## **Homework 5. Butterworth Filters**

The Matlab and Python signal processing modules have an overwhelming array of options for designing and implementing filters, but for many geo-scientific applications we can use very simple filters. In this exercise we are going to explore the properties and use of a Butterworth IIR digital filter – perhaps the most commonly used filter.

Read the butter () documentation on Matlab or Python. It is important to remember that

- The order of the filter n is the order of the polynomials defined by a and b (i.e., the number of poles or zeros). Usually n is chosen to be even. The higher the order the sharper the frequency cutoff.
- The frequency cutoff limit(s) of the filter wn are specified in units that go from 0 to <u>1 at</u> the Nyquist frequency
- b and a are as defined in class. That is, the response of the filter is

$$G(z) = \frac{B(z)}{A(z)} = \frac{b_0 + b_1 z + b_2 z^2 + \dots + b_M z^M}{a_0 + a_1 z + a_2 z^2 + \dots + a_N z^N}$$

Once you've obtained the coefficients, to apply a filter to the sequence x, execute the command lfilter() in python or filter() in Matlab.

This command calculates

$$y_{j} = \overset{N}{\overset{o}{\circ}} b_{i} x_{j-i} + \overset{M}{\overset{o}{\circ}} a_{i} y_{j-i}$$

Note that the indexing is different in Python and Matlab conventions. Now for any given output  $y_j$ , the filter requires the N previous input values of x and the M previous output values of y and so will not be getting all the inputs until j = max(N,M) + 1. Furthermore the feedback component will typically require several times M prior values to fully settle down (if you think about  $y_{M+1}$ , it will be getting prior values of y but they will in turn have been calculated without the necessary prior values) – Unless the time series starts off with zeros, always give the filter a little extra data and delete the first part of the output.

- 1. For this question start with a 128-sample time vector with unit sample interval and a start time of 0 and an input time series function comprising a delta function at 0.
- (a) Create a 2<sup>nd</sup>, 4<sup>th</sup>, 8<sup>th</sup> and 16<sup>th</sup> order low-pass Butterworth filters with a cutoff at 0.5 of the Nyquist frequency. Apply them individually to the delta function time series and plot the results.
- (b) For each filter in (a) plot its amplitude and phase.
- (c) Comment on the differences of the time- and frequency-domain responses of the filters.
- (d) Repeat (a) except for a high pass filter with Wn = [0.8]. Comment on the results.
- (e) Repeat (a) except for a notch (stop) filter with  $Wn = [0.4 \ 0.6]$ . Comment on the results.

- (f) The filtfilt command in both Python and Matlab implements a zero-phase filter by running the filter through the filtered data in both directions. Use a 128-sample time vector with unit sample interval and a start time of -63 and an input time series function comprising a delta function at t=0. Apply the filters from part (a) to the delta function and plot the results. Comment on the results.
- 2. Consider the Mauna Loa CO2 concentration time series from homework 2.
- (a) Plot the data and its amplitude spectra
- (b) Construct a 4<sup>th</sup> order Butterworth high-pass filter with an appropriate cutoff frequency so that you remove the multi-decadal trend and leave only the annual variability signal. Plot the filtered data.
- (c) Construct a 4<sup>th</sup> order Butterworth low-pass filter with an appropriate cutoff frequency so that you remove the annual trend and leave only the long-term signal. Plot the filtered data.