DSP Assignment 3 IIR filtering Report

Partheepan Shiyamsunthar, Benjamin Frazer, Cameron Richards* January 3, 2022

Contents

1	Introduction							
2 Working Principles								
3	Filter design objectives 3.1 Analogue Prototype 3.2 Sensor noise 3.3 Settling Time vs Cut-off Frequency 3.4 Filter order 3.5 Final design	2 4 4 5 6 7						
4	4.3.3 Generic High-Pass Unit test design:	8 8 9 10 11 11 12						
5	Results 1							
6	Design Review							
7	7.1 Digital Comms	14 14 14						
8	8.1 Links	15 15 15 15 15						

 $^{*2531525}S@student.gla.ac.uk,\ 2704250F@student.gla.ac.uk,\ 2391317R@student.gla.ac.uk$

8.2	Code		15
	8.2.1	realtime_iir_main.py	15
	8.2.2	rununittest.py	16
	8.2.3	calcAngles.py	19
	8.2.4	realtime_iir_main.py	20
	8.2.5	realtime_plot.pv	22

1 Introduction

This project aims to deliver an accelerometer based angel measurement device capable of returning the angle between a vector drawn between the device and the centre of the earth and three orthogonal coordinate vectors and as such providing a measurement of the tilt of the sensor in 3d space.

The high-level design objectives for this project are summarised as follows:

- return angle measurements for pitch and yaw
- ensure smooth measurements (un-disturbed by noise)
- make responsive measurements (if the user changes the angle the change in measurement is perceived as instantaneous)

2 Working Principles

The tilt sensor is predicated on the knowledge that the gravity vector will always be pointed directly down towards the centre of the earth. If the assumption is then made that the device is otherwise at rest, we may infer the pitch and roll of the device by observing the projection of the gravity vector on the respective axes.

The sensor used to measure acceleration is the ADXL335, this is a three axis accelerometer which outputs its values via DAC as three analogue signals. These analogue signals are digitised by the Arduino which is acting purely as an acquisition device. Real-time data is then passed python using the firmatta protocol, where data processing such as filtering and angle measurement occurs. Figure 1 shows the high-level structure of the tilt sensor.

3 Filter design objectives

The high-level design objectives can be translated into requirements for the filter design, namely:

- Eliminate sensor noise
- Eliminate environmental noise where possible i.e. vibrations
- Good transient behaviour for DC changes i.e. fast response, no overshoot

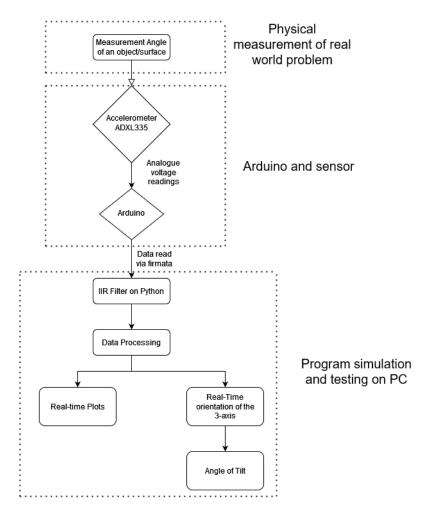


Figure 1: Tilt sensor High-level Dataflow Diagram

Acceleration at rest will be a dc value, as such a low-pass filter is required.

The primary requirement on the filter implementation is whether the filter can be realised in real time. Since python is an interpreted language, the speed of execution is fundamentally limited and as such the amount of computation required for filtering should be minimised. In this regard, IIR filters are certainly the most optimal, since on a per computation basis they offer better performance than FIR filters. Higher order IIR filters will of course still come at the cost of more computation and as such, the order of the filters should be kept minimal as a design goal.

In order to address the design goal of "responsiveness" we must first identify a metric which can suitably quantify this characteristic in the filter design. Since an instantaneous change in orientation will appear as a step change in the acceleration of a given accelerometer channel, it seems suitable to use the settling time as a metric for this goal. For the purpose of this report settling time is defined as the time taken for the step response to be bounded by ± 10

3.1 Analogue Prototype

The first design decision to be made is that of the analogue filter which will form the basis of the IIR filter design the following constitute the leading candidates:

Table 1: Qualitative comparison of candidate analogue prototype filters											
filter type	Overshoot	transition	rise	pass-band	Stop-band						
		width	$_{ m time}$	Flatness	Att						
Chebychev	poor	best	poor	poor	good						
Butterworth	poor	medium	poor	best	good						
Bessel	best	poor	best	poor	good						

Acceleration measured due to gravity is at steady state a dc value, as such the filter flatness in the pass band is not of interest, since even the distortion at DC caused by the Chebychev will not affect the angle measurement as long as it is consistent between channels. Further more, it is assumed that sensor noise will be flatly distributed in the frequency domain meaning there is no pressing requirement for a sharp transition width. With the aforementioned considerations mentioned it is clear that the benefits of the Chebychev and the Butterworth filters are largely negated by the design requirements.

The requirement for good transient characteristics penalises the Butterworth and Chebychev filters since both suffer from overshoot while the Bessel filter doesn't. The Bessel filter also has a faster settling time when compared to the other analogue designs.

3.2 Sensor noise

The first factor influencing the choice of filter cut-off is the distribution of filter noise. The noise content of the accelerometer data is confirmed by taking a recording of the filter at rest and taking a Fourier transform for a single channel. ¹

As we can see from Figure 2, sensor noise is relatively evenly distributed in the frequency domain. with only a slight concentration of harmonics around 100Hz. This means that there is no particular requirement to have a steep transition width since we are only concerned with unity response at DC and have roughly

¹After investigation noise is similar in all three channels but only one is plotted here

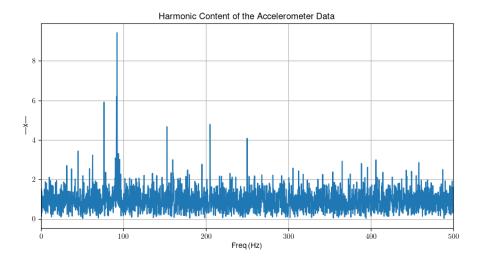


Figure 2: Fourier transform of raw accelerometer data showing sensor noise

two decades of roll-off before the dominant frequencies at 100Hz at which point even a first order bessel will suffice.

3.3 Settling Time vs Cut-off Frequency

Since the objective of the filter is to reject all but DC but also remain responsive (settling time) we must consider if and how these two requirements interact. As shown in Figure 3 there exists a fundamental trade-off between settling time and cut-off frequency.

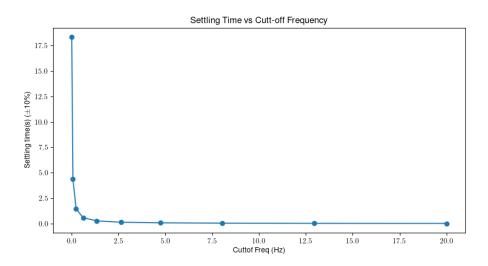


Figure 3: Trade-off between settling time and cut-off frequency for a second order Bessel filter

This plot was generated numerically by applying a step filter designs with different cutoff frequencies and logging the settling time (for Bessel this is simply the first intersection between the step response and line y = 0.9) denoted in Figure 4. The cut-off frequencies are exponentially spaced between 0.2 and 20 Hz in order to have good coverage of the corner between the two asymptotes.

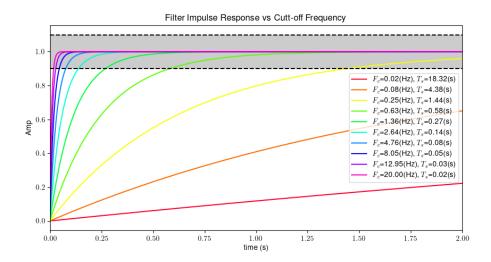


Figure 4: Impulse, frequency domain responses of second order Bessel filters with various cutoff frequencies.

For the final design, a cut-off frequency of 5 Hz will be used since this preforms adequately at removing noise while incurring little to no penalty in responsiveness (settling time).

3.4 Filter order

A further degree of freedom in the design is that of the filter order used. Typically the impact of higher filter order is faster transition width, which, as mentioned previously is only a very minor consideration here, while the increased computational cost associated with this incur a substantial penalty. It is plausible however that increased filter order might have an impact on the rise time of the filter. Figure 5 shows the trade-off plots for different orders of Bessel Filter.

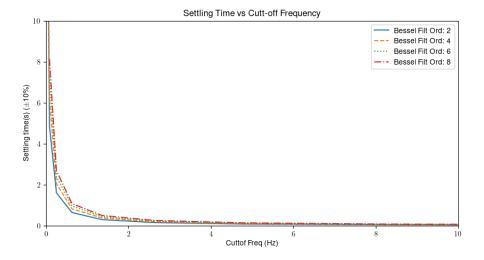


Figure 5: Trade-off between settling time and cut-off frequency for multiple orders Bessel filter orders

From this figure we can see that increasing filter order has a detrimental impact on the settling time

trade-off curve. It is a simple decision therefore to limit the filter order to first order given that there is little benefit decreased transition width. This plot was again generated numerically using exponentially spaced cut-off frequencies between 0 and 20Hz but as well as even filter orders between 2 and 8. The corresponding impulse and frequency responses of these filters are shown below.

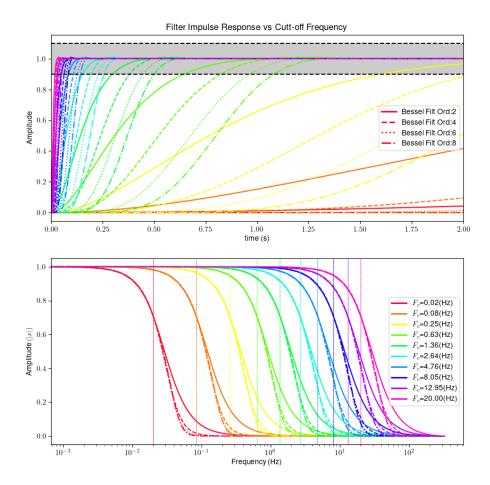


Figure 6: Impulse, frequency domain responses of second order Bessel filters with various cutoff frequencies and filter orders.

3.5 Final design

The Final Filter design parameters are given in Table 2. 2

²Filter coefficients will be presented in the following sections, since this is a function of sample rate.

Table 2: final filter design parameters
Analogue Filter Cut-off
Prototype order Frequency (Hz)

Bessel 1st 5

4 Implementation

4.1 Sample rate Verification

To get the sampling rate the number of samples recorded of a period of time were counted and then divided by the period length:

$$fs = NT \tag{1}$$

The period that samples were counted for was chosen based on the knowledge that the measurement period must be much larger than the sample rate period. The system should run at 1kHz sample rate so will have a 1ms sampling period. If the measurement period is too large the sample rate won't be calculated often enough to be practical. Therefore, a measurement period of 500ms was selected to balance both these factors.

In practice to get the half second timing the animation callback feature of matplotlib which is used to update the plot every ~100ms. A counter system was used so that the sample rate is calculated every five times the plot is updated. However, the animation callback does not have precise timing so the "time" module was used to precisely measure the time between the sample rate calculations.

This method was implemented and tested single input channel at 1kHz where the measured value stayed between 950Hz and 1050Hz. However, when more input channels are added the sample rate starts to reduce. This may be because either the Arduino or the python script can't run fast enough to process that many samples.

4.2 Filter design

The final design coefficients were derived using the scipy signal library bessel design command source-code shown below:

```
from scipy import signal
fs = 650 # sample rate
fn = fs/2
fc = 2 # cutoff frequency (Hz)
sos = signal.bessel(1, fc / fn, "lowpass", output="sos", norm="mag")
```

This design function outputs filter coefficients grouped into bi-quads. The basic structure of such a biquad is shown in Figure 7. This is the coefficient naming convention that will be used when discussing and presenting filter designs in this report.

Since the achieved sample rate was substantially lower than the requested 1kHz, the sample rate used to derive the normalised cut-off frequency was modified to account for this.

This results in the following filter coefficients:

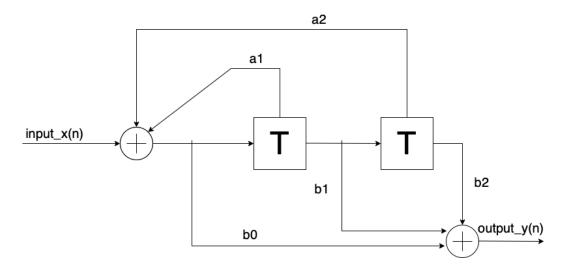


Figure 7: Direct form II Biquad IIR filter showing coefficient naming convention

4.3 Filter Unit Testing

Unit testing is conducted by first generating a generic high and lowpass design and then hand calculating the expected output by using the difference equation for a second order IIR filter.

The transfer function for a second order biquad filter is given in the z domain by:

$$H(z) = \frac{Y(z)}{X(z)} = \frac{b_0 + b_1 z^{-1} + b_2 z^{-2}}{a_0 + a_1 z^{-1} + a_2 z^{-2}}$$
(2)

This yields the following difference equation in the time-domain:

$$v[n] = x[n] - a_1 \times y[n-1] - a_2 \times y[n-2] \tag{3}$$

$$y[n] = v[n] \times b_0 + b_1 \times y[n-1] + b_2 \times y[n-2]$$
(4)

For the purpose of simplifying notation in the hand calculations the following substitutions will be made:

$$y[n-1] = T1 \tag{5}$$

$$y[n-2] = T2 \tag{6}$$

These difference equations allow the expected output of the filter to be calculated recursively.

4.3.1 Generic Low-Pass Unit test design:

This is a generic 2nd order IIR Filter with a cut-off at a normalized frequency of 0.1

4.3.2 Low-Pass hand calculations

Hand calculation for the lowpass filter when given an input signal x(n) = [1,3,5]:

```
T1 = T2 = 0
2
  Input = x(n) - (a1*T1) - (a2*T2)
  y(n) = [input]*bo + b1*T1 + b2*T2
  After each iteration, T1 = input & T2 = T1
  y(1) = [x(1) - (a1*T1) - (a2*T2)]*b0 + b1*T1 + b2*T2
       = [1 - (0) - (0)] * 0.06745527 + 0 + 0
10
       = 0.06745527
11
       = 0.0675
12
13
  T2 = T1 = 0
  T1 = input = 1
15
16
17
  y(2) = [x(2) - (a1*T1) - (a2*T2)]*b0 + b1*T1 + b2*T2
      = [3 - (-1.1429805 * 0.06745527) - (0)] * 0.06745527 + 0.13491055
19
   = [4.1429805] * 0.06745527 + 0.13491055 * 1 + 0
       = 0.4143764182
       = 0.4144
22
23
  T2 = T1 = 1
  T1 = input = 4.1429805
25
26
   y(3) = [x(3) - (a1*T1) - (a2*T2)]*b0 + b1*T1 + b2*T2
27
       = [5 - (-1.1429805 * 4.1429805) - (0.4128016 * 1)] * 0.06745527 +
     0.13491055
               * 4.1429805 + 0.06745527 * 1
29
       = [9.322544323] * 0.06745527 + 0.13491055 * 4.1429805 +
30
   → 0.06745527 * 1
       = 1.255241792
31
       = 1.2552
32
33
   T2 = T1 = 4.1429805
  T1 = input = 9.322544323
```

Therefore our output y(n) = [0.0675, 0.4144, 1.2552]

4.3.3 Generic High-Pass Unit test design:

This is a 2nd order IIR Filter with a cut-off at a normalized frequency of 0.3.

Compare these output values with the filter output when used in the unittest.py program.

4.3.4 Highpass hand calculations

Hand calculation for the highpass filter when given an input signal x(n) = [6,-8,3]:

```
T1 = T2 = 0
   Input = x(n) - (a1*T1) - (a2*T2)
  y(n) = [input]*bo + b1*T1 + b2*T2
  After each iteration, T1 = input & T2 = T1
           = [x(1) - (a1*T1) - (a2*T2)]*b0 + b1*T1 + b2*T2
       = [6 - (0) - (0)] * 0.20657208 + 0 + 0
       = 1.23943248
11
       = 1.2394
12
13
  T2 = T1 = 0
   T1 = input = 6
15
16
           = [x(2) - (a1*T1) - (a2*T2)]*b0 + b1*T1 + b2*T2
17
       = [-8 - (0.36952738 *6) - (0)] * 0.20657208 + (-0.41314417 * 6) + 0
18
       = [-10.21716428] * 0.20657208 + (0.41314417 * 6) + 0
19
       = -4.589445897
20
       = -4.5894
21
22
  T2 = T1 = 6
23
   T1 = input = -10.21716428
25
          = [x(3) - (a1*T1) - (a2*T2)]*b0 + b1*T1 + b2*T2
   y(3)
26
       = [3 - (0.36952738 * -10.21716428) - (0.19581571 * 6)] *
27
    \rightarrow 0.20657208+ (-0.41314417
               * -10.21716428) + 0.20657208 * 6
28
       = [5.600627687] * 0.20657208+ (-0.41314417 * -10.21716428) +
       0.20657208 * 6
       = 6.617527647
30
       = 6.6175
31
```

```
32

33  T2 = T1 = -10.21716428

34  T1 = input = 5.600627687

35

36  Therefore our output y(n) = [1.2394, -4.5894, 6.6175]
```

4.4 Angle measurement

The three angels defining the orientation of the device are simply defined as the angel between the orientation vector and the three orthogonal coordinate axes. Since this formula is invariant to the magnitude of the vectors involved, the unit vector is used for simplicity (denoted as \hat{x}).

$$xangle = cos^{-1} \frac{\vec{V}_{xyz} \cdot \hat{x}}{|\hat{x}| \times |\vec{V}_{xyz}|}$$
 (7)

The source code for the is shown below, this calculation is implemented for all three angels in a for loop.

```
def calcAngles(vec):
       """returns the angles of a 3d vector relative to the orthoginal
       unit vectors"""
       angle = np.zeros(3)
       referenceFrame = []
       referenceFrame.append([1.0,0.0,0.0])
       referenceFrame.append([0.0,1.0,0.0])
       referenceFrame.append([0.0,0.0,1.0])
       for i in range(3):
           if DoDebug:
               print(f"unitVec{i} = {referenceFrame[i]}")
           part1=np.dot(vec,referenceFrame[i])/
       (ln.norm(vec)*ln.norm(referenceFrame[i]))
           angle1 = np.arccos(part1)
12
           angle[i] = np.rad2deg(angle1)
13
```

4.5 Setup

Figure 8 shows the setup used for testing the tilt sensor. The Arduino board is connected to the ADXL335 breakout board via three jumper wires, while power and ground is supplied from the arduino onboard power pins.

5 Results

A full demonstration of the filter operating in real time can be found here (full URL in appendices):

 $^{^{3}}$ The final angle used in the display is chosen based on which corresponds to the longest side of the breadboard, as such tilt is only returned for one axis in the demonstration video

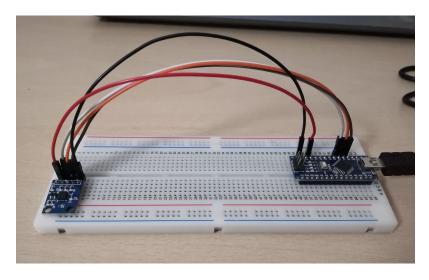


Figure 8: Tilt sensor Hardware setup

- Demonstrating Angle Measurement
- Demonstrating Responsiveness

Figures 9 and 10 show the effect of filtering on the raw accelerometer data. It can be seen that the filter acts to smooth the data while also introducing a time lag.

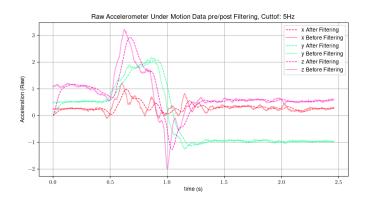


Figure 9: Raw accelerometer data before and after filtering at rest

From the video demonstration it can be seen that the tilt sensor returns accurate and stable angle measurements. Further more the tilt sensor reacts quickly to changes in angle.

6 Design Review

Given that the testing of the filter revealed that it provided accurate angle measurement, wile remaining responsive, it can be said that the initial design objectives of the project have been met.

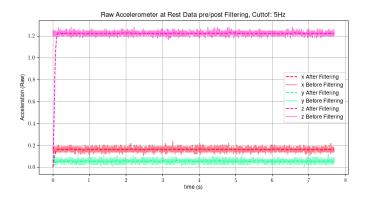


Figure 10: Raw accelerometer data before and after filtering at rest

One potential weakness of the design is the hard coded nature of the filter design while the sample rate seems constrained by the hardware in use and may as such be variable between users. This however should not substantially affect the usage of the filter since changes in sample rate for a given set of filter coefficient will merely act to move the effective cut-off frequency of the filter which, as discussed in prior sections is not a critical component of the design.

7 Future work

7.1 Digital Comms

One possible improvement could come in the form of a digital interface rather than analogue voltages between the arduino such as the I2C protocol. This was initially attempted but abandoned due to difficulties in using and debugging I2C with the firmatta library. the benefit of such a comms protocol would be eliminating a potential source of noise in the wires, and not being limited to the analogue resolution of the arduino ADCs which is substantially lower than those on the accelerometer itself.

7.2 Dynamic recalculation of filter coefficient

To combat the aforementioned drift of cut-off frequency due to hardware dependent sample rate, the filter could be recalculated "on the fly" based on the measure sample rate.

8 Appendices

- 8.1 Links
- 8.1.1 Project Source code
- 8.1.2 Filter Source code
- 8.1.3 Angle Measurment Demo https://youtu.be/plzuRrrh6NQ
- 8.1.4 Responsiveness Demo https://youtu.be/6KL1qg4h90E
- 8.2 Code
- 8.2.1 realtime iir main.py

```
""" Modules """
  from pyfirmata2 import Arduino
  from scipy import signal
  import numpy as np
  import realtime_plot as rtp
  import iir_filter as iir
  """ Constants """
  # Arduino Sample Rate
                           # BF 3/01/22 - changed to "true" sample rate
_{11} fs = 650
   → Max 1000
  # Nyquist
 fn = fs / 2
 fc = 5
15
  """ IIR Filter Design """
  # Noise Removal
  sos = signal.bessel(1, fc / fn, "lowpass", output="sos", norm="mag") #
   → BF 3/01/22 - changed to bessel
19
  x_filter = iir.IIR_filter(sos)
  y_filter = iir.IIR_filter(sos)
21
  z_filter = iir.IIR_filter(sos)
22
  """ Real Time Plotters """
  sample_plot = rtp.RealtimePlots(fs, 2, ["X", "Y", "Z"],

    sample_limits=[-0.25,0.25], channels=3)
  orientation_plot = rtp.RealtimeVectorPlot()
27
   """ Sample Process Function """
28
  def addX(data):
29
      # Zeroing from measurements
       data -= 0.335
31
       f = x_filter.filter(data)
32
33
```

```
sample_plot.addSample(data, f, channel=0)
34
       orientation_plot.addSample(f, channel=0)
35
36
   def addY(data):
37
       # Zeroing from measurements
38
       data -= 0.340
       f = y_filter.filter(data)
40
41
       sample_plot.addSample(data, f, channel=1)
42
       orientation_plot.addSample(f, channel=1)
43
44
   def addZ(data):
45
       # Zeroing from measurements
       data -= 0.340
47
       f = z_filter.filter(data)
48
49
       sample_plot.addSample(data, f, channel=2)
50
       orientation_plot.addSample(f, channel=2)
51
52
   """ Aurdino Data Aquisition """
53
   board = Arduino(Arduino.AUTODETECT)
   board.samplingOn(1000/fs)
55
   board.analog[0].register_callback(addX)
56
   board.analog[0].enable_reporting()
  board.analog[1].register_callback(addY)
   board.analog[1].enable_reporting()
   board.analog[2].register_callback(addZ)
   board.analog[2].enable_reporting()
62
   """ Show Real Time Plots """
63
   rtp.plt.show()
   board.exit()
65
```

8.2.2 rununittest.py

```
import unittest
   import iir_filter
   import scipy.signal as signal
3
4
   """Function used to collect the IIR filter output influenced by the
   → specifications"""
   def getCoefficients(w, order, system_type, input_signal):
       coeff = []
       irr_result = []
10
       b, a = signal.butter(order, 2 * w, system_type)
11
12
       for i in b:
13
```

```
coeff.append(i)
14
15
       for i in a:
16
           coeff.append(i)
17
       f = iir_filter.IIR2_filter(coeff)
20
       for i in range(len(input_signal)):
21
           irr_result.append(round(f.filter(input_signal[i]), 4))
22
23
       return irr_result
24
25
   """Function used to collect the IIR filter output influenced by SOS at
27
   → given specifications"""
28
29
   def getSOSCoefficients(w, order, system_type, input_signal):
30
       sos = signal.butter(order, 2 * w, system_type, output='sos')
31
       sos check = []
32
       fi = iir_filter.IIR_filter(sos)
34
35
       for i in range(len(input_signal)):
36
           sos_check.append(round(fi.filter(input_signal[i]), 4))
37
38
       return sos_check
39
40
41
   """Unit test class function for a lowpass and highpass filters"""
42
43
44
   class TestStringMethods(unittest.TestCase):
45
46
       def test lowpass 2nd order filter(self):
47
           cutoff_freq = 0.1 # define the normalized cutoff frequency
48
           filter_order = 2 # state the filter order
49
           filter_type = 'lowpass' # state the type of filter used
50
           test_input = [1, 3, 5] # define the input signal coefficients
51
52
           hand_calculated_values = [0.0675, 0.4144, 1.2552] # array
53
       that contains the hand calculated values for the
           # above specification
54
55
           filter_calculated_values = getCoefficients(cutoff_freq,
56
       filter_order, filter_type,
                                                         test_input)
57
       array contains the IIR filter output values for the
           # same specification
58
```

```
59
           self.assertEqual(hand_calculated_values,
60
                             filter_calculated_values) # compare both
61
      arrays to check if they aer similar
62
       def test_lowpass_sos_filter(self):
           cutoff_freq = 0.1 # define the normalized cutoff frequency
64
           filter_order = 2 # state the filter order
65
           filter_type = 'lowpass' # state the type of filter used
66
           test_input = [1, 3, 5] # define the input signal coefficients
67
68
           hand_calculated_values = [0.0675, 0.4144, 1.2552] # array
69
       that contains the hand calculated values for the
           # above specification
70
71
           filter_sos_calculated_values = getSOSCoefficients(cutoff_freq,
72
       filter_order, filter_type,
                                                               test input)
73
       # array contains the sos influenced IIR filter
           # output values for the same specification
74
75
           self.assertEqual(hand_calculated_values,
76
                             filter_sos_calculated_values) # compare both
77
      arrays to check if they aer similar
78
       def test_highpass_2nd_order_filter(self):
79
           cutoff_freq = 0.3 # define the normalized cutoff frequency
           filter_order = 2 # state the filter order
           filter_type = 'highpass' # state the type of filter used
82
           test_input = [6, -8, 3] # define the input signal
83
       coefficients
84
           hand_calculated_values = [1.2394, -4.5894, 6.6175] # array
85
       that contains the hand calculated values for the
           # above specification
86
87
           filter calculated values = getCoefficients(cutoff freq,
88
       filter_order, filter_type,
                                                        test_input) #
89
       array contains the IIR filter output values for the
           # same specification
90
91
           self.assertEqual(hand_calculated_values,
92
                             filter_calculated_values) # compare both
93
      arrays to check if they aer similar
94
       def test_highpass_sos_filter(self):
95
           cutoff_freq = 0.3 # define the normalized cutoff frequency
96
           filter_order = 2 # state the filter order
97
```

```
filter_type = 'highpass' # state the type of filter used
98
            test_input = [6, -8, 3] # define the input signal
99
      coefficients
100
            hand_calculated_values = [1.2394, -4.5894, 6.6175] # array
101
       that contains the hand calculated values for the
           # above specification
102
103
            filter_sos_calculated_values = getSOSCoefficients(cutoff_freq,
104
       filter_order, filter_type,
                                                                test_input)
105
       # array contains the sos influenced IIR filter
            # output values for the same specification
106
107
            self.assertEqual(hand_calculated_values,
108
                              filter_sos_calculated_values) # compare both
109
    → arrays to check if they aer similar
110
   try: # BF 20/12/21 - check if i am being run in the ipython shell
111
        IPYTHON
112
   except:
113
       __IPYTHON__ = False
114
115
   if not __IPYTHON__:
116
       if __name__ == '__main__':
117
            unittest.main()
118
   else: # BF 20/21/21 - This code allows the unit test to run in the
119
    → IPyhton shell
       unittest.main(argv=['first-arg-is-ignored'], exit=False)
120
```

8.2.3 calcAngles.py

```
1 #!/usr/bin/env ipython
  import unittest
  import numpy as np
  import numpy.linalg as ln
6
   def calcAngles(vec, DoDebug=False):
7
       """returns the angles of a 3d vector relative to the orthoginal
       unit vectors"""
       angle = np.zeros(3)
9
       referenceFrame = []
10
       referenceFrame.append([1.0,0.0,0.0])
11
       referenceFrame.append([0.0,1.0,0.0])
       referenceFrame.append([0.0,0.0,1.0])
13
       for i in range(3):
14
           if DoDebug:
15
               print(f"unitVec{i} = {referenceFrame[i]}")
16
```

```
part1=np.dot(vec,referenceFrame[i])/
17
       (ln.norm(vec)*ln.norm(referenceFrame[i]))
           angle1 = np.arccos(part1)
18
           angle[i] = np.rad2deg(angle1)
19
          #angle1 = np.arcsin(abs(vec[i])/np.linalg.norm(vec))
20
           #angle[i] = 90 - np.rad2deg(angle1)
           if DoDebug:
22
               print(f"angle[{i}] = {angle[i]}")
23
       return angle
24
25
26
  27
  testVec = [99.1, 99.1, 99.1]
  class TestCalcAngles(unittest.TestCase):
30
      def test1(self):
31
           testvec = [1,1,1]
32
           angles = calcAngles(testVec)
33
           didPass =
34
      np.testing.assert_array_almost_equal([54,54,54],angles)
          #print(didPass)
36
37
38
  try: # BF 20/12/21 - check if i am being run in the ipython shell
39
       __IPYTHON__
40
  except:
41
       __IPYTHON__ = False
42
43
  if not IPYTHON :
44
       if __name__ == '__main__':
45
           unittest.main()
46
  else: # BF 20/21/21 - This code allows the unit test to run in the
47
   → IPyhton shell
       unittest.main(argv=['first-arg-is-ignored'], exit=False)
```

8.2.4 realtime iir main.py

```
1 """ Modules """
2 from pyfirmata2 import Arduino
3 from scipy import signal
4 import numpy as np
5
6 import realtime_plot as rtp
7 import iir_filter as iir
8
9 """ Constants """
10 # Arduino Sample Rate
```

```
# BF 3/01/22 - changed to "true" sample rate
   fs = 650
    → Max 1000
  # Nyquist
  fn = fs / 2
13
  fc = 5
14
   """ IIR Filter Design """
16
   # Noise Removal
17
   sos = signal.bessel(1, fc / fn, "lowpass", output="sos", norm="mag") #
   → BF 3/01/22 - changed to bessel
19
  x_filter = iir.IIR_filter(sos)
   y_filter = iir.IIR_filter(sos)
  z_filter = iir.IIR_filter(sos)
23
   """ Real Time Plotters """
24
   sample_plot = rtp.RealtimePlots(fs, 2, ["X", "Y", "Z"],
   \rightarrow sample_limits=[-0.25,0.25], channels=3)
   orientation_plot = rtp.RealtimeVectorPlot()
26
27
   """ Sample Process Function """
28
   def addX(data):
29
       # Zeroing from measurements
30
       data -= 0.335
31
       f = x_filter.filter(data)
32
33
       sample_plot.addSample(data, f, channel=0)
34
       orientation_plot.addSample(f, channel=0)
36
   def addY(data):
37
       # Zeroing from measurements
38
       data -= 0.340
39
       f = y_filter.filter(data)
40
41
       sample_plot.addSample(data, f, channel=1)
42
       orientation_plot.addSample(f, channel=1)
43
44
   def addZ(data):
45
       # Zeroing from measurements
46
       data -= 0.340
47
       f = z_filter.filter(data)
48
49
       sample_plot.addSample(data, f, channel=2)
       orientation_plot.addSample(f, channel=2)
51
   """ Aurdino Data Aquisition """
53
   board = Arduino(Arduino.AUTODETECT)
54
   board.samplingOn(1000/fs)
55
   board.analog[0].register_callback(addX)
```

```
board.analog[0].enable_reporting()
board.analog[1].register_callback(addY)
board.analog[1].enable_reporting()
board.analog[2].register_callback(addZ)
board.analog[2].enable_reporting()

""" Show Real Time Plots """
rtp.plt.show()
board.exit()
```

8.2.5 realtime plot.py

```
from mpl_toolkits.mplot3d import Axes3D
  from matplotlib.animation import FuncAnimation
  from matplotlib.widgets import Button
  import matplotlib.pyplot as plt
  import numpy as np
   import time
   from calcAngles import calcAngles
9
10
   """ Rolling Buffer """
11
  class RollingBuffer:
12
       def __init__(self, size):
13
           self.size = size
           self.plot_buffer = np.zeros(self.size)
15
           self.data_buffer = []
16
17
       def update(self):
18
           self.plot_buffer = np.append(self.plot_buffer,
19
       self.data_buffer)
           self.plot_buffer = self.plot_buffer[-self.size:]
           self.data_buffer = []
21
           return self.plot_buffer
22
23
       def add(self, v):
24
           self.data_buffer.append(v)
25
26
   """ Real-Time Plotter Variable Channel """
27
   class RealtimePlots:
28
       def __init__(self, fs, window_time, labels, sample_limits=[-0.5,
29
       0.5], channels=1):
           # Buffer
30
           self.buffer_size = fs * window_time
31
           self.r_buffers = []
32
           self.f_buffers = []
33
34
           # Figure Plot
35
```

```
self.fig, self.ax = plt.subplots(nrows=2)
36
37
           # Sample Plot
38
           self.ax[0].plot([0,
39
       self.buffer_size-1],[0,0],"r--",label="Zero")
           self.ax[0].set_ylim(sample_limits[0], sample_limits[1])
40
           self.ax[0].set_title("Un-Filtered")
41
           self.r_lines = []
42
43
           # Filter Plot
44
           self.ax[1].plot([0,
45
       self.buffer_size-1],[0,0],"r--",label="Zero")
           self.ax[1].set_ylim(sample_limits[0], sample_limits[1])
46
           self.ax[1].set_title("Filtered")
47
           self.f_lines = []
48
49
           # Plot Buffers
50
           for i in range(channels):
51
                self.r_buffers.append(RollingBuffer(self.buffer_size))
52
                line, = self.ax[0].plot(self.r_buffers[i].update(),
53
       label=labels[i])
                self.r_lines.append(line)
54
55
                self.f_buffers.append(RollingBuffer(self.buffer_size))
56
                line, = self.ax[1].plot(self.f_buffers[i].update(),
57
       label=labels[i])
                self.f_lines.append(line)
58
           self.ax[1].legend(loc=4)
60
           self.anim = FuncAnimation(self.fig, self.update, interval=100)
61
           self.update_count = 0
62
63
           # Sample Rate
64
           self.sample_count = 0
65
           self.label = self.ax[0].text(0, sample_limits[0], "Sample
       Rate: -", ha="left", va="bottom", fontsize=15)
           self.last = 0
67
68
       def update(self, x):
69
           # Buffer
70
           for i in range(len(self.r_lines)):
71
                self.r_lines[i].set_ydata(self.r_buffers[i].update())
72
                self.f_lines[i].set_ydata(self.f_buffers[i].update())
73
74
           # Sample Rate Calc
75
           if self.update_count % 5 == 0: # Reduce updates for
76
      performance
                current_time = time.time()
77
```

```
sample_rate = self.sample_count / (current_time -
78
        self.last)
                self.last = current time
79
                self.sample_count = 0
80
81
                self.label.set_text(f"Fs: {sample_rate:.1f}Hz")
            self.update_count += 1
83
84
        def addSample(self, v, f, channel=0):
85
            if channel == 0:
86
                self.sample_count += 1
87
            self.r_buffers[channel].add(v)
            self.f_buffers[channel].add(f)
90
   """ Real-Time Vector Plotter """
91
   class RealtimeVectorPlot:
92
        def __init__(self):
93
            self.vec = np.zeros(3)
94
95
            self.fig, self.ax =
        plt.subplots(subplot_kw=dict(projection="3d"))
            self.ax.set_xlim(-1.0, 1.0)
97
            self.ax.set_ylim(-1.0, 1.0)
98
            self.ax.set_zlim(-1.0, 1.0)
99
            self.ax.set_xticks([])
100
            self.ax.set_yticks([])
101
            self.ax.set_zticks([])
102
103
            self.anim = FuncAnimation(self.fig, self.update, interval=100)
104
            self.q = self.ax.quiver(0, 0, 0, 0, 0, 1)
105
            self.x = self.ax.quiver(0, 0, 0, 1, 0, 0, color="red")
106
            self.y = self.ax.quiver(0, 0, 0, 0, 1, 0, color="green")
107
            self.z = self.ax.quiver(0, 0, 0, 0, 0, 1, color="blue")
108
109
            self.label = self.ax.text(0, 0, 0, "test",
110
        transform=self.ax.transAxes)
            self.update delay = 0
111
112
            # For calibration
113
            self.offset_angles = np.zeros(3)
114
            axes = plt.axes([0.81, 0.05, 0.1, 0.075])
115
            self.button = Button(axes, "Calibrate")
116
            self.button.on_clicked(self.setOffset)
117
118
        def update(self, x):
119
            m = np.sqrt(self.vec[0] * self.vec[0] + self.vec[1] *
120
       self.vec[1] + self.vec[2] * self.vec[2])
            if m != 0:
121
                self.q.remove()
122
```

```
self.q = self.ax.quiver(0, 0, 0, self.vec[0] / m,
123
       self.vec[1] / m, self.vec[2] / m, color="black")
124
                # Angle Calcs
125
                if self.update_delay % 5 == 0:
126
                    # Angle between the vector and the x plane
127
                    angle = calcAngles(self.vec)
128
                    ma = f"Measured Angles (deg), x: {angle[0]:.1f}, y:
129
       {angle[1]:.1f}, z: {angle[2]:.1f}\n"
                    angle = np.abs(angle - self.offset_angles)
130
                    fa = f"Final Angle (deg): {angle[2]:.1f}"
131
                    self.label.set_text(ma + fa)
132
                self.update_delay += 1
134
        def setOffset(self, x):
135
            self.offset_angles = calcAngles(self.vec)
136
137
       def addSample(self, v, channel=0):
138
            self.vec[channel] = v
139
```