

**Aarhus Institue of Technology**

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## Filtre

This project utilizes two types of filter a FIR (finite impulse responds) filter and an IIR (infinite impulse responds) filter. The FIR filter is designed and implemented in matlab using the window method.

### FIR Filter Design

First step in designing a FIR filter is to design an ideal IIR filter before truncating it with by multiplying the IIR filter with a finite length window function.

By using our spectral analysis from the earlier sections, we qualitatively decided to make the cutoff frequency  $f_c = 10$ . The sample frequency  $f_s$  is given from our dataset,  $f_s = 47.7774$ .

Lastly, the filters made order  $M = 250$  thus using 250 filter coefficients.

### Resolution

Next step in designing our filter, we determine the frequency resolution which provides specifications for the FIR transfer function.

$$f_{res} = \frac{f_s}{M}$$

### Transfer function

Using  $f_{res}$  and  $f_c$ , we can determine which frequency bin corresponds to frequencies below  $f_c$ . This must be in done in integer values i.e. rounded to closes integer value.

$$f_{bin} = \left\lfloor \frac{f_c}{f_{res}} \right\rfloor = 52$$

In this case the bin number corresponds to  $f_c$  is 52, which we design our lowpass filter around see ??.

These specification help determine which frequencies should be passed and which should be removed. In this case we remove everything above frequency 10.

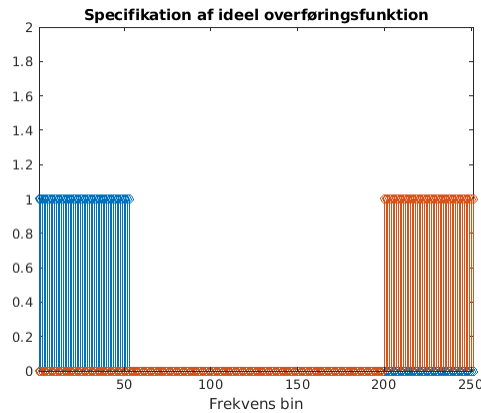


Figure 1: Specification for our transfer function

In matlab we can now make our transfer function  $h$  using the following code. Which when combined with, in our case a hanning window function serves to become our filter which we can run our data set through.

```

                                matlabStuff/filter.m
1 | H_left  = [1 ones(1,freq_bin_round) zeros(1,(M/2)-freq_bin_round)];
2 | H_right = fliplr(H_left(2:end));
3 | h = fftshift(real(ifft(H)));
4 | w_hanning = hanning(M+1)';
5 | h_win = h.*w_hanning;
6 |
7 | H_without_win = fft(h,f_sample_round);
8 | H_with_win = fft(h_win,f_sample_round);

```

Figure 2: matlab code for making a transfer function

Now that we have our transfer function, we can apply it to our data to see if we can reduce the potential noise in data. we can plot its coefficients together with the window function, along with the resultat transfer function to see if it does indeed “cover” the peak around  $10Hz$ .

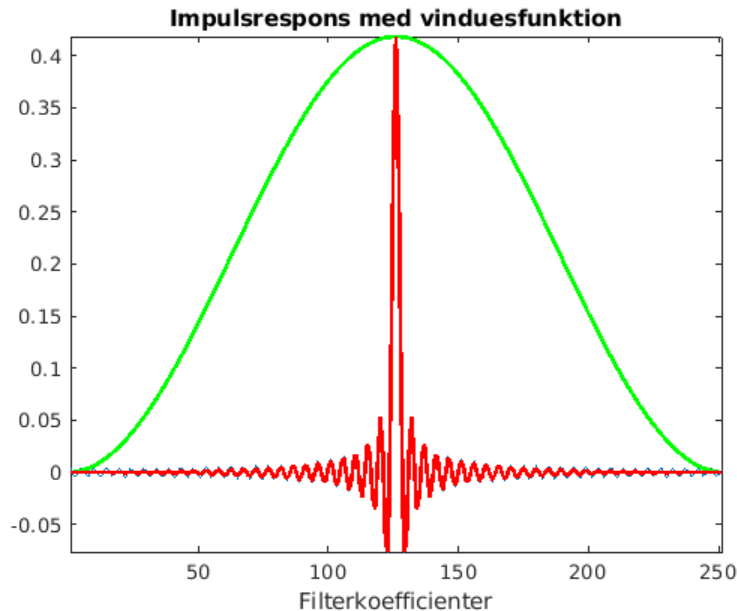


Figure 3: Impulsrespons overlaid with the window function, showing its koefficient values as functions of index numbers

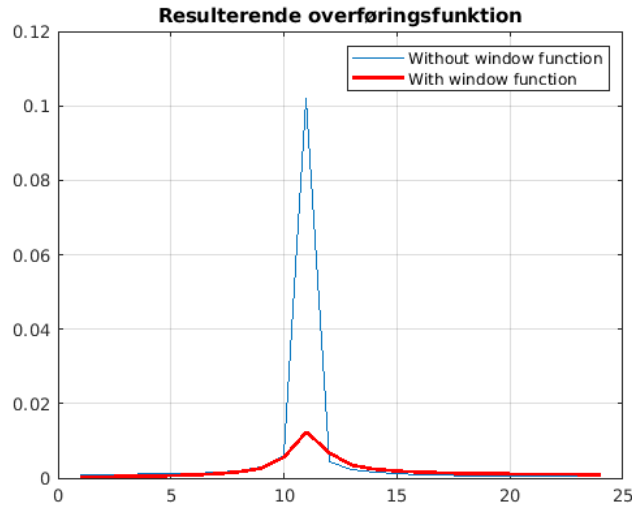


Figure 4: Resulting transfer function overlaid with and without the window function

### Applying the FIR filter

With all of the preparation work done, we can now apply our filter to our dataset to see the results.

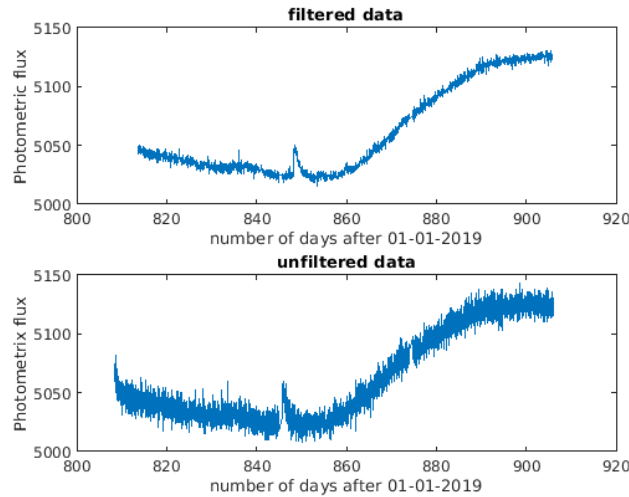


Figure 5: Side by side comparison of the data before and after the filter has been applied

Its clear from this comparison that we have manage to remove a some of the signal noise, but since this data set does not show anything usefull, we wont be able to tell anything from this filtering.