

	<p>Chapter 7 Digital filters</p>

1

	<h2>Digital Filtering</h2>
	<ul style="list-style-type: none">■ Introduction■ Frequency response■ Fir Filters■ Fir design

2

Introduction

- Digital filters are commonplace and an essential elements of everyday electronics such as digital radios, MP3 players, cellphones, and stereo receivers.

3

Type of filters

- There are two types of filters:

1. Analogue filters.
2. Digital filters.

Analogue filters use discrete components such as resistors, capacitors and inductors.

4

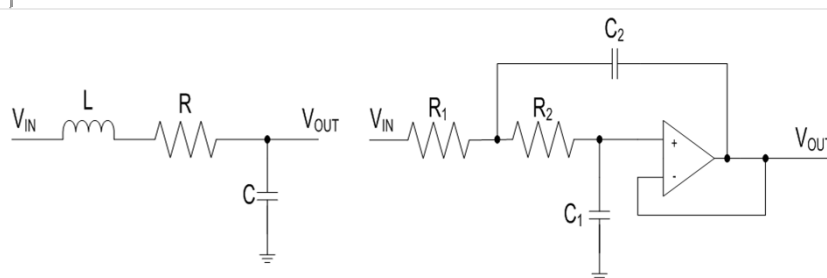
Analogue filters

There are also two types of analogue filters:

1. Passive filters: analogue filters that use discrete components such as resistors, capacitors and inductors.
2. Active filters: analogue filters that uses resistors, capacitors and operational amplifiers, (op-amp).

5

Analogue filters, examples

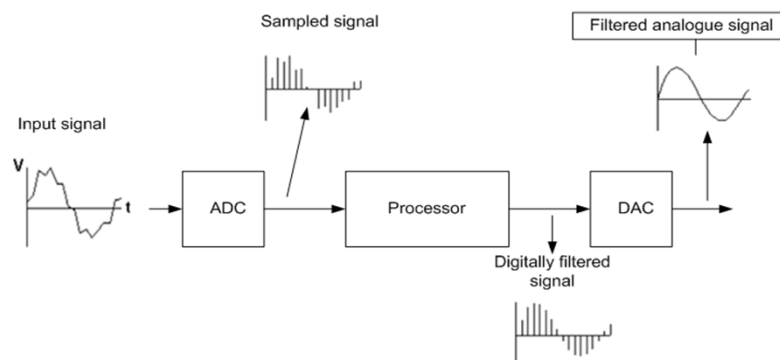


Second-Order Passive Low-Pass and Second-Order Active Low-Pass

6

Digital Filters

Digital filters use digital processors to perform numerical operations on the sampled waveform.



Types of digital filters

There are two main types of digital filters, recursive and non-recursive:

1. FIR filters (Finite Impulse Response Filters): These are mainly used for their linear phase response. (non-recursive).
2. IIR Filters (Infinite Impulse Response Filters): These are mainly used for the low number of "taps" but have non linear phase response. (recursive).

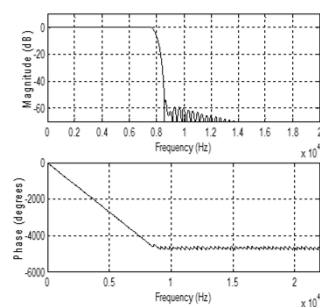
Advantage of an FIR filter

- Amongst all the obvious advantages that digital filters offer, the FIR filter can guarantee linear phase characteristics.
- Neither analogue or IIR filters can achieve this.
- There are many commercially available software packages for filter design. However, without basic theoretical knowledge of digital filters, it will be difficult to use them.

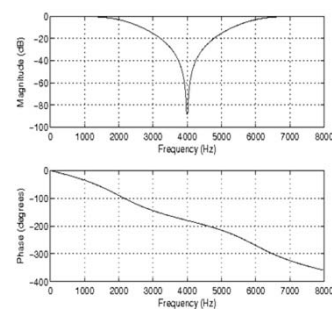
9

Type of digital filters

- Phase response of an FIR filter:



- Phase response of an IIR filter:



10

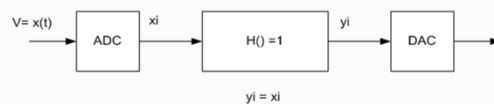
Advantages and disadvantages in using Digital Filters

- Can have linear phase response.
- Digital filters can be reprogrammable.
- They are low cost.
- They consume low power.
- They have small size (implemented in software)
- Easy to design.
- Have a very precise transfer function.
- Unlike the analogue filters, digital Filters don't:
 - Drift
 - Age
 - Change with temperature

11

How non-recursive digital filters work

Eg 1: Filter with a unite gain



$x_i : x_0, x_1, x_2, x_3, x_4, \dots, x_N$

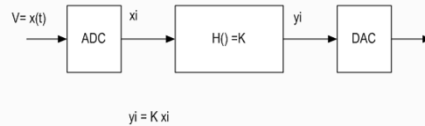
T: the sampling period

These digital filter has no effect on the input signal

$y_0 = x_0,$
 $y_1 = x_1,$
 $y_2 = x_2,$
 $y_3 = x_3,$
 $y_4 = x_4,$
 \dots
 \dots
 $y_N = x_{N,2}$

How digital filters work

Filter with a gain, K



$x_i : x_0, x_1, x_2, x_3, x_4, \dots, x_N$

The digital filter act as an amplifier

$$y_0 = K x_0,$$

$$y_1 = K x_1,$$

$$y_2 = K x_2,$$

$$y_3 = K x_3,$$

$$y_4 = K x_4,$$

...

...

$$y_N = K x_N$$

13

How digital filters work

Filter with a gain, K

K value	Effect of the digital filter
0 or 1	Switch
-1	Inverter
$0 < K < 1$	Attenuator
$-1 < K < 0$	Inverting attenuator
$K > 1$	Amplificator
$K < -1$	Inverting amplificador

$$y_0 = K x_0,$$

$$y_1 = K x_1,$$

$$y_2 = K x_2,$$

$$y_3 = K x_3,$$

$$y_4 = K x_4,$$

...

...

$$y_N = K x_N$$

14

How digital filters work

Eg 2: Two-term filter

$$y_i = \frac{x_i + x_{i-1}}{2}$$

$$y_0 = \frac{x_0 + x_{-1}}{2} \quad y_3 = \frac{x_3 + x_2}{2}$$

$$y_1 = \frac{x_1 + x_0}{2} \quad y_4 = \frac{x_4 + x_3}{2}$$

$$y_2 = \frac{x_2 + x_1}{2} \quad y_5 = \frac{x_5 + x_4}{2}$$

15

How digital filters work

Eg 3: Three-term filter

$$y_i = \frac{x_i + x_{i-1} + x_{i-2}}{3}$$

$$y_0 = \frac{x_0 + x_{-1} + x_{-2}}{3} \quad y_3 = \frac{x_3 + x_2 + x_1}{3}$$

$$y_1 = \frac{x_1 + x_0 + x_{-1}}{3} \quad y_4 = \frac{x_4 + x_3 + x_2}{3}$$

$$y_2 = \frac{x_2 + x_1 + x_0}{3} \quad y_5 = \frac{x_5 + x_4 + x_3}{3}$$

16

How digital filters work

- N-term filter

$$y_i = \frac{x_i + x_{i-1} + \dots + x_{i-(N-1)}}{N}$$

17

Digital filter coefficients

- General form of a digital filter:

$$y_i = a_0x_i + a_1x_{i-1} + a_2x_{i-2} + \dots + a_Nx_{i-(N-1)}$$

Eg:

$$y_i = 3x_i - x_{i-1} + 2x_{i-6} + 5x_{i-(106)}$$

What is the order of this filter? 106!

18

Order of digital filters

The order of digital filter is the number of previous inputs that need to be stored in memory

Zero Order:

$$y_i = a_0 x_i$$

First order

$$y_i = a_0 x_i + a_1 x_{i-1}$$

Second order

$$y_i = a_0 x_i + a_1 x_{i-1} + a_2 x_{i-2}$$

N order

$$y_i = a_0 x_i + a_1 x_{i-1} + a_2 x_{i-2} + \dots + a_N x_{i-(N-1)}$$

19

How recursive digital filters work?

General form

$$y_i = b_0 x_i + b_1 x_{i-1} + \dots + b_N x_{i-N} + a_0 y_i + a_1 y_{i-1} + \dots + a_N y_{i-N}$$

Eg:

$$y_i = 2x_i - x_{i-1} + 3x_{i-2} + y_{i-1}$$

20

What is digital filter design?

- The design of a digital filter is carried out in three steps:
 1. **Specifications:** they are determined by the applications
 2. **Approximations:** once the specification are defined, we use various concepts and mathematics that we studied so far to come up with a filter description that approximates the given set of specifications.
 3. **Implementation:** The product of the above step is a filter description in the form of either a difference equation, or a system function $H(z)$, or an impulse response $h(n)$. From this description we implement the filter in hardware or through software on a computer.

21

Specifications

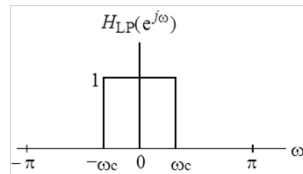
- Specifications are required in the frequency-domain in terms of the desired **magnitude** and **phase** response of the filter.
- Generally a **linear phase response** in the passband is desirable.
 - In the case of **FIR** filters, It is possible to have exact linear phase.
 - In the case of **IIR** filters, a linear phase in the passband is not achievable.

22

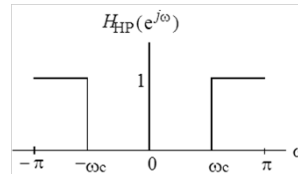
Filter Types

- Frequency responses for four basic types of **ideal filters**

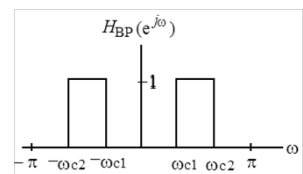
LOW PASS



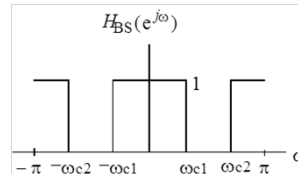
HIGH PASS



BAND PASS

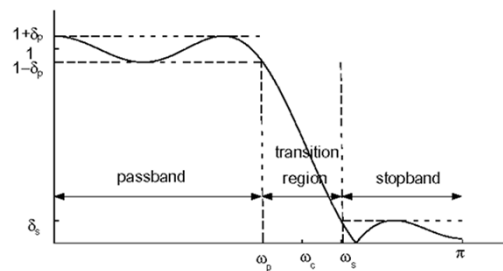


BAND STOP



23

Filter Specification



δ_p passband ripple

δ_s stopband ripple

passband: $|\omega| < \omega_p$ in passband: $|H(e^{j\omega}) - 1| \leq \delta_p$

stopband: $|\omega| > \omega_s$ in stopband: $|H(e^{j\omega})| \leq \delta_s$

transition band: $\omega_p < |\omega| < \omega_s$

24

Properties of an FIR Filter

- Filter coefficients:

$$y[n] = \sum_{k=0}^{N-1} b_k \cdot x[n-k]$$

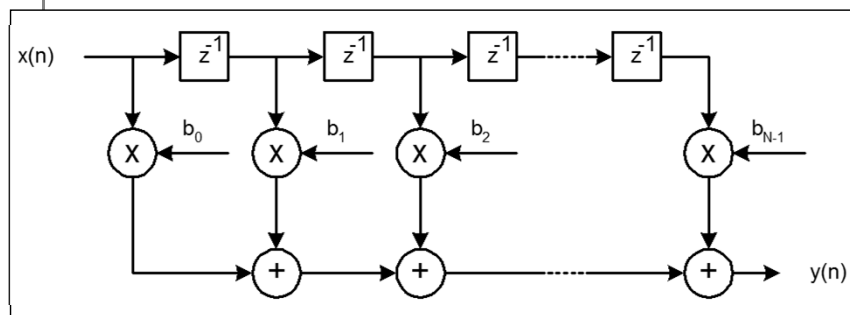
$x[n]$ represents the filter input,
 b_k represents the filter coefficients,
 $y[n]$ represents the filter output,
 N is the number of filter coefficients
 (order of the filter).

25

FIR Filter Structure

$$y[n] = \sum_{k=0}^{N-1} b_k \cdot x[n-k]$$

FIR equation



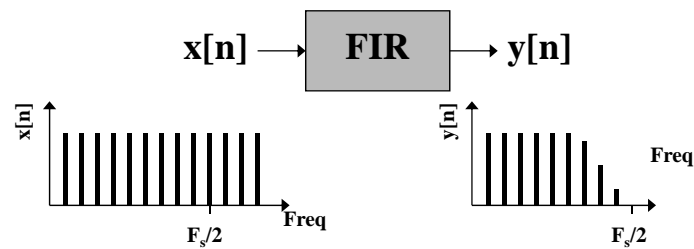
FIR Filter structure

26

Frequency Response of an FIR Filter

- Frequency response:

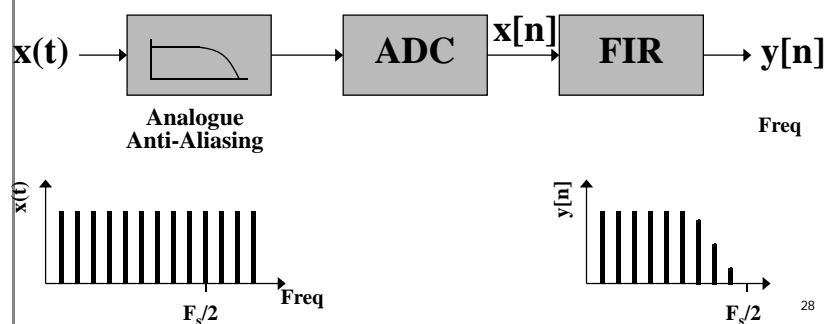
$$H(e^{j\omega+2k\pi}) = H(e^{j\omega})$$



27

Frequency Response of an FIR Filter

- Solution: Use an anti-aliasing filter.



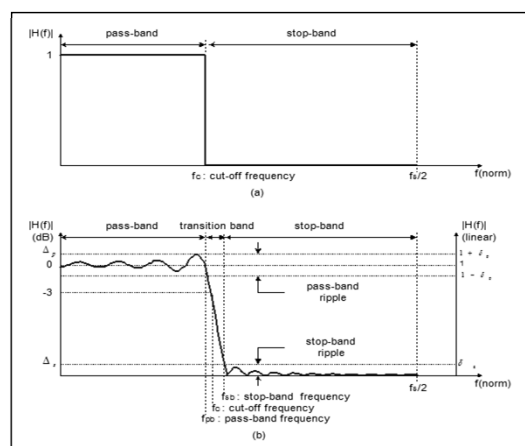
28

Design Procedure

- To fully design and implement a filter five steps are required:
 - (1) Filter specification.
 - (2) Coefficient calculation.
 - (3) Structure selection.
 - (4) Simulation (optional).
 - (5) Implementation.

29

Filter Specification - Step 1



30

Coefficient Calculation - Step 2

- There are several different methods available, the most popular are:
 - Window method.
 - Frequency sampling.
 - Parks-McClellan.
- We will just consider the window method.

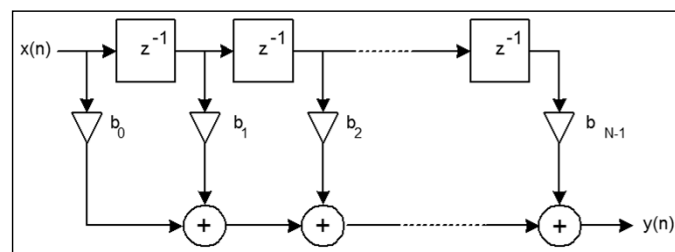
31

Realisation Structure Selection - Step 3

$$H(z) = \sum_{k=0}^{N-1} b_k z^{-k}$$

$$Y(z) = H(z) \cdot X(z)$$

$$y(n) = b_0 x(n) + b_1 x(n-1) + \dots + b_{N-1} x(n-N+1)$$



32

Implementation - Step 5

b_0
b_1
b_2
b_3

x_0
x_1
x_2
x_3

$$y_0 = b_0 * x_0 + b_1 * x_1 + b_2 * x_2 + b_3 * x_3$$

33

Implementation - Step 5

b_0
b_1
b_2
b_3

x_4
x_1
x_2
x_3

$$\begin{aligned} y_0 &= b_0 * x_0 + b_1 * x_1 + b_2 * x_2 + b_3 * x_3 \\ y_1 &= b_0 * x_4 + b_1 * x_1 + b_2 * x_2 + b_3 * x_3 \end{aligned}$$

34

Implementation - Step 5

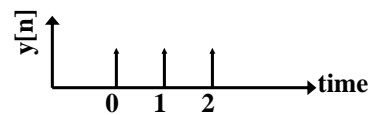
b_0
b_1
b_2
b_3

x_4
x_5
x_2
x_3

$$y_0 = b_0 \cdot x_0 + b_1 \cdot x_1 + b_2 \cdot x_2 + b_3 \cdot x_3$$

$$y_1 = b_0 \cdot x_4 + b_1 \cdot x_1 + b_2 \cdot x_2 + b_3 \cdot x_3$$

$$y_2 = b_0 \cdot x_4 + b_1 \cdot x_5 + b_2 \cdot x_2 + b_3 \cdot x_3$$



35

Digital filters

- End -

36