

Project Status

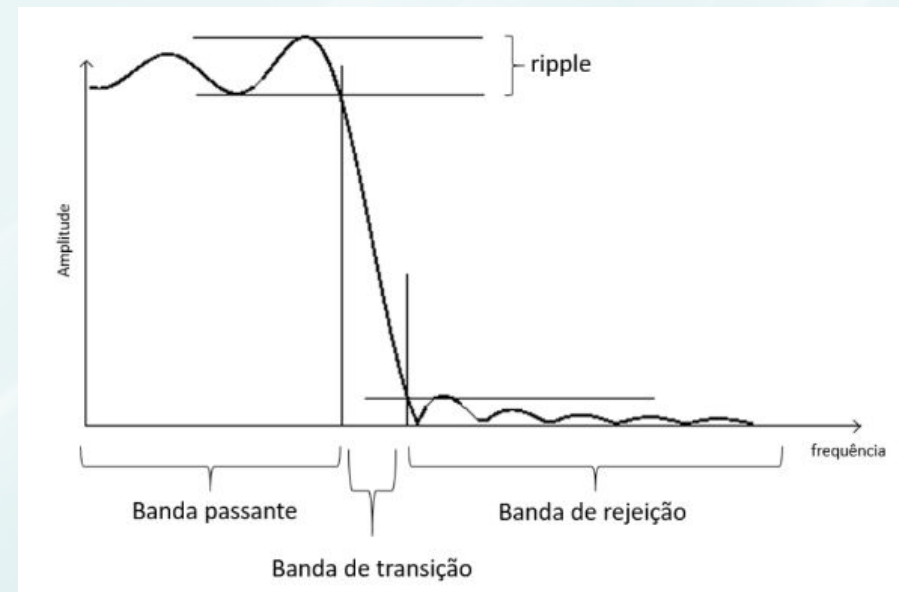
PIEIC

Oscilloscope based in FPGA

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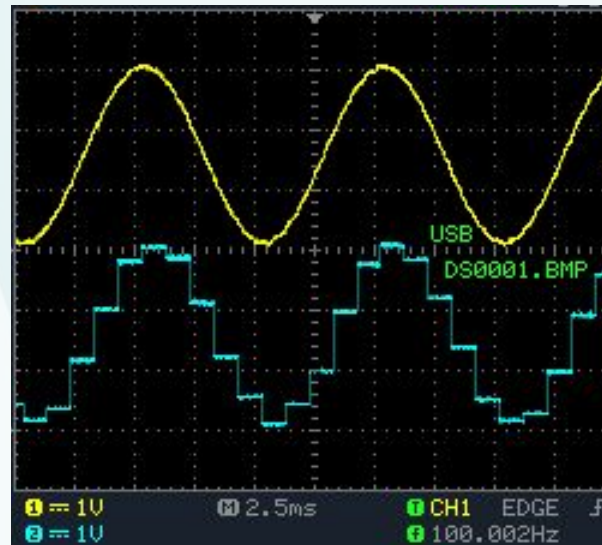
- Digital Filters
 - FIR
 - IIR
- Sampling and Filtering
- Low Pass Filter
 - Filter Calculation
 - Frequency Response
 - Results
- High Pass Filter
 - Filter Calculation
 - Frequency Response
 - Results
- Band Pass Filter
 - Filter Calculation
 - Frequency Response
 - Results

- IIR (Infinite Impulse Response)
 - Similar to analog filters
 - More efficient
 - Unstable
- FIR (Finite Impulse Response)
 - Better response
 - Stable
 - Robust
 - Linear phase



$$\begin{cases} t = nh \rightarrow x^*(t) = x(nh) \\ t \neq nh \rightarrow x^*(t) = 0 \end{cases}$$

$$f_s \geq 2 \cdot f_M$$



Sample Frequency: $f_s = 1 \text{ kHz}$

$$H(z) = \sum_{n=-\infty}^{+\infty} h(n) z^{-n}$$

$$y(n) = h(n) * x(n) = \sum_{k=-\infty}^{+\infty} h(k) x(n - k)$$

$$y[n] = a_1 y[n-1] + \dots + a_N y[n-N] + b_0 x[n] + b_1 x[n-1] + \dots + b_M x[n-M]$$

$a_k = 0$ for FIR filter

Low Pass Filter

```
%----- Low Pass Filter (LPF)
% sampling frequency [Hz]
fsamp = 1000;
% stopband and passband frequencies [Hz]
fcuts = [25 50];
% ripples
devs = [0.01 0.01];

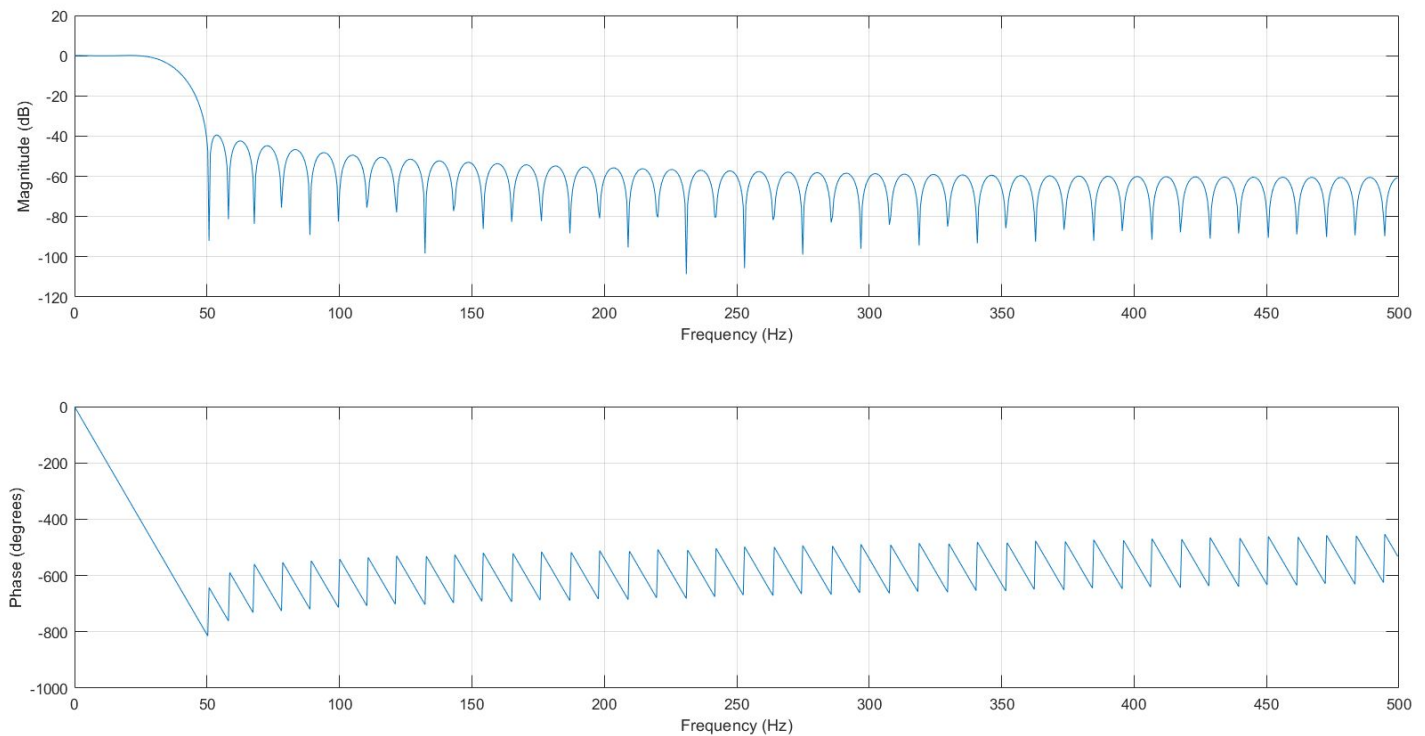
% low pass filter
mags = [1 0];

% get kaiser window
[n,Wn,beta,ftype] = kaiserord(fcuts,mags,devs,fsamp);
% calculate coefficients
hh = fir1(n,Wn,ftype,kaiser(n+1,beta),'noscale');

figure('Name','Low Pass Filter');
freqz(hh,1,1024,fsamp)

fprintf("M [%d]", n);
coefs = regexprep(num2str(hh),'\s+',',')
```

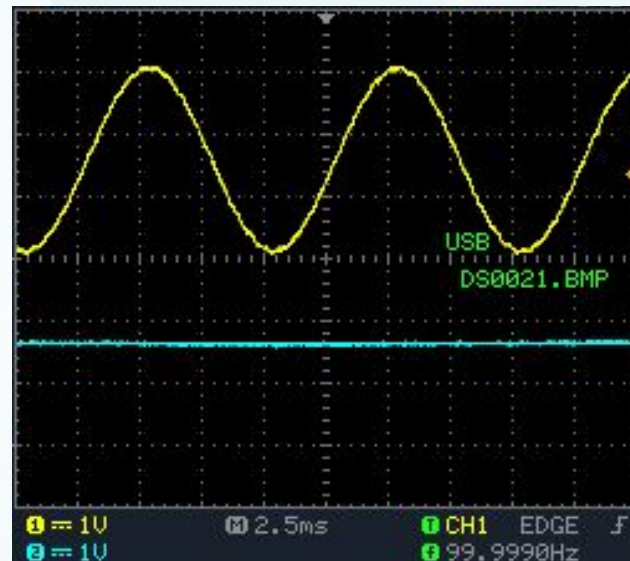
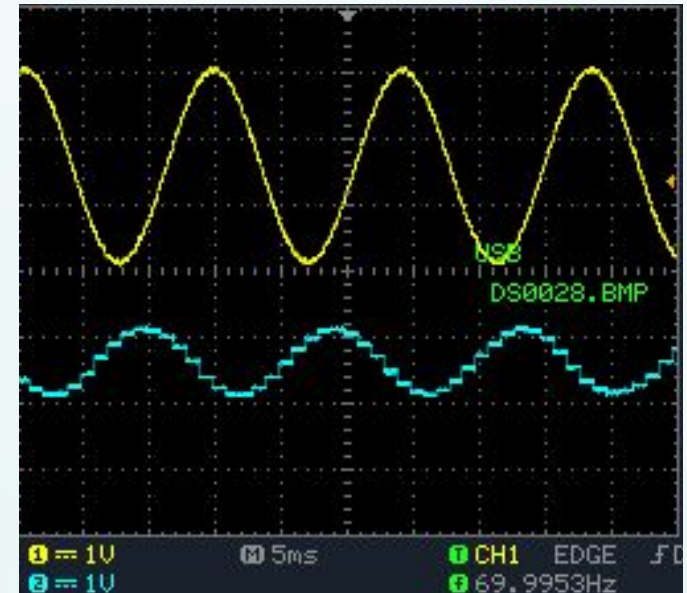
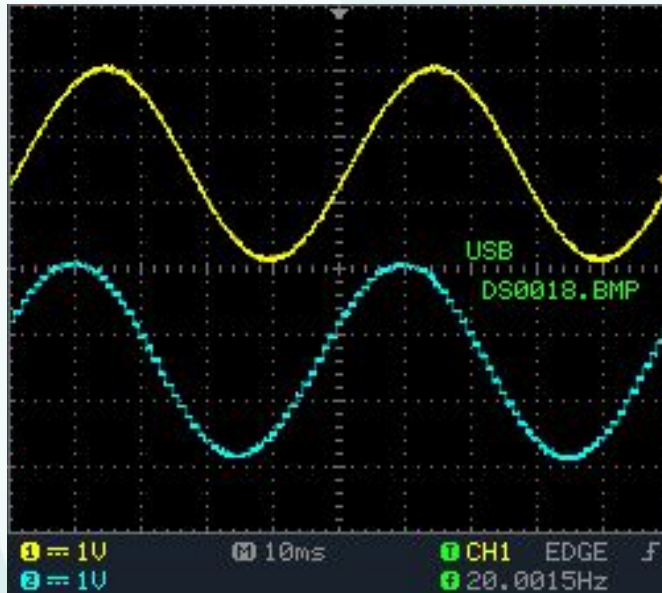
Low Pass Filter



Low Pass Filter



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High Pass Filter

```
%----- High Pass Filter (HPF)
% sampling frequency [Hz]
fsamp = 1000;
% stopband and passband frequencies [Hz]
fcuts = [50 75];
% ripples
devs = [0.01 0.01];

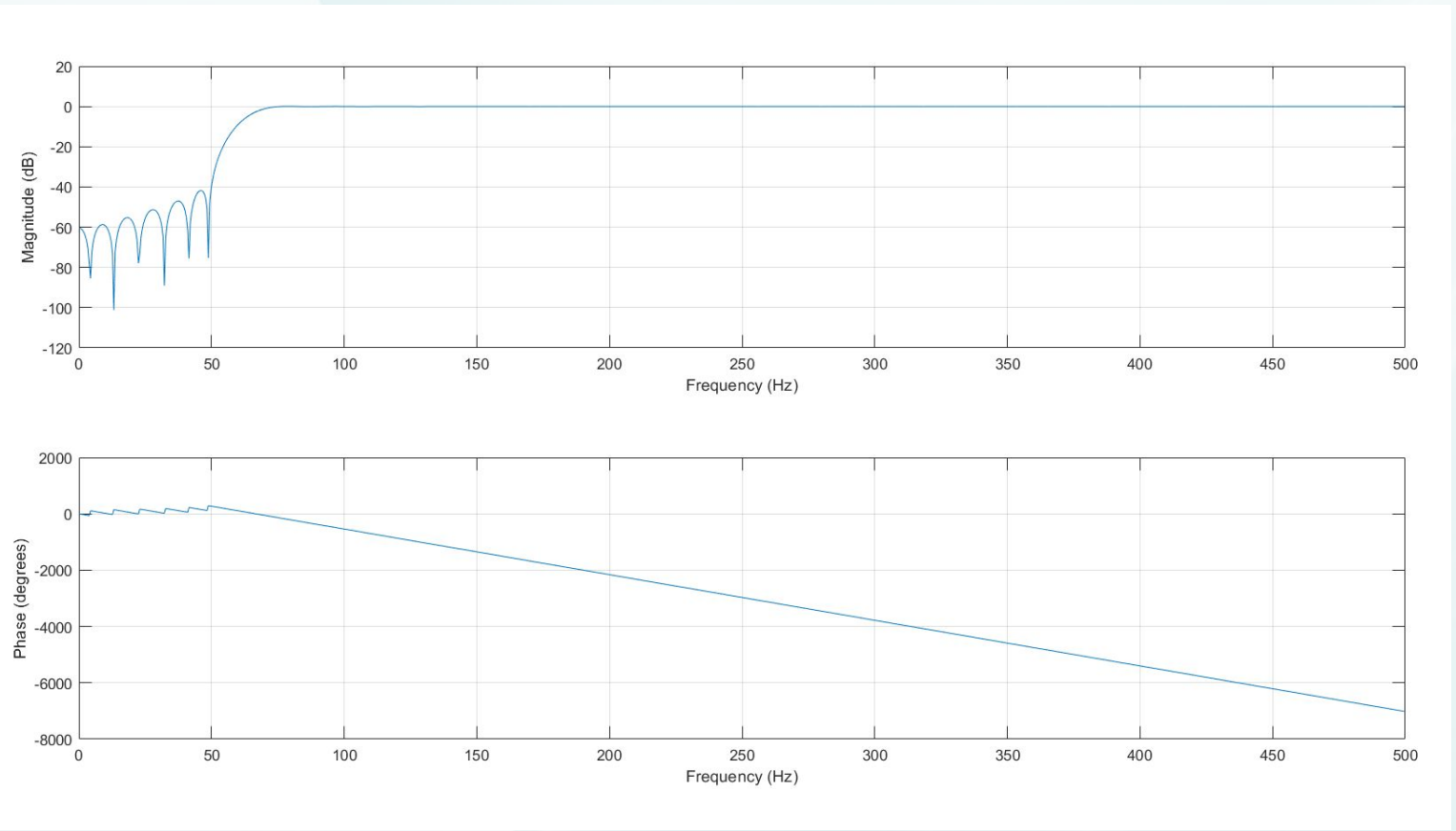
% high pass filter
mags = [0 1];

% get kaiser window
[n,Wn,beta,ftype] = kaiserord(fcuts,mags,devs,fsamp);
% calculate coefficients
hh = fir1(n,Wn,ftype,kaiser(n+1,beta),'noscale');

figure('Name','High Pass Filter');
freqz(hh,1,1024,fsamp)

fprintf("M [%d]", n);
coefs = regexprep(num2str(hh),'\s+',',')
```

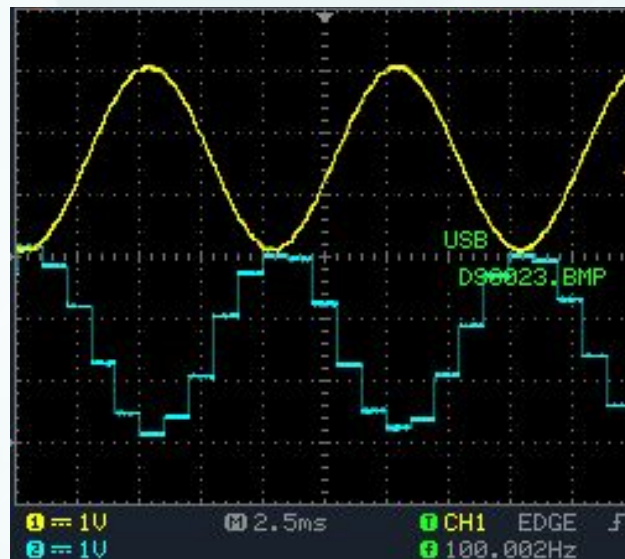
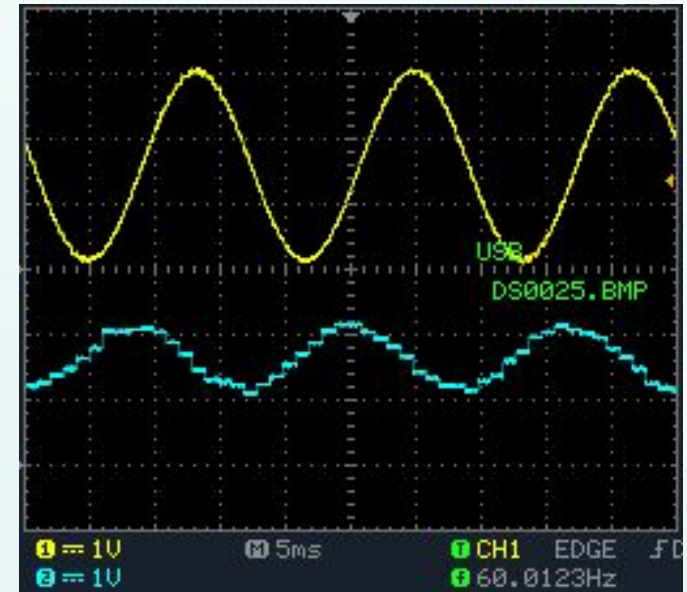
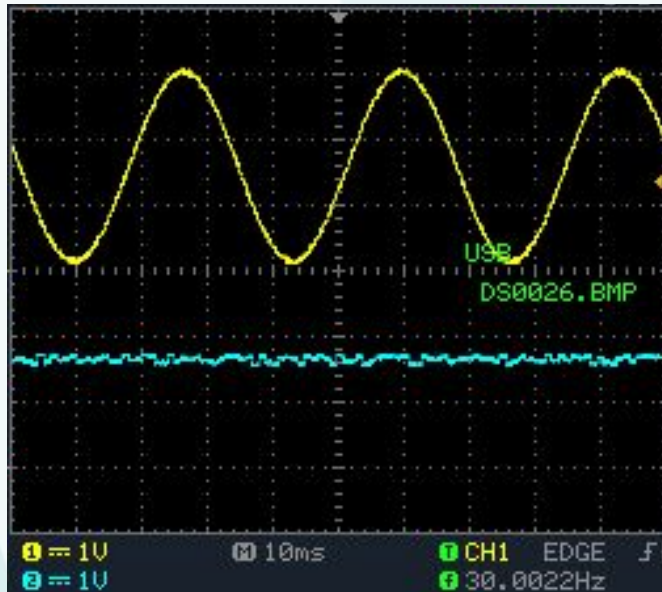

High Pass Filter



High Pass Filter



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Band Pass Filter

```
%----- Band Pass Filter (BPF)
% sampling frequency [Hz]
fsamp = 1000;
% stopband and passband frequencies [Hz]
fcuts = [50 100 150 200];
% ripples
devs = [0.01 0.01 0.01];

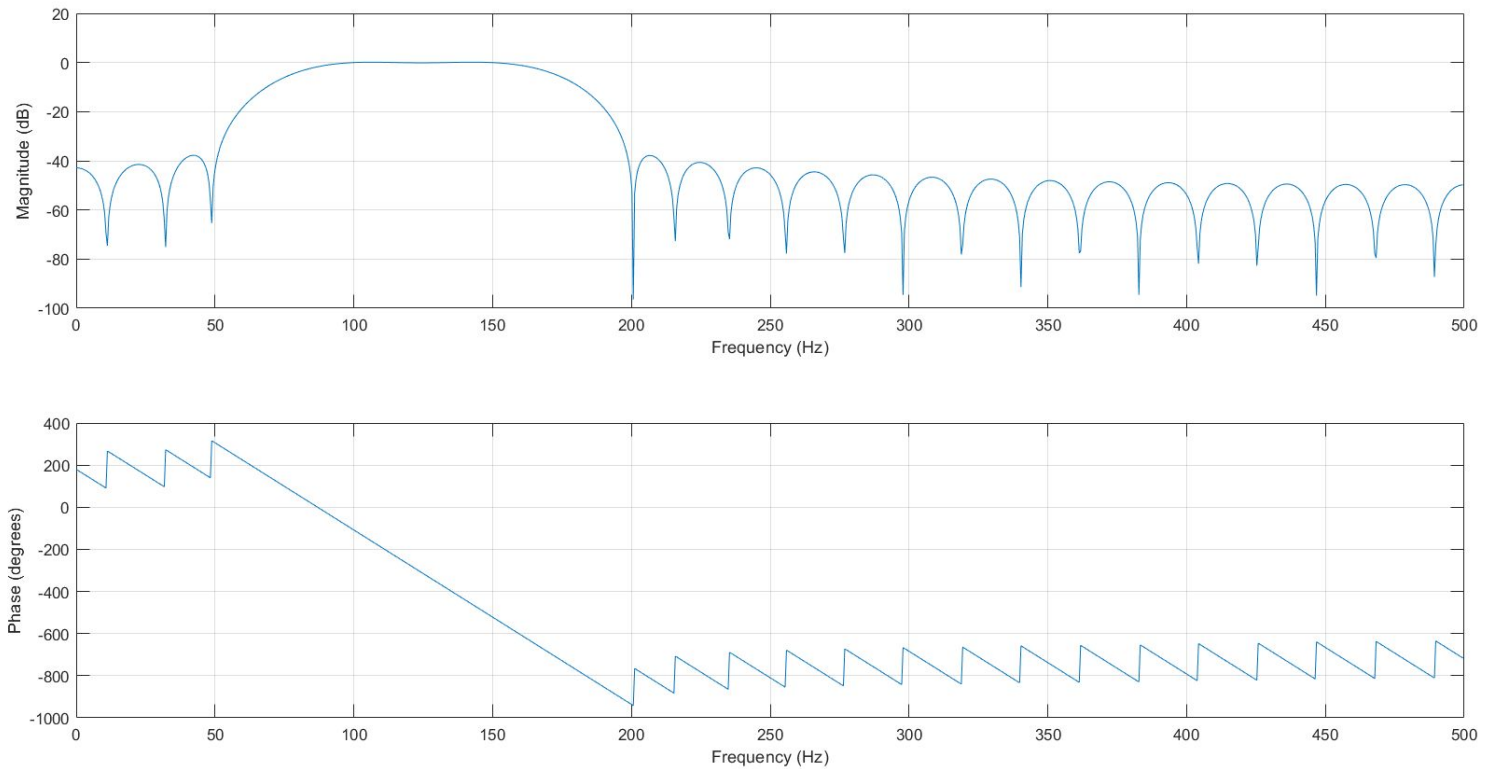
% band pass filter
mags = [0 1 0];

% get kaiser window
[n,Wn,beta,ftype] = kaiserord(fcuts,mags,devs,fsamp);
n = n + rem(n,2);
% calculate coefficients
hh = fir1(n,Wn,ftype,kaiser(n+1,beta),'noscale');

figure('Name','Band Pass Filter');
freqz(hh,1,1024,fsamp)

fprintf("M [%d]", n);
coefs = regexprep(num2str(hh),'\s+',' ','')
```

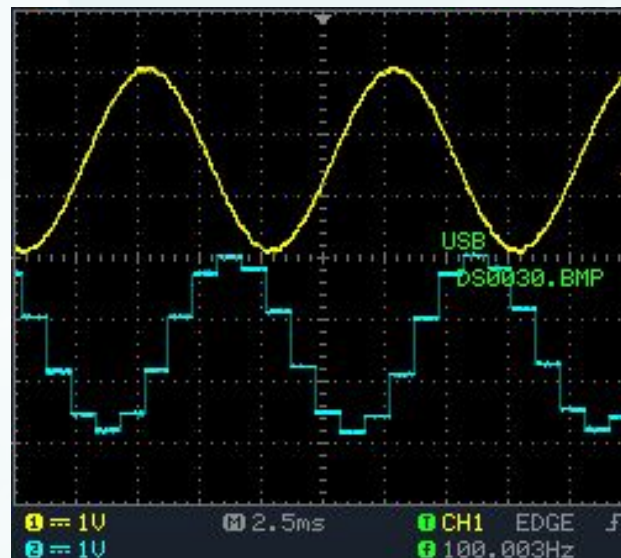
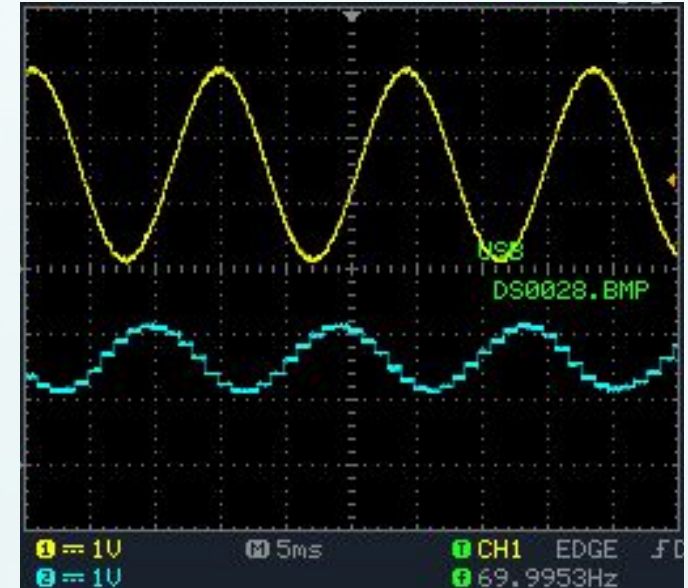
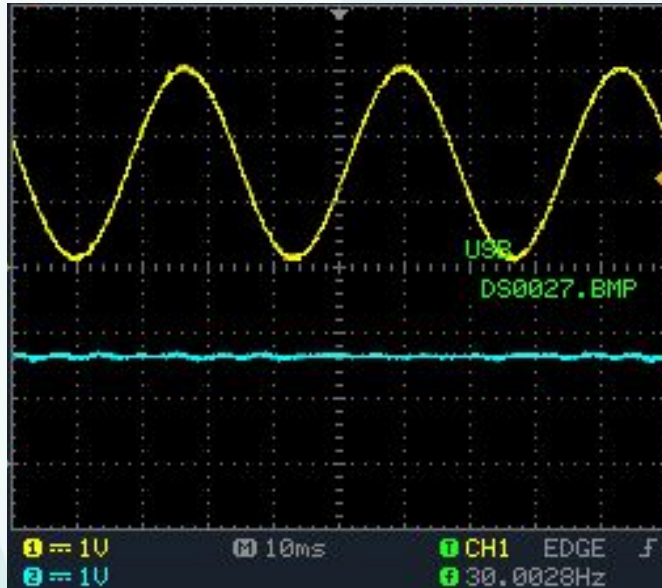
Band Pass Filter



Band Pass Filter



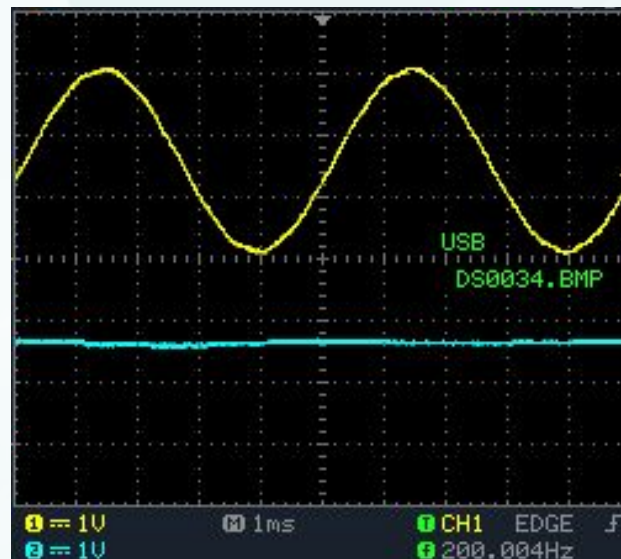
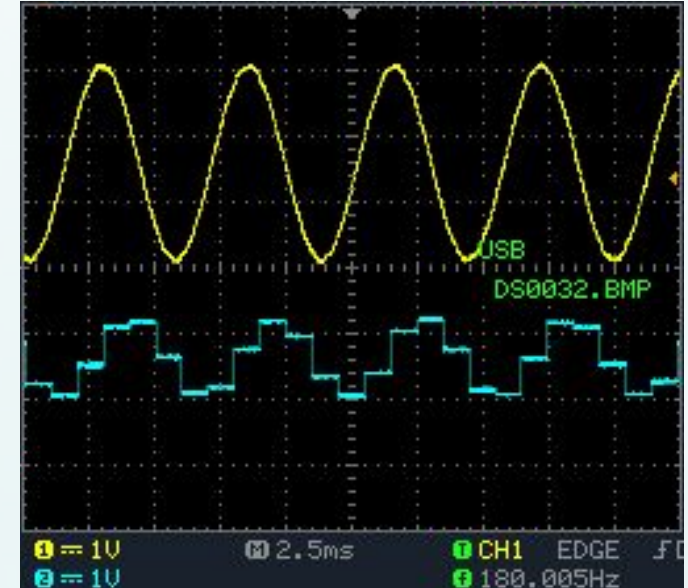
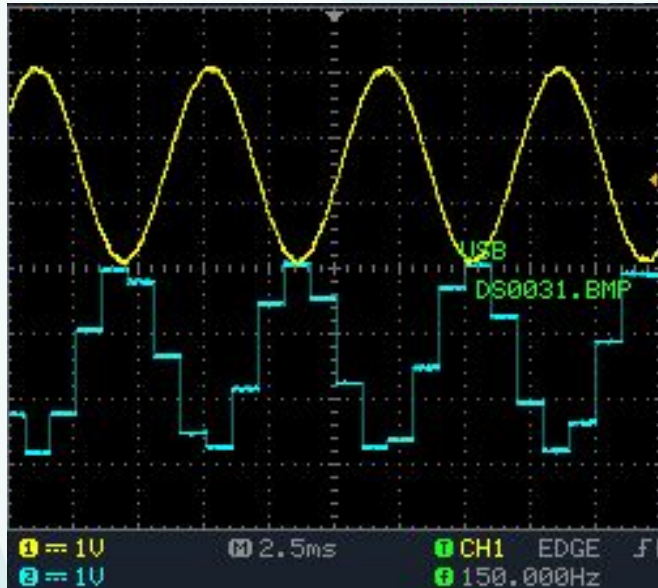
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Band Pass Filter



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Questions?