

Example 1 - Warped low pass filter

January 25, 2017

0.1 Example:

Design a **warped low pass filter** with cuto frequency of $0.05 * \pi$ (π is the Nyquist frequency). Observe: here this frequency is the end of passband, with frequency warping close to the Bark scale of human hearing.

First as a comparison: design an **unwarped filter** with 4 coefficients/taps with these specifications:

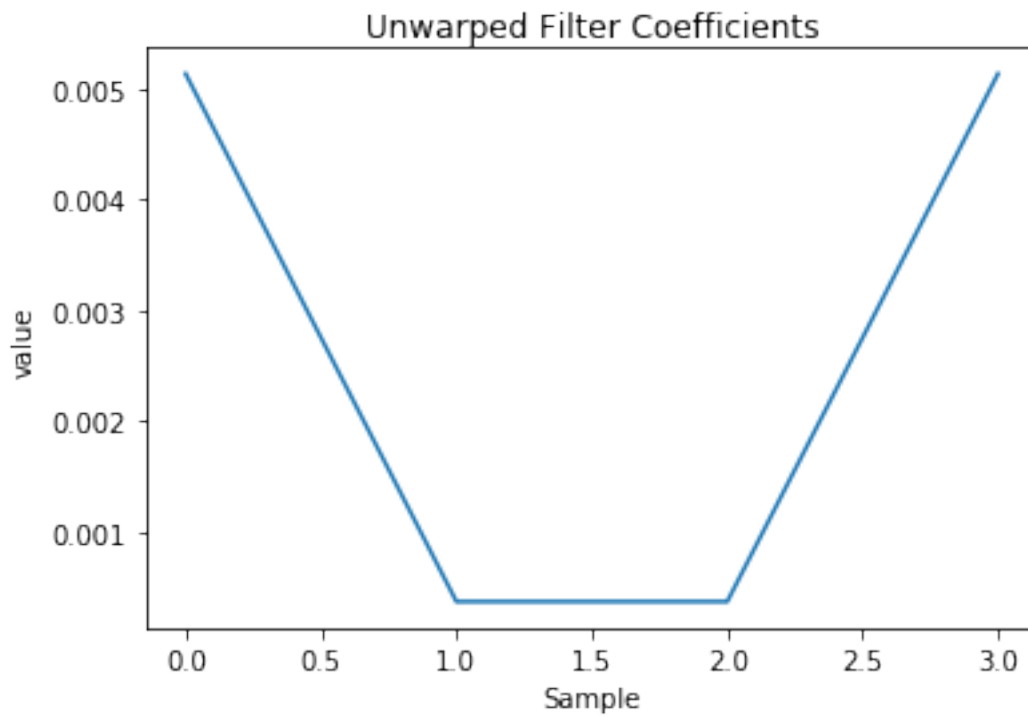
In Python:

```
In [1]: import scipy.signal as sp
        cunw = sp.remez(4, [0, 0.025, 0.025+0.025, 0.5], [1,0], [1, 100])
        print cunw
```

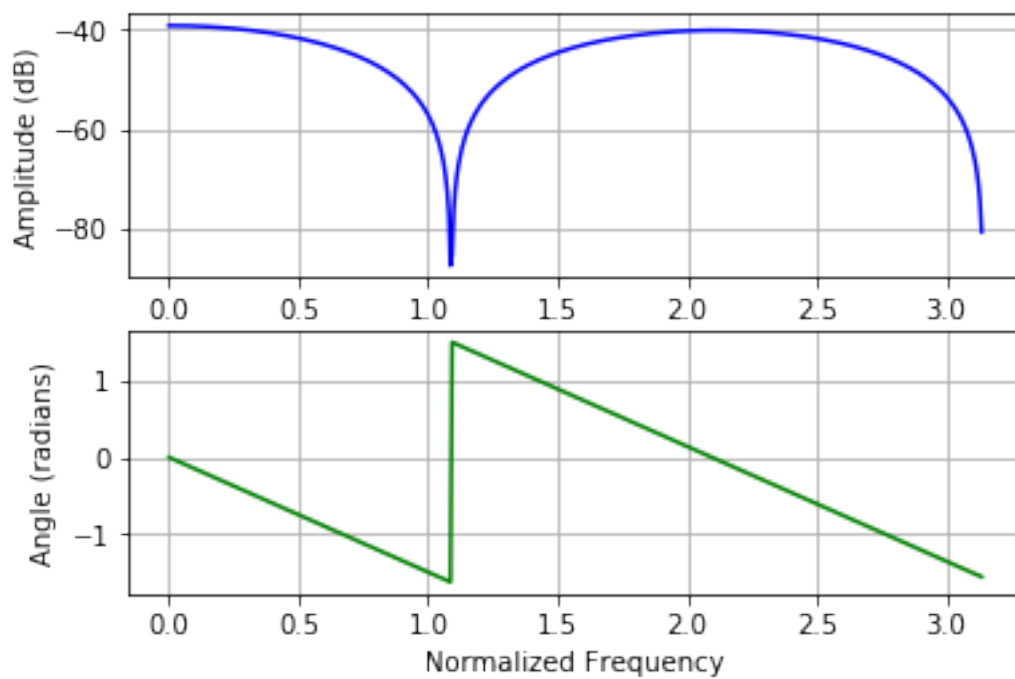
```
[ 0.00513655  0.00037423  0.00037423  0.00513655]
```

```
In [2]: #impulse response:
        %matplotlib inline
        import matplotlib.pyplot as plt
        plt.plot(cunw)
        plt.xlabel('Sample')
        plt.ylabel('value')
        plt.title('Unwarped Filter Coefficients')
```

```
Out[2]: <matplotlib.text.Text at 0x10d844d0>
```



```
In [3]: #frequency response:
from freqz import freqz
freqz(cunw, 1)
```



Here we can see that this is not a good filter. The passband is too wide (up to about 0.15), and there is almost no stopband attenuation (in the range of 0.5 to 0.9). So this filter is probably **useless** for our application.

Now design the FIR low pass filter (4th order), which we then want to frequency warp in the next step, with a warped cutoff frequency. First we have to compute the allpass coefficient a for our allpass filter which results in an approximate Bark warping, according to Julius O. Smith and Jonathan S. Abel, "Bark and ERB Bilinear Transforms," IEEE Transactions on Speech and Audio Processing, vol. 7, no. 6, pp. 697 – 708, November 1999: ###

$$a = 1.0674 \cdot \left(\frac{2}{\pi} \cdot \arctan(0.6583 \cdot f_s) \right)^{0.5} - 0.1916$$

with f_s as the sampling frequency in kHz. Our warped design is then,

```
In [4]: from warpingphase import *
import numpy as np
#warping allpass coefficient:
a = 1.0674 * (2 / np.pi * np.arctan(0.6583 * 32)) ** 0.5 - 0.1916
print a
```

0.859559698991

```
In [5]: # with f_s=32 in kHz. from [1]
# The warped cutoff frequency then is:
fcw = -warpingphase(0.05 * np.pi, 0.85956)
print fcw
```

1.61200904751

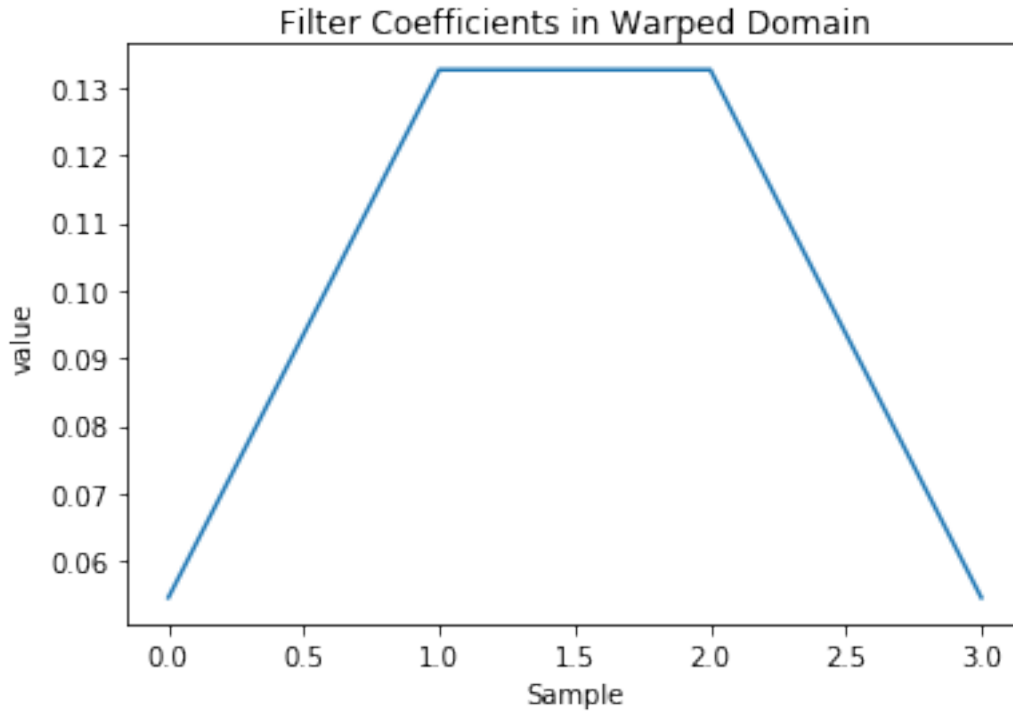
```
In [6]: # fcw = 1.6120; %in radians
# filter design:
# cutoff frequency normalized to nyquist:
fcny=fcw/np.pi
print fcny
```

0.513118416439

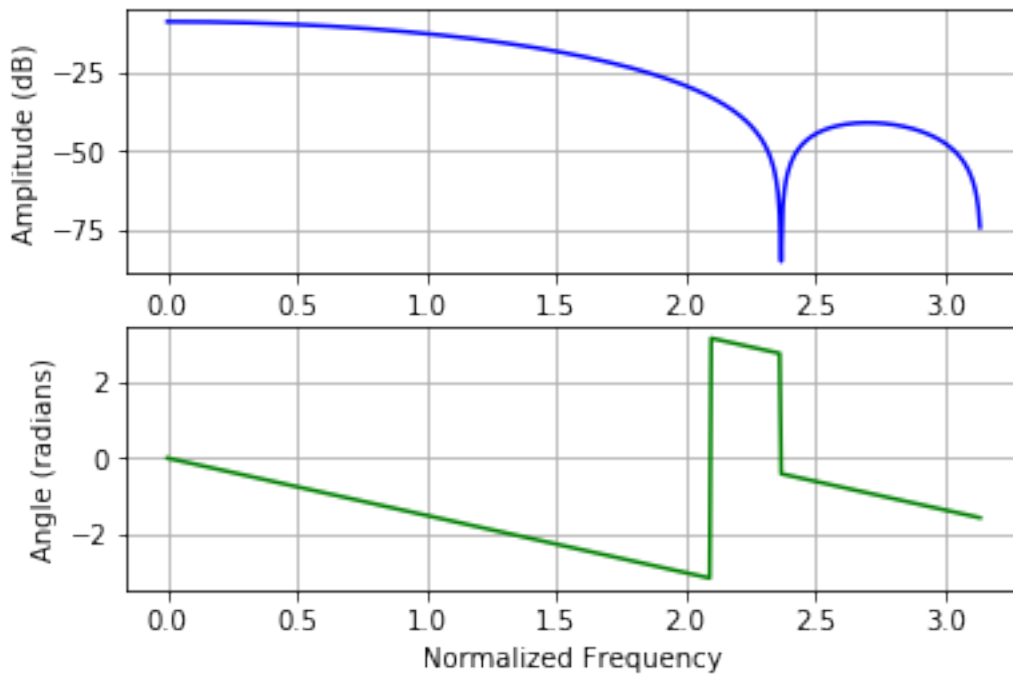
```
In [7]: c = sp.remez(4, [0, fcny/2.0, fcny/2.0+0.1, 0.5], [1, 0],[1, 100])
```

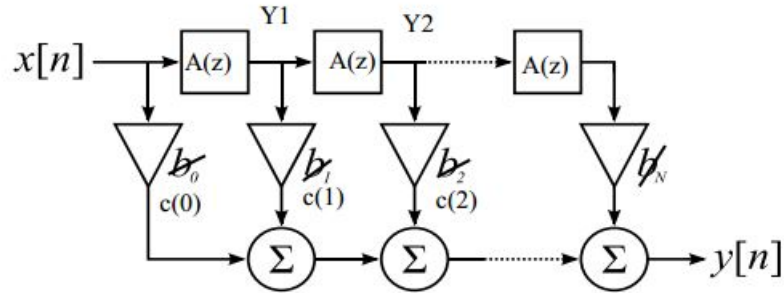
```
In [8]: #The resulting Impulse Response:
plt.plot(c);
plt.xlabel('Sample')
plt.ylabel('value')
plt.title('Filter Coefficients in Warped Domain')
```

Out[8]: <matplotlib.text.Text at 0x11022150>



In [9]: *#The resulting Frequency response:*
`freqz(c,1)`





WarpedFilter

This is the filter we obtain from the c coefficients if we don't replace the delays by allpasses. Here we can see that in the warped domain, we obtain a reasonable low pass filter. In the passband from 0 to somewhat above 0.5 it has a drop of about 10 dB, and in the stopband we obtain about -30 dB attenuation, which is much more than before (it might still not be enough for practical purposes though).

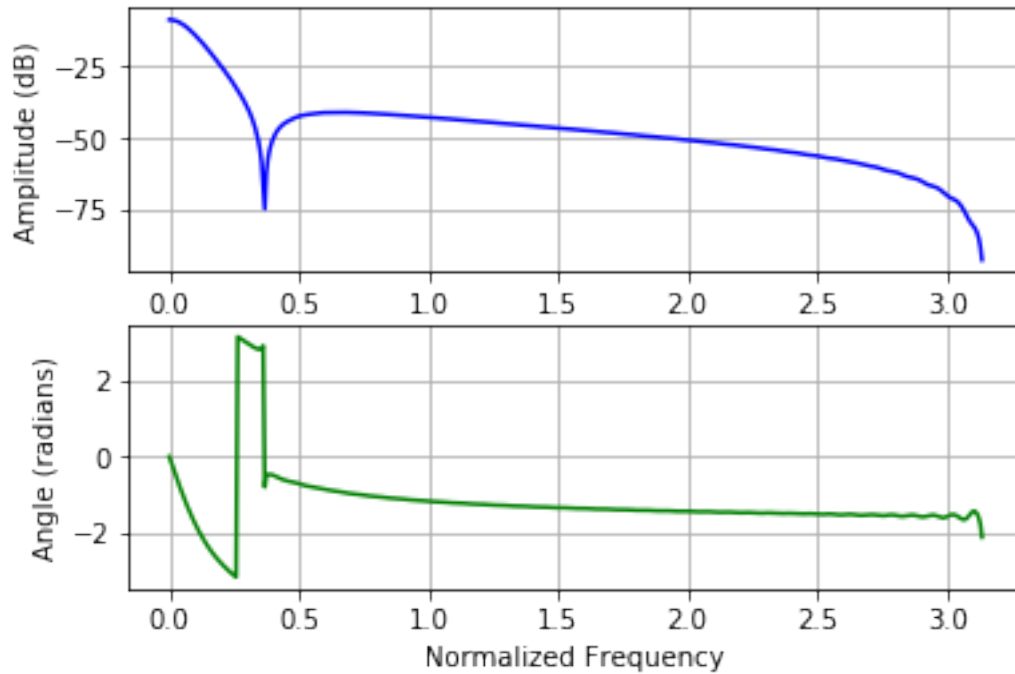
Now we use the same c coefficients, but replace the Delays in the FIR filter with Allpass filters (in this way we go from frequency response $H(z)$ to $H(1/A(z))$):

```
In [10]: # Warping Allpass filters:
#Numerrator:
B = [-a.conjugate(), 1]
#Denominator:
A = [1, -a]
# Impulse with 80 zeros:
Imp = np.zeros(80)
Imp[0] = 1
x = Imp

# Y1(z)=A(z), Y2(z)=A^2(z),...
# Warped delays:
y1 = sp.lfilter(B,A,x)
y2 = sp.lfilter(B,A,y1)
y3 = sp.lfilter(B,A,y2)

# Output of warped filter with impulse as input:
yout = c[0]*x+c[1]*y1+c[2]*y2+c[3]*y3

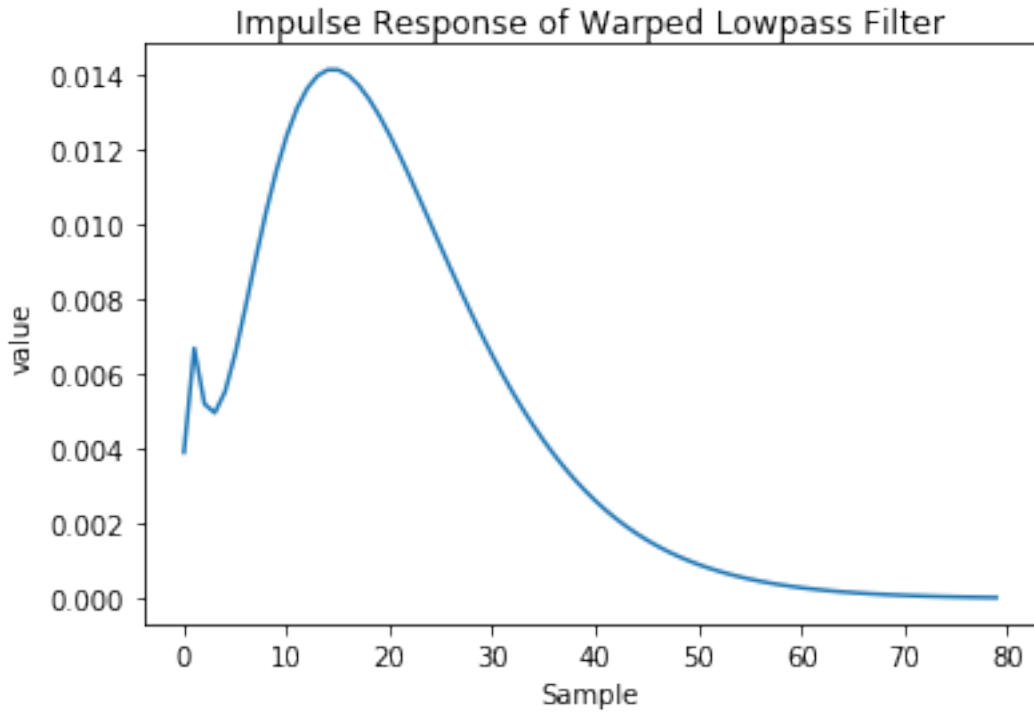
In [11]: # frequency response:
freqz(yout, 1)
```



Here we can now see the frequency response of our final warped low pass filter. We can see that again we have a drop of about 10 dB in the passband, now from 0 to 0.05π , and a stopband attenuation of about 30dB, which is somewhat reasonable.

```
In [12]: #Impulse response:
plt.plot(yout);
plt.xlabel('Sample')
plt.ylabel('value')
plt.title('Impulse Response of Warped Lowpass Filter')
```

```
Out[12]: <matplotlib.text.Text at 0x11772970>
```



This is the resulting impulse response of our warped filter. What is most obvious is its length. Instead of just 4 samples, as our original unwarped design, it easily reaches 80 significant samples, and in principle is infinite in extent. This is also what makes it a much better filter than the unwarped original design!