

$$a_m(m) = k_m$$

$$a_m(k) = a_{m-1}(k) + a_m(m) a_{m-1}(m - k)$$

where,

$$1 \leq k \leq m - 1 \text{ and } m = 1, 2, \dots, M - 1$$

- **Conversion of Direct Form Coefficients to Lattice Coefficients :**

➤ In general,

$$a_m(0) = 1$$

$$k_m = a_m(m)$$

$$a_m(k) = a_{m-1}(k) + a_m(m) a_{m-1}(m - k)$$

$$a_{m-1}(k) = \frac{a_m(k) - a_m(m) a_m(m - k)}{1 - \{a_m(m)\}^2}$$

where,

$$1 \leq k \leq m - 1 \text{ and } m = 1, 2, \dots, M - 1$$

□ IIR Filter, Structures for IIR Filter:

- Causal IIR systems are characterized by the difference equation as:

$$y[n] = -\sum_{k=1}^N a_k y[n-k] + \sum_{k=0}^M b_k x[n-k] \quad \text{.....1}$$

Or, $\sum_{k=0}^N a_k y[n-k] = \sum_{k=0}^M b_k x[n-k] , a_0 = 1$

- Taking z-transform on both sides of equation 1, we get

$$Y(z) = -\sum_{k=1}^N a_k z^{-k} Y(z) + \sum_{k=0}^M b_k z^{-k} X(z)$$

$$Y(z)[1 + \sum_{k=1}^N a_k z^{-k}] = \sum_{k=0}^M b_k z^{-k} X(z)$$

$$H(z) = \frac{Y(z)}{X(z)} = \frac{\sum_{k=0}^M b_k z^{-k}}{1 + \sum_{k=1}^N a_k z^{-k}} \quad \text{.....2}$$

where, $H(z)$ = system function, $a_0 \neq 0$ and $M \leq N$

- From 1 and 2, we observe that the ***realization of IIR systems, i. e. filters , involves a recursive computational algorithm.***

- **Notes:**

- A. Non-recursive and Recursive Systems:**

- **Non-recursive System:**

- If the ***output of a system is the function of the present and past values of the inputs*** only then ***the system is known as non-recursive system.*** Mathematically,

$$y[n] = F\{x[n], x[n - 1], \dots x[n - M]\} \quad \dots 1$$

- A ***causal FIR system is non-recursive*** system. consider a causal FIR system,

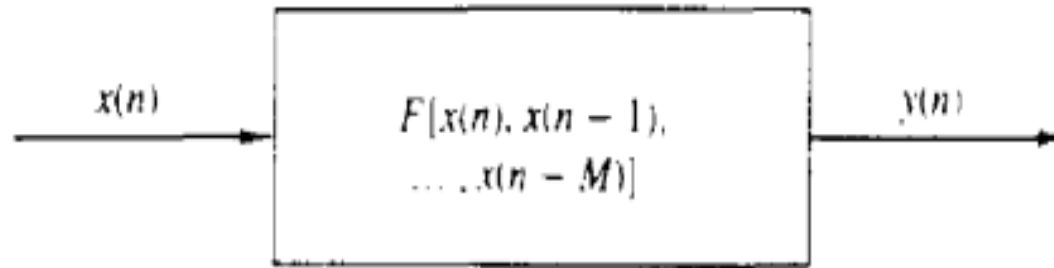
$$y[n] = \sum_{k=0}^{M-1} h[k]x[n - k] \quad \dots 2$$

where the output of the system is the function of present and past inputs. Thus, the causal FIR system is non-recursive.

- A **non-recursive system** can be represented in terms of **difference equation** as

$$y[n] = \sum_{k=0}^M b_k x[n - k] \quad \text{.....3}$$

- Non-recursive systems **does not have feedback path**.



- **Recursive System:**

- If the **output of a system is the function of the present and past values of the inputs as well past outputs** then **the system is known as recursive system**. Mathematically,

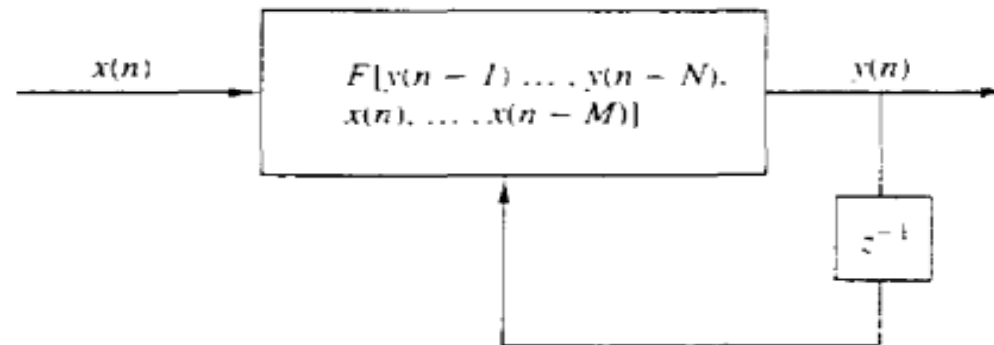
$$y[n] = F\{y[n-1], \dots, y[n-N], x[n], x[n-1], \dots, x[n-M]\} \quad \text{.....4}$$

(causal recursive system)

- A **recursive LTI system** is characterized by **difference equation** as

$$y[n] = -\sum_{k=1}^N a_k y[n-k] + \sum_{k=0}^M b_k x[n-k] \quad \text{.....5}$$

- **Recursive realization reduces the memory requirements, additions, and multiplications.**
- Recursive systems **have feedback path.**



B. FIR (Finite-Duration Impulse Response) and IIR (Infinite-Duration Impulse Response) System:

➤ Let $h[n]$ be the impulse response of a LTI system. Then LTI system can be subdivided into two types

1. FIR (Finite-Duration Impulse Response) System, and
2. IIR (Infinite-Duration Impulse Response) System

1. FIR System:

➤ For causal FIR system, the convolution sum formula is

$$y[n] = \sum_{k=0}^{M-1} h[k]x[n - k] \quad \text{.....1}$$

An **FIR system has finite memory of length M and non-recursive.**

➤ The difference equation of FIR system is

$$y[n] = \sum_{k=0}^M b_k x[n - k] \quad \text{.....2}$$

2. IIR System:

- For causal IIR system, the **convolution sum** formula is

$$y[n] = \sum_{k=0}^{\infty} h[k]x[n-k] \quad \text{.....1}$$

An ***FIR system has finite memory of length M and non-recursive.***

- The difference equation of IIR system is

$$y[n] = -\sum_{k=1}^N a_k y[n-k] + \sum_{k=0}^M b_k x[n-k] \quad \text{.....2}$$

- An ***IIR system cannot be realized using convolution sum*** but it is ***realized using difference equation (LCCDE)***.

1. Methods for the Implementation of IIR Systems:

- a. Direct form structures
- b. Cascade form structures
- c. Parallel form structures
- d. Lattice and lattice-ladder structures

a. Direct Form Structures:

➤ The **system function for IIR** system is

$$H(z) = \frac{Y(z)}{X(z)} = \frac{\sum_{k=0}^M b_k z^{-k}}{1 + \sum_{k=1}^N a_k z^{-k}}, a_0 = 1 \quad \text{.....1}$$

➤ Since, the **multiplier coefficients** (a_k and b_k) in the structures are exactly the **coefficients of the system function**, they are called **direct form structures**.

➤ Direct form structures can be studied under:

- i. Direct form I structure
- ii. Direct form II structure

i. Direct form I structure:

- We know that the multiplier coefficients are the coefficients of system function

$$H(z) = \frac{Y(z)}{X(z)} = \frac{\sum_{k=0}^M b_k z^{-k}}{1 + \sum_{k=1}^N a_k z^{-k}}, a_0 = 1 \quad \text{.....1}$$

Or,
$$H(z) = H_1(z) \cdot H_2(z)$$

where,
$$H_1(z) = \sum_{k=0}^M b_k z^{-k}, \text{ all-zero system (non-recursive)} \quad \text{.....2}$$

and
$$H_2(z) = \frac{1}{1 + \sum_{k=1}^N a_k z^{-k}}, \text{ all-pole system (recursive)} \quad \text{.....3}$$

- Then the **direct form I structure** is obtained by **cascading the structures for $H_1(z)$ and $H_2(z)$** .

• **Note:**
$$H(z) = \frac{Y(z)}{X(z)} = \frac{\sum_{k=0}^M b_k z^{-k}}{1 - \sum_{k=1}^N a_k z^{-k}}, a_0 = 1 \quad (\text{in Oppenheim book})$$

- From figure, we have

$$\frac{W(z)}{X(z)} = H_1(z)$$

$$\frac{W(z)}{X(z)} = \sum_{k=0}^M b_k z^{-k}$$

$$W(z) = X(z) \sum_{k=0}^M b_k z^{-k}$$

- Or,

$$W(z) = b_0 X(z) + b_1 z^{-1} X(z) + b_2 z^{-2} X(z) + \dots + b_M z^{-M} X(z) \quad \dots 1$$

- Taking inverse z-transform, we get

$$w[n] = b_0 x[n] + b_1 x[n-1] + b_2 x[n-2] + \dots + b_M x[n-M]$$

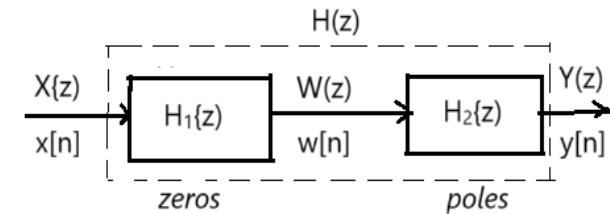
$$\therefore w[n] = \sum_{k=0}^M b_k x[n-k]$$

.....2

- Similarly,

$$\frac{Y(z)}{W(z)} = H_2(z)$$

$$\frac{Y(z)}{W(z)} = \frac{1}{1 + \sum_{k=1}^N a_k z^{-k}}$$



➤ Or,

$$Y(z) + \sum_{k=1}^N a_k z^{-k} Y(z) = W(z)$$

$$Y(z) = -\sum_{k=1}^N a_k z^{-k} Y(z) + W(z)$$

.....3

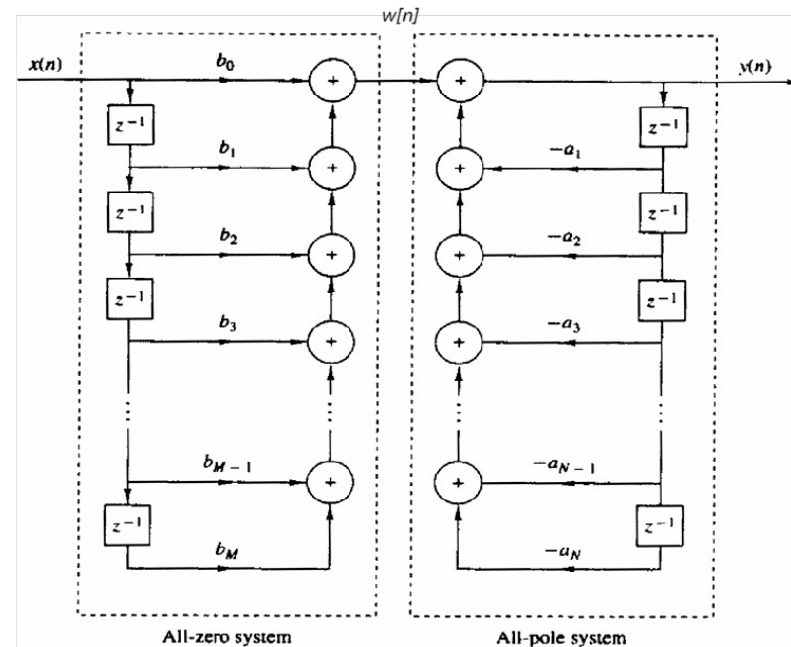
➤ Taking inverse z-transform, we get

$$y[n] = -\sum_{k=1}^N a_k y[n-k] + w[n]$$

.....4

$$y[n] = -a_1 y[n-1] - a_2 y[n-2] - \dots - a_N y[n-N] + w[n]$$

➤ Therefore, direct form-I realization becomes as shown in fig below:



- **Computational Complexity:**

➤ This realization requires:

- i. Number of additions = $M+N$
- ii. Number of multiplications = $M+N+1$
- iii. Number of memory locations (delay elements) = $M+N+1$

ii. Direct form II structure:

- Since, we are dealing with the LTI systems, we can interchange the positions of $H_1(z)$ and $H_2(z)$. This property gives the direct form-II structure.
- In direct form-II realization(or structure), poles of $H(z)$ is realized first and then the zeros second.

$$H(z) = H_1(z).H_2(z)$$

where, $H_1(z) = \frac{1}{1+\sum_{k=1}^N a_k z^{-k}}$, all-pole system (recursive)

and $H_2(z) = \sum_{k=0}^M b_k z^{-k}$, all- zero system (non-recursive)

- From figure, we have

$$\frac{W(z)}{X(z)} = H_1(z)$$

$$\frac{W(z)}{X(z)} = \frac{1}{1 + \sum_{k=1}^N a_k z^{-k}}$$

$$W(z) = X(z) \frac{1}{1 + \sum_{k=1}^N a_k z^{-k}}$$

- Or,

$$W(z) \{ 1 + \sum_{k=1}^N a_k z^{-k} \} = X(z)$$

$$W(z) = - \sum_{k=1}^N a_k z^{-k} W(z) + X(z)$$

.....1

- Taking inverse z-transform, we get

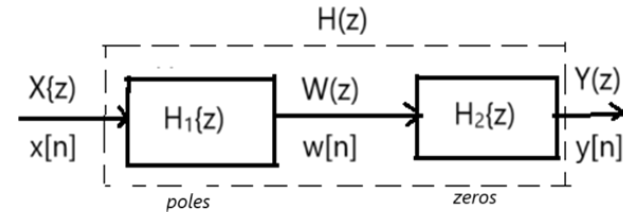
$$w[n] = - \sum_{k=1}^N a_k w[n - k] + x[n]$$

.....2

- Similarly,

$$\frac{Y(z)}{W(z)} = H_2(z)$$

$$\frac{Y(z)}{W(z)} = \sum_{k=0}^M b_k z^{-k}$$



➤ Or,

$$\frac{Y(z)}{W(z)} = \sum_{k=0}^M b_k z^{-k}$$

$$Y(z) = \sum_{k=0}^M b_k z^{-k} W(z)$$

➤ Taking inverse z-transform, we get

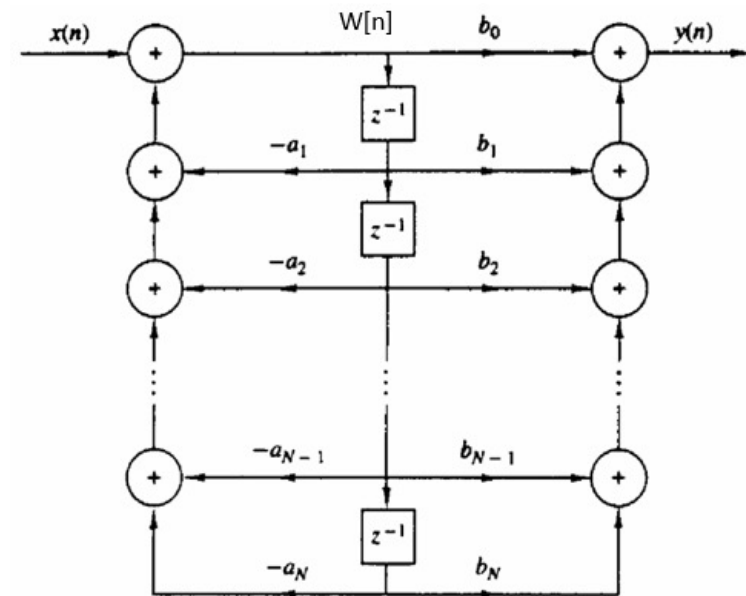
$$y[n] = \sum_{k=0}^M b_k w[n - k] \quad \text{.....3}$$

➤ The direct form-II realization (or structure) is shown in figure below (for **N=M**):

- **Computational Complexity:**

➤ This realization requires:

- Number of additions = M+N
- Number of multiplications = M+N+1
- Number of memory locations (delay elements) is equal to the order of the filter (or, order of the system function or difference equation), hence canonical structure



b. Cascade Form Structures:

- The cascade form realization of an IIR system or filter is obtained by decomposing the system function $H(z)$ into a product of simpler transfer functions as:

$$H(z) = AH_1(z)H_2(z) \dots H_K(z)$$

$$H(z) = A \prod_{k=1}^K H_k(z)$$

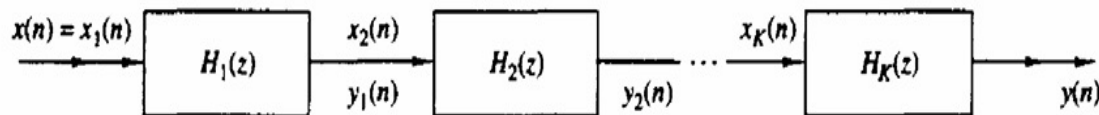
.....1

where,

$A = \text{a constant}$

$k = \text{integer part of } \frac{N+1}{2}$

and it is assumed that $M \leq N$



c. Parallel Form Structures:

- By using partial fraction expansion, the overall system function $H(z)$ can be expressed as:

$$H(z) = C + H_1(z) + H_2(z) + \cdots + H_K(z) \quad \dots 1$$

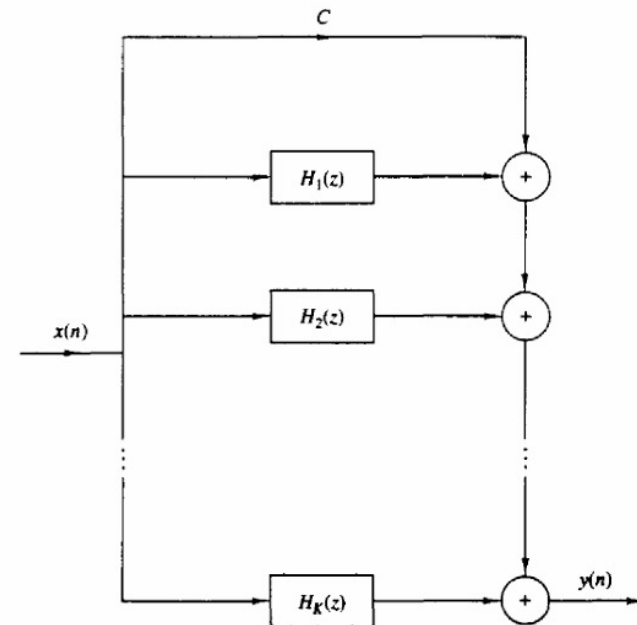
where, $C = \text{a constant}$

$H_1(z), H_2(z), \dots, H_K(z) = \text{second order sub-systems}$

and,
$$H(z) = \frac{Y(z)}{X(z)} = \frac{\sum_{k=0}^M b_k z^{-k}}{1 + \sum_{k=1}^N a_k z^{-k}}, a_0 = 0$$

• Application:

- For **high-speed filtering**. Because the processing of filtering operation is performed parallelly.



d. Lattice and Lattice-ladder Structure:

- Lattice filters are used in **digital speech processing** and the implementing of **adaptive filters**.

a. Lattice Structure:

- An all-pole system with system function

$$H(z) = \frac{1}{1 + \sum_{k=1}^N a_N(k)z^{-k}} = \frac{1}{A_N(z)} \quad \text{.....1}$$

and the difference equation is

$$y[n] = -\sum_{k=1}^N a_N(k) y[n-k] + x[n] \quad \text{.....2}$$

Or,

$$x[n] = y[n] + \sum_{k=1}^N a_N(k) y[n-k] \quad \text{.....3}$$

- **For $N = 1$:**

- We have,
$$\begin{aligned} x[n] &= y[n] + a_1(1) y[n-1] & \because N = k = 1 \\ y[n] &= x[n] - a_1(1) y[n-1] \end{aligned} \quad \text{.....4}$$

- We have,
$$\begin{aligned} x[n] &= y[n] + a_1(1) y[n-1] & \because N = k = 1 \\ y[n] &= x[n] - a_1(1) y[n-1] \end{aligned} \quad \text{.....4}$$

- From figure, we have
- $$\begin{aligned} x[n] &= f_1[n] \\ y[n] &= f_0[n] = f_1[n] - k_1 g_0[n-1] \end{aligned}$$
- Or, $y[n] = x[n] - k_1 y[n-1]$ 5

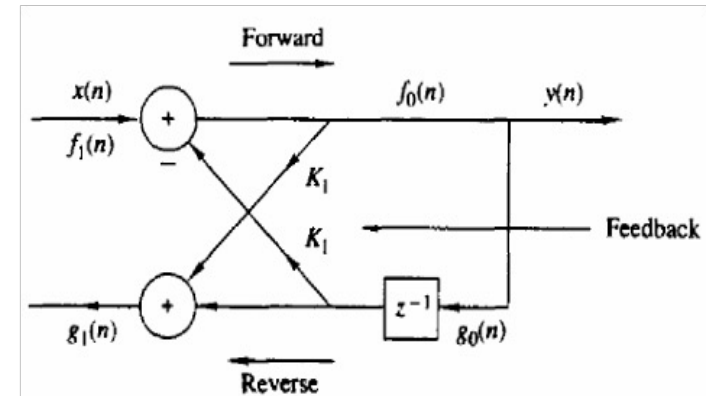


Fig: Single-stage all-pole lattice filter

- Also, $g_1[n] = k_1 y[n] + y[n-1]$
- From equations (4) and (5), we know
- $$k_1 = a_1(1) \quad \text{.....6}$$

• **For $N = 2$:**

- We have,
$$\begin{aligned} x[n] &= y[n] + a_2(1) y[n-1] + a_2(2) y[n-2] & \because N = k = 1 \\ y[n] &= x[n] - a_2(1) y[n-1] - a_2(2) y[n-2] \end{aligned} \quad \text{.....7}$$
- This output can be achieved from two-stage lattice structure as shown in figure below:

- From figure, we have

$$x[n] = f_2[n]$$

$$f_1[n] = f_2[n] - k_2 g_1[n - 1]$$

$$g_2[n] = k_2 f_1[n] + k_2 g_1[n - 1]$$

$$f_0[n] = f_1[n] - k_1 g_0[n - 1]$$

$$g_1[n] = k_1 f_0[n] + g_0[n - 1]$$

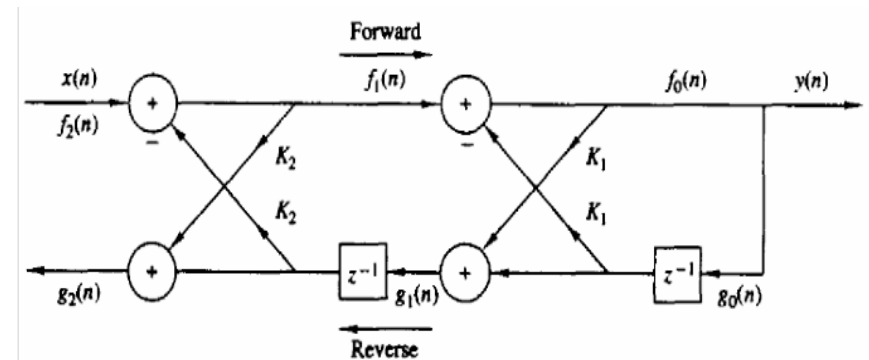


Fig: Two-stage all-pole lattice filter

- Therefore,

$$y[n] = f_0[n] = g_0[n]$$

$$y[n] = f_1[n] - k_1 g_0[n - 1]$$

$$y[n] = \{f_2[n] - k_2 g_1[n - 1]\} - k_1 g_0[n - 1]$$

$$y[n] = f_2[n] - k_2 \{k_1 f_0[n - 1] + g_0[n - 2]\} - k_1 g_0[n - 1]$$

$$y[n] = f_2[n] - k_2 k_1 f_0[n - 1] - k_2 g_0[n - 2] - k_1 g_0[n - 1]$$

$$y[n] = x[n] - k_2 k_1 y[n - 1] - k_2 y[n - 2] - k_1 y[n - 1]$$

- Therefore, $y[n] = x[n] - k_1(1 + k_2)y[n - 1] - k_2 y[n - 2]$

.....8

- In similar manner, we obtain

$$g_2[n] = k_2 y[n] + k_1(1 + k_2)y[n - 1] + y[n - 1] \quad \text{.....9}$$

- From equations (7) and (8), we have

$$a_2(0) = 1$$

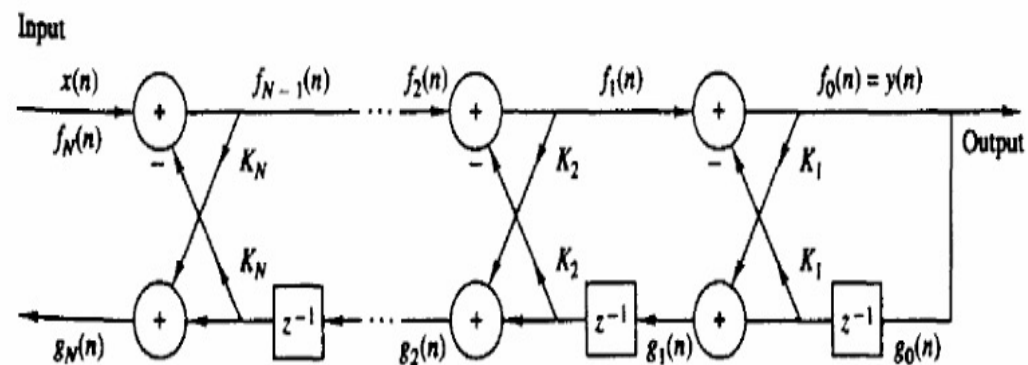
$$a_2(1) = k_1(1 + k_2)$$

and,

$$k_1(1 + k_2)$$

$$a_2(2) = k_2$$

- Therefore, N-stage lattice structure of IIR filter is obtained as shown in figure below:



➤ From figure, we have

$$\begin{aligned} f_N[n] &= x[n] \\ f_{m-1}[n] &= f_m[n] - k_m g_{m-1}[n-1], & m = N, N-1, \dots, 1 \\ g_m[n] &= k_m f_{m-1}[n] + g_{m-1}[n-1], & m = N, N-1, \dots, 1 \\ y[n] &= f_0[n] = g_0[n] \end{aligned}$$

b. Lattice-ladder Structure:

➤ A general IIR filter connecting both poles and zeros can be realized (or implemented) using all-pole lattice as building block. To develop appropriate structure, let us consider an IIR system with system function:

$$H(z) = \frac{\sum_{k=0}^M c_M(k)z^{-k}}{1 + \sum_{k=1}^N a_N(k)z^{-k}} = \frac{C_M(z)}{A_N(z)} \quad \dots 1$$

where, $N \geq M$

- The lattice structure for equation (1) is constructed first by realizing all-pole lattice coefficients k_m , where, $1 \leq m \leq N$ for the denominator $A_N(z)$, and then adding the ladder part by taking the output as a weighted linear combination of $g_m[m]$.
- The result is the pole-zero IIR (lattice-ladder) structure.
- The lattice-ladder structure for $M = N$ is shown in figure below:

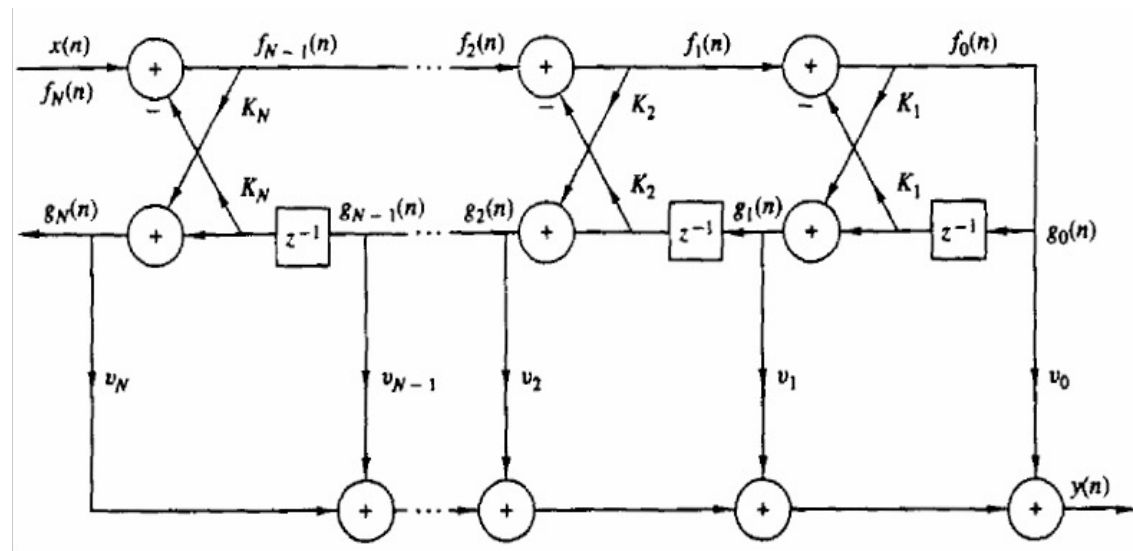


Fig: Lattice-ladder structure for the pole-zero system

➤ The output is given by

$$y(n) = \sum_{m=0}^M v_m g_m(n) \quad \text{.....2}$$

where, v_m = ladder coefficients and obtained by the equation

$$v_m = c_m - \sum_{i=1+m}^M v_i a_i(i-m), \quad m = M, M-1, \dots, 0 \quad \text{.....3}$$

- **Conversion from Lattice Structure to Direct-form Structure:**

➤ In general,

$$\begin{aligned} a_m(0) &= 1 \\ a_m(m) &= k_m \\ a_m(k) &= a_{m-1}(k) + a_m(m) a_{m-1}(m-k) \end{aligned}$$

- **Conversion of Direct Form Coefficients to Lattice Coefficients :**

➤ In general,

$$a_m(0) = 1$$

$$k_m = a_m(m)$$

$$a_{m-1}(k) = \frac{a_m(k) - a_m(m)a_m(m-k)}{1 - \{a_m(m)\}^2}$$

❑ Quantization Effect (Truncation and Rounding):