# Department of Electronic and Telecommunication Engineering University of Moratuwa, Sri Lanka

EN2570 - Digital Signal Processing



# Design of a Finite Duration Impulse Response Bandpass Digital Filter

(For Prescribed Specifications Using the windowing method in conjunction with the Kaiser window)

**Project Report** 

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#### Abstract

Design procedure of a Finite Duration Impulse Response(FIR) bandpass Digital Filter which satisfies a set of prescribed specifications, is described in this report where windowing method in conjunction with the Kaiser window is used for the designing procedure. Operation of the filter was analyzed with a combination of sine functions. The design was implemented and tested using MATLAB R2018a of the MathWorks Inc. Therefore implementation is not guaranteed to work on the previous version of the software.

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Note:

 $Additionally\ all\ the\ materials\ related\ to\ Task\ can\ also\ be\ found\ at$ 

# 1 Introduction

This report describes the design procedure of an FIR bandpass digital filter.

# 2 Method

### 2.1 Filter Implementation

Filter implementation consists of the steps mentioned below. Subsections of this section of the report describes each one of them for designing an FIR bandpass filter.

- 1. Identifying the prescribed filter specifications
- 2. Derivation of the filter Parameters
- 3. Derivation of the Kaiser Window Parameters

#### 2.1.1 Prescribed Filter specifications

Following table describes the desired specifications of the bandpass filter which need to be implemented. The notation used here is the same as the notation used in the reference material[1] and they will be used throughout the report.

Parameter	Symbol	Value
Maximum passband ripple	$ ilde{A_p}$	$0.09~\mathrm{dB}$
Minimum stopband attenuation	$ ilde{A_a}$	48  dB
Lower passband edge	$\omega_{p1}$	400  rad/s
Upper passband edge	$\omega_{p2}$	800  rad/s
Lower stopband edge	$\omega_{a1}$	250  rad/s
Upper stopband edge	$\omega_{a2}$	900  rad/s
Sampling frequency	$\omega_s$	2600  rad/s

Table 1: Prescribed Filter specifications

Following figure illustrates the aforementioned specifications for an idealized frequency responses of Bandpass filter.  $\delta$  in the figure has the following relationship with peak to peak passband ripple  $A_p$  and the minimum stopband attenuation  $A_a$ .

$$\tilde{A}_p \ge A_p = 20 \log \left( \frac{1+\delta}{1-\delta} \right)$$
 (1)

$$\tilde{A}_a \le A_a = -20\log(\delta) \tag{2}$$

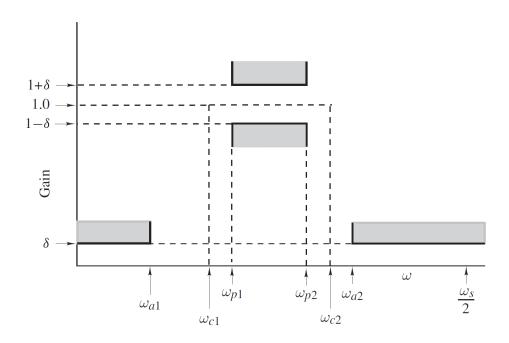


Figure 1: Idealized frequency response of a Bandpass filter[1]

#### 2.1.2 Derivation of filter Parameters

According to the given specifications following parameters are calculated.

Parameter	Symbol	Calculation	Value
Lower transition width	$B_{t1}$	$\omega_{p1} - \omega_{a1}$	150  rad/s
Upper transition width	$B_{t2}$	$\omega_{a2} - \omega_{p2}$	100  rad/s
Critical transition width	$B_t$	$\min(B_{t1}, B_{t2})$	100  rad/s
Lower cutoff frequency	$\omega_{c1}$	$\omega_{p1} - B_t/2$	350  rad/s
Upper cutoff frequency	$\omega_{c2}$	$\omega_{p2} + B_t/2$	850  rad/s
Sampling period	T	$2\pi/\omega_s$	$0.0024~\mathrm{s}$

Table 2: Derivation of filter Parameters

#### 2.1.3 Derivation of the Kaiser Window Parameters

Following equation represents the Kaiser window which will be used to truncate the Infinite duration Impulse Response to obtain the Finite duration Impulse Response for our filter design.

$$w_K(nT) = \begin{cases} \frac{I_0(\beta)}{I_0(\alpha)} & for |n| \le \frac{N-1}{2} \\ 0 & Otherwise \end{cases}$$
 (3)

where  $\alpha$  is an independent parameter and  $I_0(x)$  is the zeroth-order modified Bessel function of the first kind.

$$\beta = \alpha \sqrt{1 - \left(\frac{2n}{N-1}\right)^2} \qquad I_0(x) = 1 + \sum_{k=1}^{\infty} \left[\frac{1}{k!} \left(\frac{x}{2}\right)^k\right]^2$$

- 3 Results
- 4 Discussion
- 5 Conclusion

# Bibliography

 $[1]\,$  Andreas Antoniou.  $Digital\ Signal\ Processing.$  McGraw-Hill Professional, US, 2005.