

# FIR Filter Design summary

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# FIR Filter Design

- Window Method
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# FIR Filter Design

In signal processing, a ***Finite Impulse Response (FIR)*** filter is a filter whose impulse response (or response to any finite length input) is of finite duration, because it settles to zero in finite time.

This is in contrast to ***Infinite Impulse Response (IIR)*** filters, which may have internal feedback and may continue to respond indefinitely (usually decaying).

# FIR Filter Design

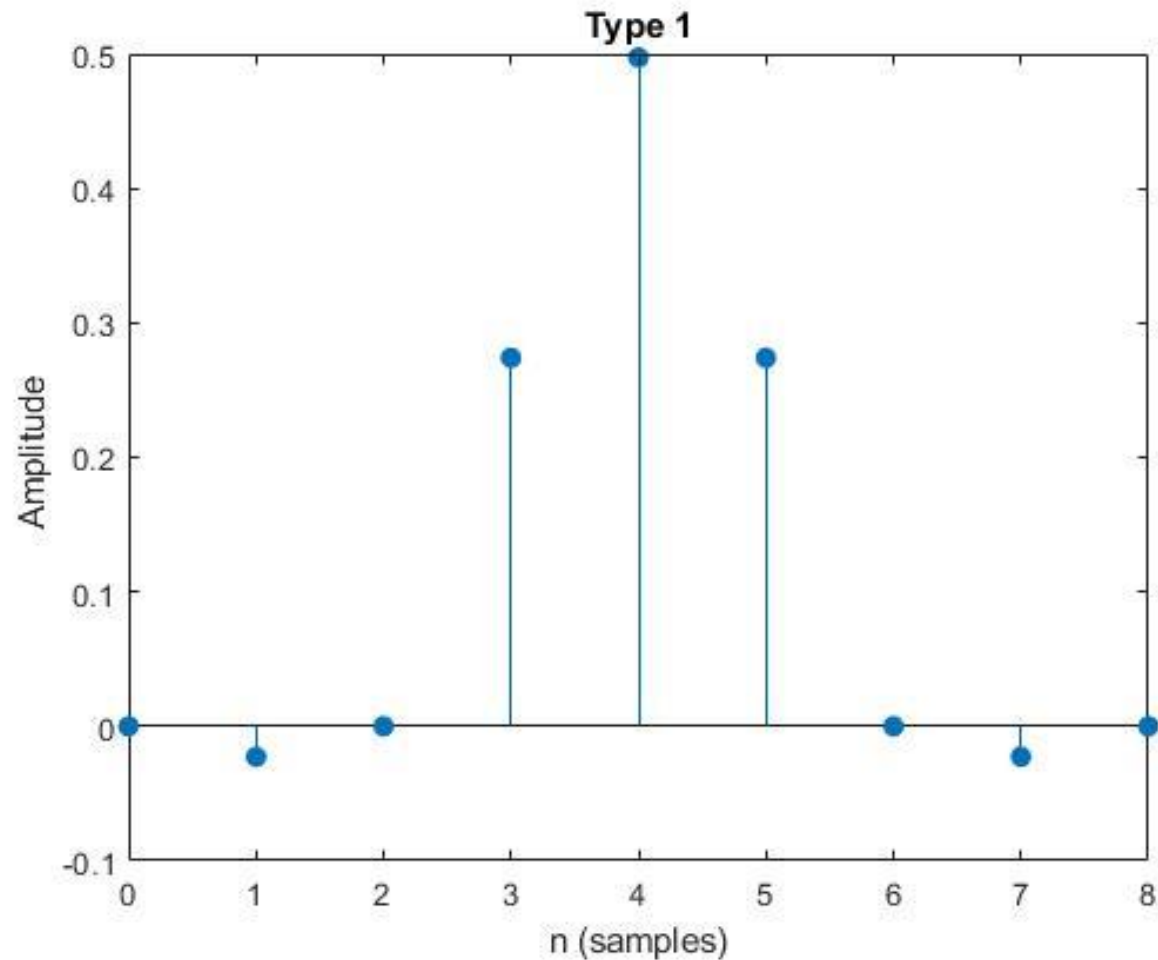
A **FIR filter** of order  $N$ , is a filter of the form:

$$H(z) = \sum_{n=0}^{N-1} h(n) z^{-1} z^{-n}.$$

Its **Frequency Response** is given by:

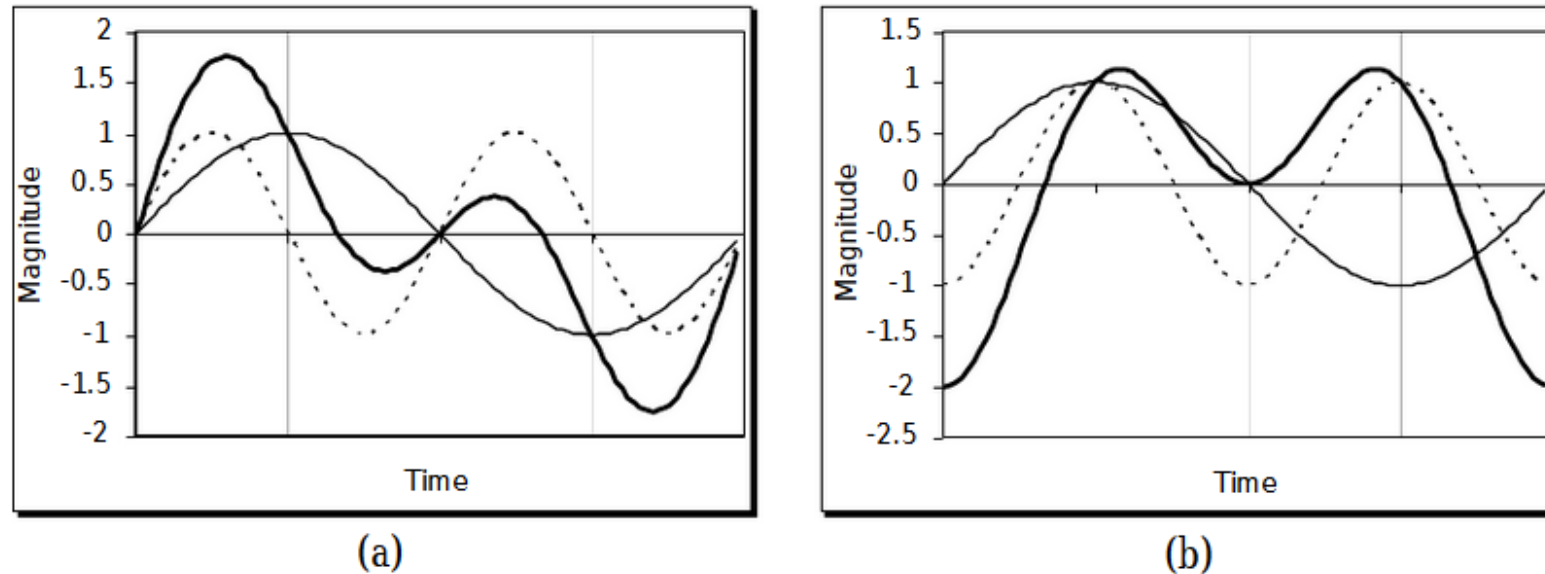
$$H(e^{iW}) = \sum_{n=0}^{N-1} h(n) e^{-iWn}.$$

# FIR - Linear Phase





# FIR - Linear Phase



**Figure 8.5.** Effects of linear and non-linear phase on signals. In both parts (a) and (b), the bold trace denotes the sum of a 1 Hz and a 2 Hz sine wave. However, in part (b), the 2 Hz sine wave has also been phase delayed by  $\pi/2$ , ie. 90°.

# Window Method

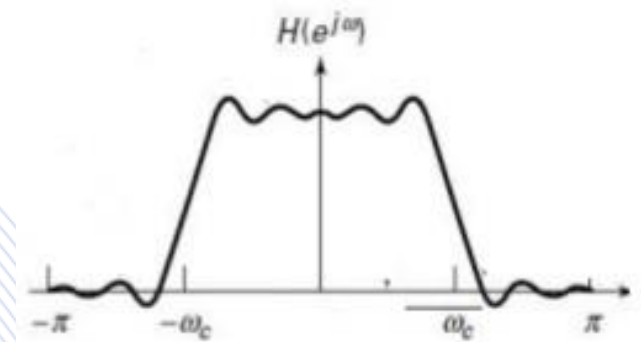
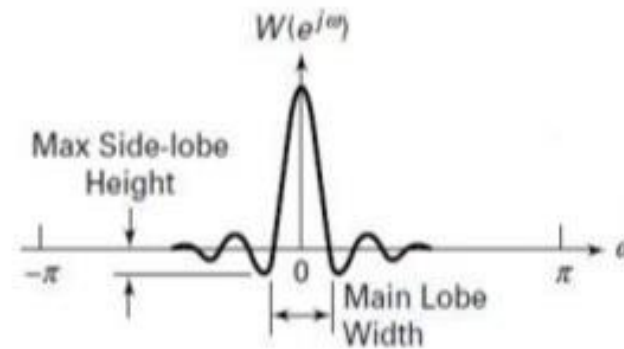
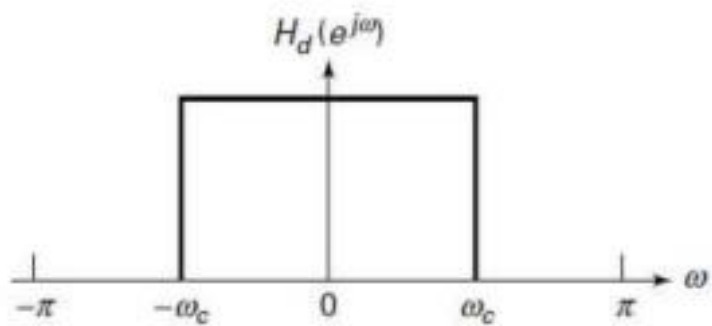
If the *Impulse Response*  $h_d(n)$ ,  $-\infty < n < \infty$  needed is known, a **FIR** filter of length  $N$  can be designed as:

$$h(n) = h_d(n)w(n)$$

The following function  $w(n)$  is called “**Window**”:

$$w(n) = \begin{cases} \neq 0, & 0 \leq n \leq N - 1 \\ 0, & \text{elsewhere} \end{cases}$$

# Window Method





# Types of Windows

- **Rectangular Window:**

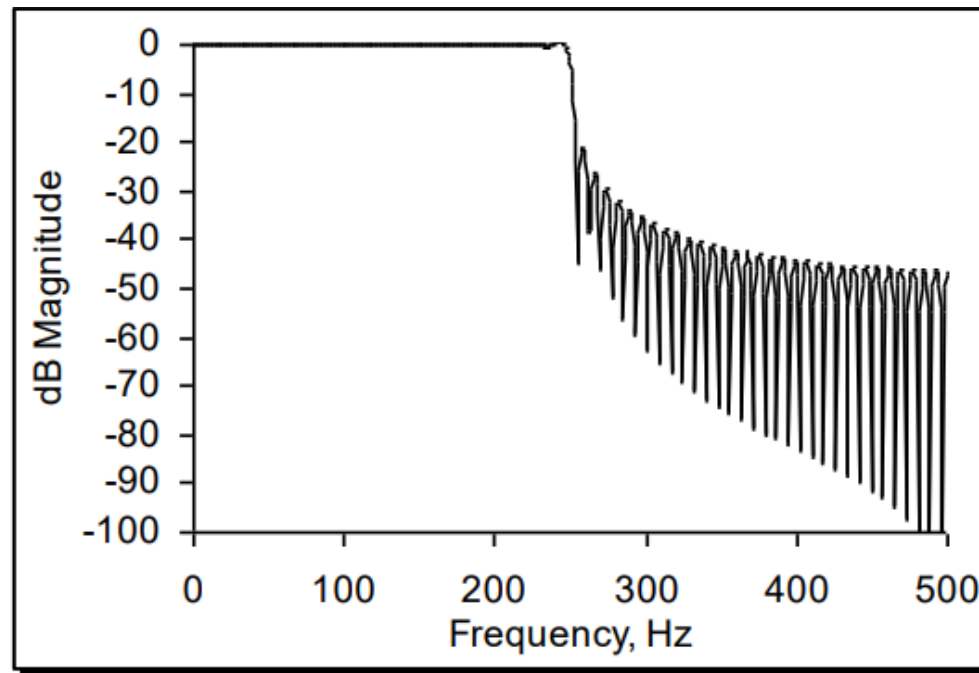
$$w(n) = 1, 0 \leq n \leq N - 1$$

- **Bartlett Window:**

$$w(n) = \begin{cases} \frac{2n}{N-1}, & 0 \leq n \leq \frac{N-1}{2} \\ 2 - \frac{2n}{N-1}, & \frac{N-1}{2} \leq n \leq N-1 \end{cases}$$

# Window Method

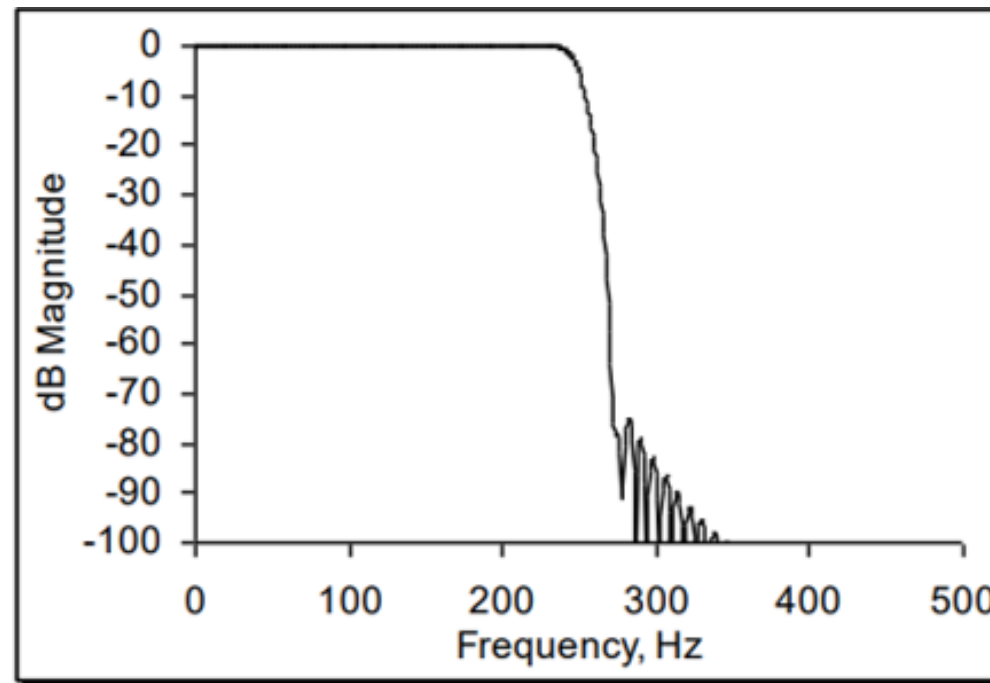
Magnitude *Frequency Responses*, shown in dBs, of a *low-pass FIR* filter with a cut-off of 250 Hz and a sample rate of 1 kHz using: (a) a ***Rectangular***



(a)

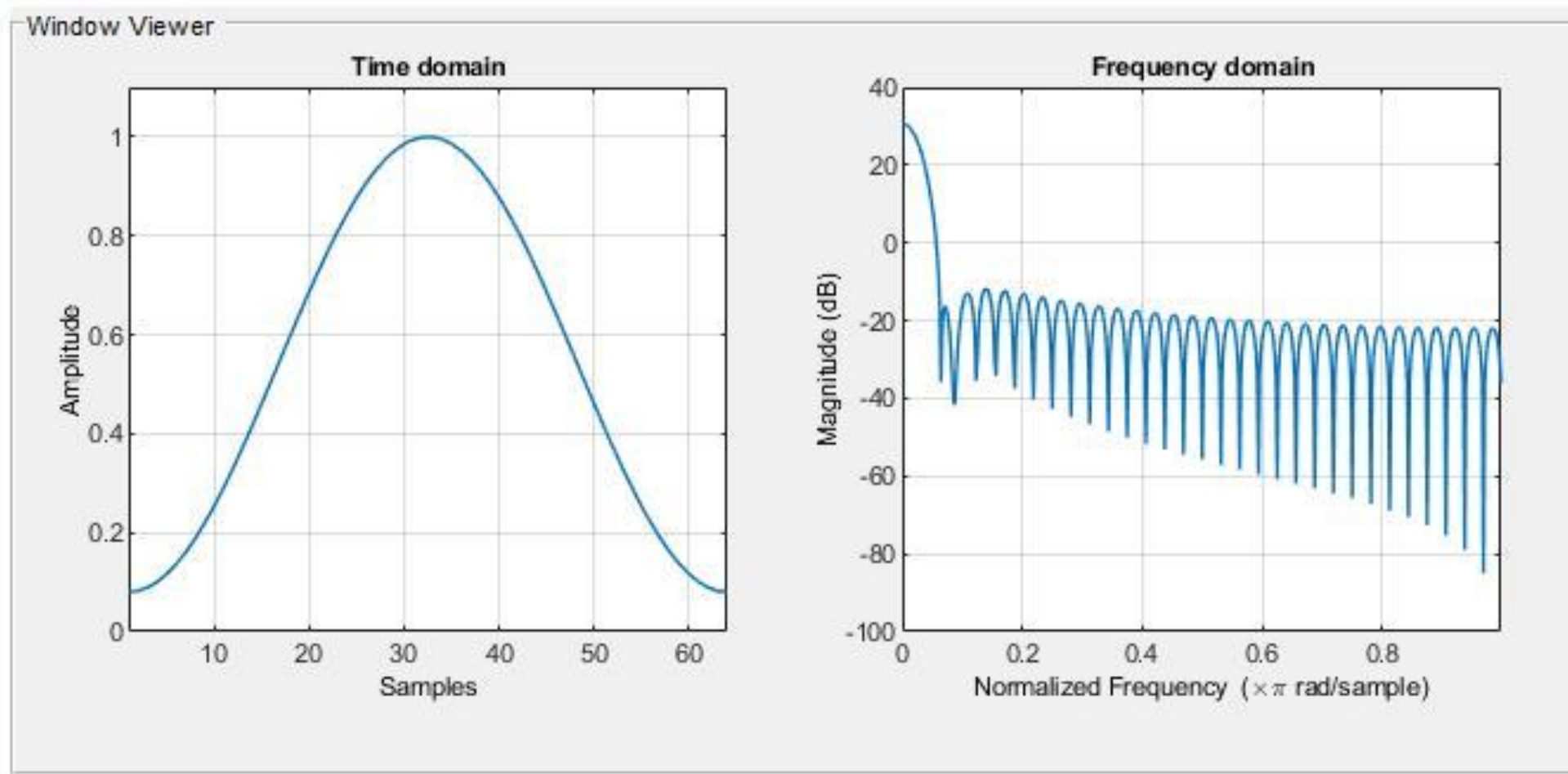
# Window Method

Magnitude *Frequency Responses*, shown in dBs, of a *low-pass FIR* filter with a cut-off of 250 Hz and a sample rate of 1 kHz using: and (d) a ***Blackman window***.



(d)

# Window Method



# Optimisation Method

An effective design method where the coefficients  $h(n)$ ,  $0 \leq n \leq N - 1$  in a way to minimize other criteria, such as  $L_2$  or the  $L_p$  of the *approximation error*.

$$E_2 = \frac{1}{2\pi} \int_{-\pi}^{\pi} |H_d(e^{iW}) - H(e^{iW})|^2 dW$$

$$E_p = \frac{1}{2\pi} \int_{-\pi}^{\pi} |H_d(e^{iW}) - H(e^{iW})|^p dW$$



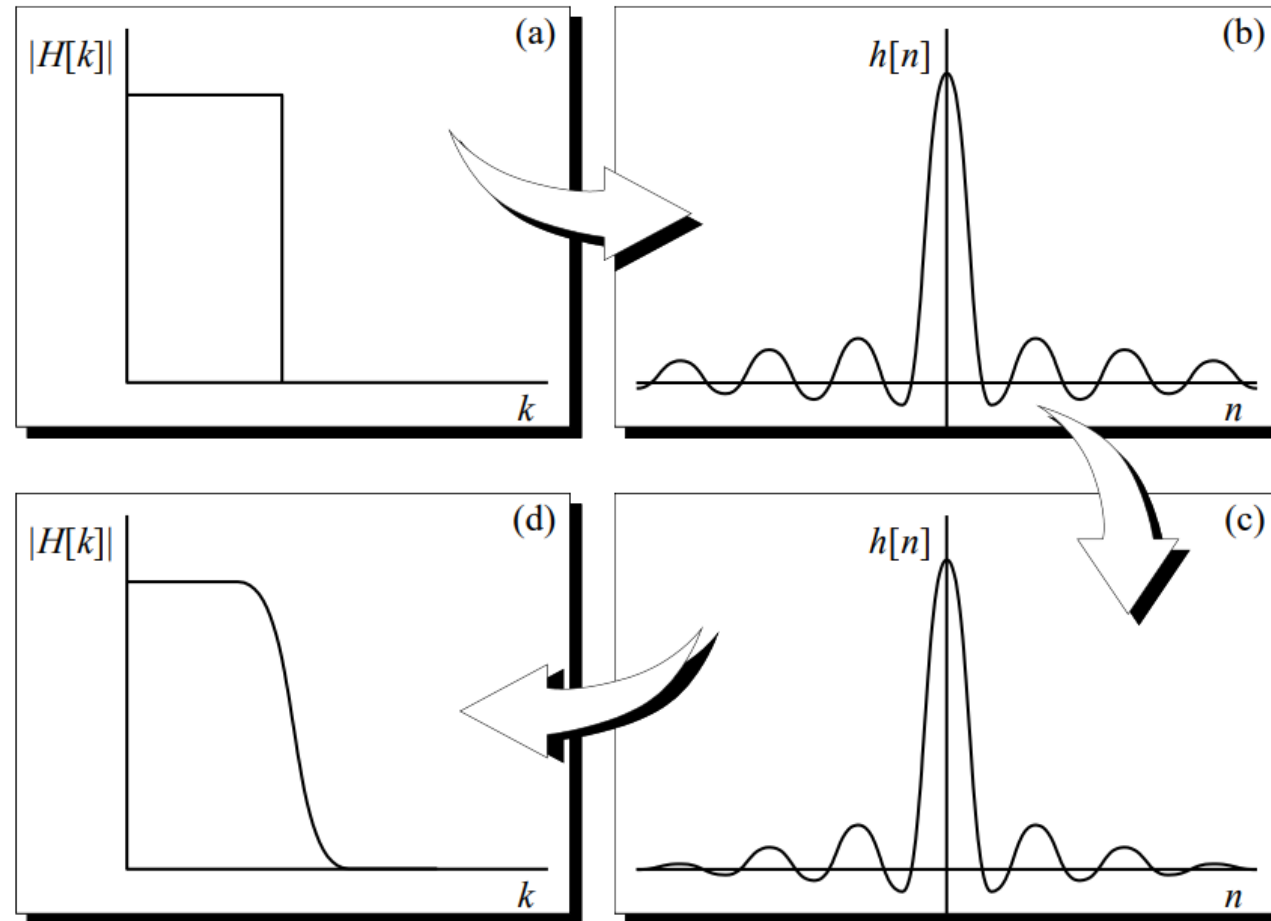
# Frequency Sampling Method

A very simple design method is by choosing:

$$\tilde{H}(k) = H_d\left(e^{i\frac{2\pi k}{N}}\right), k = 0, 1, \dots, N - 1$$

We can optimize this by choosing the values of  $\tilde{H}(k)$  so that the filter response is optimized in the *passband* and the *stopband*.

# Frequency Sampling Method



**Figure 8.6.** Design steps using the frequency sampling method.

# Equiripple FIR Filter Design

Design of *Linear Phase* **FIR** Filters

$$H(e^{iW}) = \sum_{n=-M}^M h(n)e^{-iWn}, h(n) = h(-n)$$

In this case the *Frequency Response* is *real*:

$$H(e^{iW}) = h(0) + \sum_{n=1}^M 2h(n) \cos(Wn)$$

# Equiripple FIR Filter Design

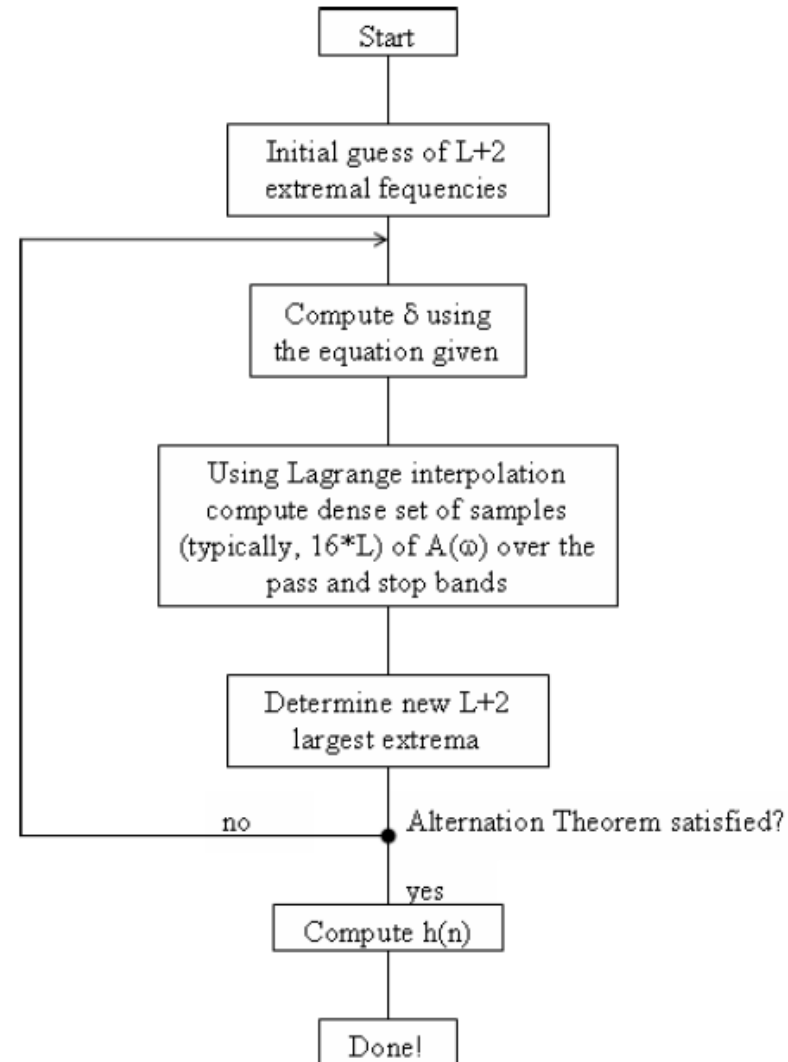
And the unknown constants are  $M+1$ .

Usually,  $M, \delta_1, \delta_2$  are *constants* and  $W_p, W_s$  are *variables*, or the opposite.

Given the  $M, W_p, W_s$ , we're trying to minimize the error:

$$E_{max} = \max |E(W)|, 0 \leq W \leq W_p, W_s \leq W \leq \pi$$

# REMEZ Algorithm





# FIR vs IIR

- The main advantage of **IIR** Filters is that they demand a less amount of operations than the **FIR** Filters.
- The main advantage of **FIR** Filters is their *Phase Linearity*.

# Bibliography

[GAY2010] Patrick Gaydecki. “Chapter 8: The Design and Implementation of Finite Impulse Response Filters” 2010.

[EQUI] “Equi-ripple FIR Filter Design”. PDF file.

[SUH2014] Suhaib Ahmed, Mudasar Bashir, Ashish Suri 2014: Low Pass FIR Filter Design and Analysis Using Hamming, Blackman and Kaiser Windows. PDF file.

# Bibliography

[FIRF] “FIR Filter Design (Windowing Technique)”. PDF file.

[VAI] P.P.Vaidyanathan, T.Q.Nguyen. “A Simple Proof of the Alternation Theorem”. PDF file.

[WAN2020] Lars Wanhammar, Tapio Saramäki, 2020: Digital Filters Using MATLAB.

[RAH2015] Rahul Kumar Sahu, 2015: Comparative Designing of Optimal FIR Filter using Parks-McClellan & Genetic Algorithm. PDF file.

# Bibliography



[OPP2013] A. Oppenheim, A. Willsky, Signals and Systems, Pearson New International, 2013.

[MIT1997] S. K. Mitra, Digital Signal Processing, McGraw-Hill, 1997.

[OPP1999] A.V. Oppenheim, Discrete-time signal processing, Pearson Education India, 1999.

[HAY2007] S. Haykin, B. Van Veen, Signals and systems, John Wiley, 2007.

[LAT2005] B. P. Lathi, Linear Systems and Signals, Oxford University Press, 2005.

[HWE2013] H. Hwei. Schaum's Outline of Signals and Systems, McGraw-Hill, 2013.

[MCC2003] J. McClellan, R. W. Schafer, and M. A. Yoder, Signal Processing, Pearson Education Prentice Hall, 2003.



# Bibliography



[PHI2008] C. L. Phillips, J. M. Parr, and E. A. Riskin, Signals, Systems, and Transforms, Pearson Education, 2008.

[PRO2007] J.G. Proakis, D.G. Manolakis, Digital signal processing. PHI Publication, 2007.

[DUT2009] T. Dutoit and F. Marques, Applied Signal Processing. A MATLAB-Based Proof of Concept. New York, N.Y.: Springer, 2009



# Bibliography

- [PIT2000] I. Pitas, “Digital Image Processing Algorithms and Applications”, J. Wiley, 2000.
- [PIT2021] I. Pitas, “Computer vision”, Createspace/Amazon, in press.
- [PIT2017] I. Pitas, “Digital video processing and analysis” , China Machine Press, 2017 (in Chinese).
- [PIT2013] I. Pitas, “Digital Video and Television” , Createspace/Amazon, 2013.
- [NIK2000] N. Nikolaidis and I. Pitas, “3D Image Processing Algorithms”, J. Wiley, 2000.

# Q & A

**Thank you very much for your attention!**

**More material in  
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