

Problem Description Matlab Source Code Simulations References

### **Copyright** © 2005-2015.

No part of this manual may be reproduced or transmitted in any form or by any means, electronic or mechanical, including photocopying, recording, or by any information storage or retrieval system, without prior permission from the publisher.

### **Notice of Liability**

Every effort has been made to ensure that this manual contains accurate and current information. However, Yankee Bush Software LLC and the author shall not be held liable for any loss or damage suffered by readers as a result of any information contained herein. This manual is for educational use only and not for commercial use.

## Trademarks & Copyrights

All trademarks and copyrights are acknowledged as belonging to their respective owners.

#### **Licensing Information**

Source code and object code can be freely distributed to students by accredited colleges and universities without requiring a license as long as there is no commercial application involved. In other words, no license is required if you use this manual for educational use only. You must obtain an Object Code Distribution license to embed code into a commercial product. This is a license to put code into a product that is sold with the intent to make a profit. There will be a license free for such situations. You must obtain a Source Code Distribution License to distribute source code. Again there is a fee for such a license. You can download shareware versions but you must obtain commercial licenses from the manufacturers to use full versions of their software and/or hardware.

Des

Ch

De (n

## **Problem Description**

Design a digital FIR lowpass filter to meet the following specifications:

$$\omega_p = 0.22\pi, \qquad R_p = 0.3 \ dB$$
$$\omega_s = 0.32\pi, \qquad A_s = 52 \ dB$$

Choose an appropriate window function from the following table.

| Window<br>name | Transition w<br>Approximate | $\Delta \omega$<br>Exact | Minimum stopband<br>attenuation in dB |
|----------------|-----------------------------|--------------------------|---------------------------------------|
| Rectangular    | $\frac{4\pi}{M}$            | $\frac{1.8\pi}{M}$       | 21                                    |
| Bartlett       | $\frac{8\pi}{M}$            | $\frac{6.1\pi}{M}$       | 25                                    |
| Hanning        | $\frac{8\pi}{M}$            | $\frac{6.2\pi}{M}$       | 44                                    |
| Hamming        | $\frac{8\pi}{M}$            | $\frac{6.6\pi}{M}$       | 53                                    |
| Blackman       | $\frac{12\pi}{M}$           | $\frac{11\pi}{M}$        | 74                                    |

Determine the impulse response and plot the frequency response of the designed filter (magnitude response in dB).

### **Matlab Source Code**

% design of low pass FIR filter

clc;

clear all;

wp=0.22;% normalize frequency

ws=0.32;

Rp=0.3;% in db

As=52; % in db

tr\_width = ws - wp;

```
M = ceil(6.6*pi/tr_width) + 1;
```

delta\_w = 2\*pi/1000;

Rp = -(min(db(1:1:wp/delta\_w+1))) ; % Passband Ripple

As = -round(max(db(ws/delta\_w+1:1:501))); % Min Stopband attenuation

% chosen hamming window

 $f = [0 \ 0.22 \ 0.32 \ 1]; m = [1 \ 1 \ 0 \ 0];$ 

b = fir2(M,f,m);

figure(1)

freqz(b,1);

title('frequency response of low pass filter design using hamming

window');

figure(2)

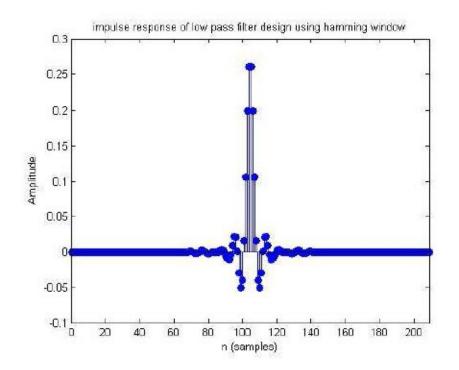
impz(b,1);

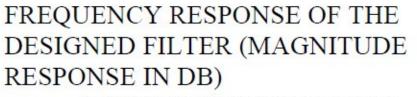
title('impulse response of low pass filter design using hamming

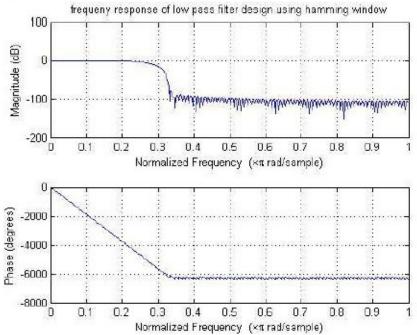
window');

II I R

# IMPULSE RESPONSE OF THE DESIGNED FILTER (MAGNITUDE RESPONSE IN DB)







### References

A. V. Oppenheim, R. W. Schafer with J. R. Buck, "Discrete-Time Signal Processing," Second Ed., Prentice Hall Signal Processing Series, 1999. Pages 1-870

J. G. Proakis and D. G. Manolakis, "Digital Signal Processing Principles, Algorithms, and Applications," Third Ed., Prentice Hall, 1996. Pages 1-1016

S. K. Mitra, "Digital Signal Processing – A Computer-Based Approach," Second Ed., McGraw Hill 2001 (comes with Digital Signal Processing Using MatlabR, Prentice Hall, 1999) Pages 1-354

R. A. Haddad and T. W. Parsons, "Digital Signal Processing – Theory, Applications, and Hardware," Computer Science Press, 1991. Pages 1-636

Joyce Van de Vegte, "Fundamentals of Digital Signal Processing," Prentice Hall 2002. Pages 1-810

N. Dahnoun, "Digital Signal Processing Implementation using the TMS320C6000TM DSP Platform," First Ed., Prentice Hall 2000. Pages 1-234

A. V. Oppenheim, R. W. Schafer, "Digital Signal Processing," First Edition, Prentice Hall, 1975. Pages 1-784

Hykin, "Adaptive Filter Theory", Prentice Hall, Fourth Edition, 2001, Pages 1-936

http://www.mathworks.com/