Team Project: Octave Band Filtering

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5.1 Octave Bands

This bit was taken care of in Matlab and this is the function to set up the filter bands table. The results are below and a copy of the code is included in appendix A: Since we have a limit on

val [units]	O_1	O_2	O_3	O_4	O_5	O_6	O_7	O_8
Lower (Hz)	32.703	65.406	130.81	261.63	523.25	1046.5	2093	4186
Lower (Rad)	0.025685	0.05137	0.10274	0.20548	0.41096	0.82192	1.6438	3.2877
Upper (Hz)	61.735	123.47	246.94	493.88	987.77	1975.5	3951.1	7902.1
Upper (Rad)	0.048487	0.096974	0.19395	0.3879	0.77579	1.5516	3.1032	6.2063
Center (Hz)	47.219	94.439	188.88	377.75	755.51	1511	3022	6044.1
Center (Rad)	0.037086	0.074172	0.14834	0.29669	0.59338	1.1868	2.3735	4.747

Table 1: Frequency Ranges for Octaves 1 to 8 starting with O_1

the bands we can recognize that arises from the use of the sampling freq of 8kHz, we can only obtain unique detection for the set of Octaves whose frequencies are below the Nyquist rate of $\frac{fs}{2}$ or 4kHz. Or, O_7 in the table above.

5.2 Octave Filter Bank

Comment on the selectivity of the bandpass filters, i.e., use the frequency response (passbands and stopbands) to explain how the filter passes one octave while rejecting the others. Are the filter's passbands narrow enough so that only one octave lies in the passband and the others are in the stop-band?

Comment here and then do the next bit. need a plot from 5.2 that Tyler made. our beta is 1 The band pass filter bank was put together in python and has the added benefit of being more portable than the use of Matlab for this application. These are the results from the creation of the x signal needed to pass on to the filter bank. Defining the expression below:

$$x(t) = \begin{cases} \cos(2\pi \cdot 220t) & 0.0 < t \le 0.25, \\ \cos(2\pi \cdot 880t) & 0.3 < t \le 0.55, \\ \cos(2\pi \cdot 440t) + \cos(2\pi \cdot 880t) & 0.60 < t \le 0.85 \\ 0 & \text{otherwise} \end{cases}$$

and now the code to do this along with the plot showing the data is what we expect:

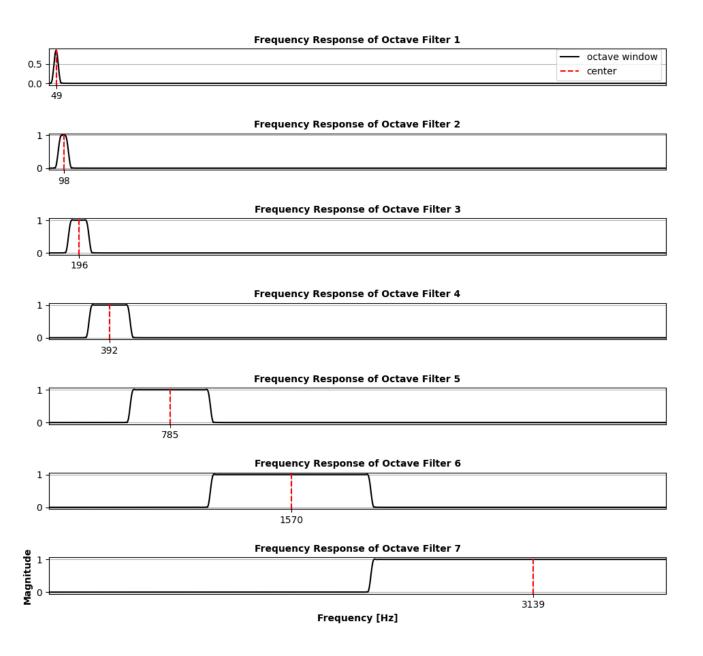


Figure 1: plot for all BPF's

The filters above in fig. 1 begin at a center of 47Hz but we only need to worry about the subsequent ones that we were asked for. Aligned at their centers is a vertical line and a tick mark indicating the center frequency. this bank of filters is what we will use to pass the time data created in the equation above.

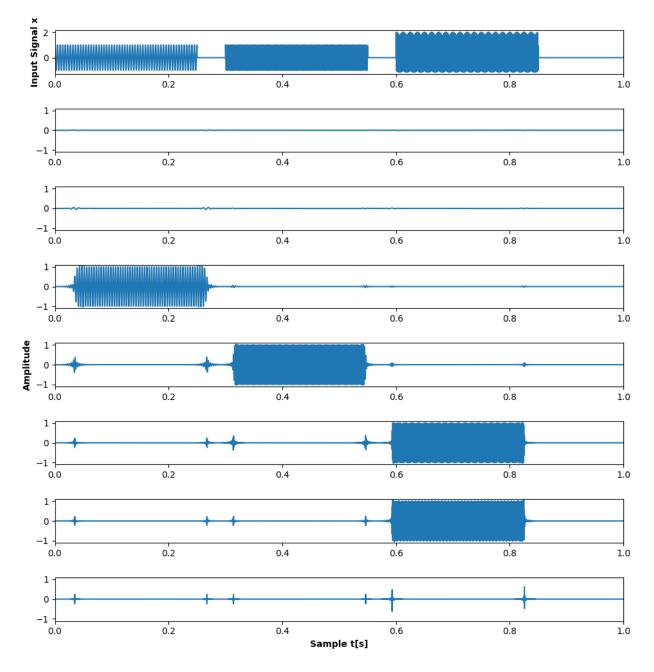


Figure 2: plot for all BPF's processing the time data at the top

Filters one and two do not pass anything since they are both too low to capture the first freq of 220Hz, but the third has a center of 189Hz, meaning it does capture the first frequency. In all cases, we can see that the transition points have the characteristic peaks associated with instant changes in frequency. Any instant change results in a sharp change which corresponds to an infinite number of frequencies. the third filter is capable of capturing the 880Hz, the fourth captures the 1760Hz in superposition with the 880Hz, and no signal is passed through the last except the transients that appear in all filters.

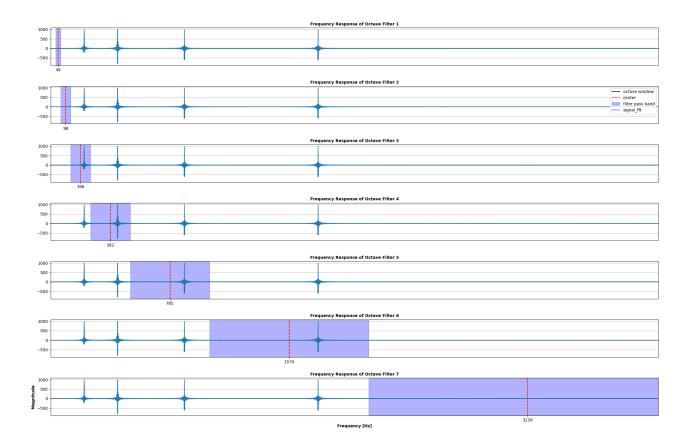


Figure 3: plot for all BPF's bandpass regions

In this figure, it is easier to see that the band-pass filters correctly capture the frequencies that we see in the time plot of fig. 2. In the figure above fig. 3, we can see that the filters that do not capture any signal do not have an overlapping note peak in the shaded region. However, the ones that we see with a signal passed through them do have a corresponding note in the shaded band-pass region.

0.1 Creating the FIR Filters

To create the FIR filters, we used a python class shown in subsection 0.4. The class works by first defining N, the number of points to be used in the FIR filter. We then declare the pass band region, the frequencies we wish for the FIR filter to pass through while the others are rejected. This is shown in fig. ??. Effectively, we are specifying the desired frequency response of the FIR filter but only using N number of points to define the filter in the frequency domain.

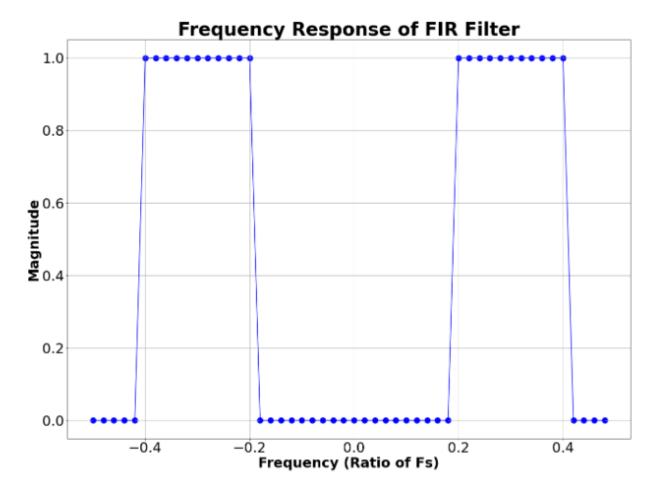


Figure 4: Band-pass region of the FIR filter being created. The filter response is specified by N number of points where N is the length of the FIR filter being created.

After defining the frequency response with N number of points, we need to also ensure that the frequency response is symmetric about the y-axis. This ensures that our FIR filter will be purely real. To obtain the time domain FIR filter, we then take the inverse fourier transform of the N number of points specifying the frequency response. The FIR filter corresponding to the frequency response shown in 4 above.

Appendix A

0.2 Matlab Code

```
function print_octaves(n,fs)
2 % n should be a range relative to A4. ex: -1:1 gives A3, A4, A5
3 A4 = 440;
4 C4 = A4.*2.^{(-9./12)};
B4 = A4.*2.^(2./12);
6 Octaves = 2.^n;
7 \text{ Cs} = C4.*Octaves;
8 Bs = B4.*Octaves;
n_range = 0:length(n)-1;
10 Centers = (Cs + Bs)./2;
cell(size(Centers));
12 octave_array = arrayfun(@(x) sprintf('Octave %d', x), n_range, 'UniformOutput',
     false);
w_Cs = Cs.*2.*pi./fs;
w_Bs = Bs.*2.*pi./fs;
u_Centers = Centers.*2.*pi./fs;
16 rows = {'Lower (Hz)','Lower (Rad)','Upper (Hz)','Upper (Rad)','Center (Hz)','Center
      (Rad)'};
17 % Summarize data in a table
18 T = array2table([Cs; w_Cs; Bs; w_Bs; Centers; w_Centers],'VariableNames',
      octave_array,'RowName',rows);
19 disp(T)
20 disp('Hz are not normalized.')
21 disp('Radians are normalized by sampling frequency.')
```

0.3 Python Code

```
1 from util.OctaveBandFilt import OctaveBandFilter, ourOctaveBandFilter
2 from util.FIR_filter import FIRFilter
3 import numpy as np
4 import matplotlib.pyplot as plt
5 from rich import print
6 import math
8 \text{ fontsize} = 20
10 def calculate_grid_dimensions(n):
      columns = round(math.sqrt(n))
12
      rows = math.ceil(n / columns)
13
      return rows, columns
14
15
16 def calculate_octave_ranges(base_freq, num_octaves, sample_rate):
      octave_ranges = []
17
      for i in range(num_octaves+1):
18
          fmin = base_freq * (2 ** i) # Minimum frequency for the octave
19
          fmax = base\_freq * (2 ** ((i*12+11)/12)) # Maximum frequency for the
20
      octave
21
           # Normalize the frequencies
23
          fmin_normalized = fmin
24
          fmax_normalized = fmax
25
          octave_ranges.append((fmin_normalized, fmax_normalized))
26
      return octave_ranges
```

```
28
29
30 def windowed_sinc(filter_length, low_freq, high_freq):
      n = np.arange(filter_length)
3.1
      mid = (filter\_length - 1) / 2
32
      window = np.hamming(filter_length)
33
34
      # Debugging: Print frequency values
      print("low_freq:", low_freq)
37
      print("high_freq:", high_freq)
38
39
      sinc_low = np.sinc(2 * low_freq * (n - mid))
      sinc_high = np.sinc(2 * high_freq * (n - mid))
40
      bp_filter = sinc_high - sinc_low
41
42
      # Debugging: Print sinc function outputs
43
      print("sinc_low sample values:", sinc_low[:5])
44
45
      print("sinc_high sample values:", sinc_high[:5])
46
      bp_filter *= window
47
      bp_filter_sum = np.sum(bp_filter)
48
49
      # Debugging: Check for zero sum
51
      print("bp_filter sum after window:", bp_filter_sum)
52
      if bp_filter_sum == 0:
          bp_filter_sum = 1 # To avoid division by zero
53
54
      bp_filter /= bp_filter_sum
55
      # Debugging: Final check
      print("Final bp_filter sample values:", bp_filter[:5])
      return bp_filter
59
60
61
62 # Example usage
63 filter_length = 150 # A typical length for FIR filters
64 center_frequency = 220 # Center frequency in Hz
65 sampling_frequency = 12000 # Sampling frequency in Hz
67 # Recreating the OctaveBandFilter instance with the updated class definition
68 octave_filter = OctaveBandFilter(filter_length, center_frequency,
      sampling_frequency)
69 octave_filter.calculate_coefficients()
7.0
71
_{72} # Recreating the OctaveBandFilter instance with the updated class definition
73 ourOctave_filter = ourOctaveBandFilter(filter_length, center_frequency,
      sampling_frequency, windowed_sinc)
74 ourOctave_filter.calculate_coefficients()
75 print("Coefficients:", ourOctave_filter.coefficients)
^{76} # Generating a test signal - a simple sine wave at the center frequency
77 test_signal = np.sin(2 * np.pi * center_frequency / sampling_frequency * np.arange
      (0, sampling_frequency))
79 # Applying the filter to the test signal
80 filtered_signal = octave_filter.apply_filter(test_signal)
81 # Applying the filter to the test signal
82 ourFiltered_signal = ourOctave_filter.apply_filter(test_signal)
83
84
85 def generate_fm_signal(sampling_frequency, duration, start_freq, end_freq):
```

```
87
       Generate a frequency modulated (FM) signal.
88
       t = np.arange(0, duration, 1/sampling_frequency) # Time vector
89
       instantaneous_frequency = np.linspace(start_freq, end_freq, len(t))
90
       phase = 2 * np.pi * np.cumsum(instantaneous_frequency) / sampling_frequency
91
       signal = np.sin(phase)
92
       return signal
93
96 # Parameters for the OctaveBandFilter
97 filter_length = 150
98 center_frequency = 440 # Center frequency in Hz
99 sampling_frequency = 8000 # Sampling frequency in Hz
100 # Lambda function for windowed sinc
101
103
104 # Creating the filter instance and calculating coefficients
105 Octave_filter = OctaveBandFilter(filter_length, center_frequency,
      sampling_frequency)
106 Octave_filter.calculate_coefficients()
107 # Creating the filter instance and calculating coefficients
108 ourOctave_filter = ourOctaveBandFilter(filter_length, center_frequency,
       sampling_frequency, windowed_sinc)
ourOctave_filter.calculate_coefficients()
110
# Generating an FM signal
duration = 5 # 1 second duration
start_freq = 1 # Starting frequency of 0 Hz
114 end_freq = 4 * ourOctave_filter.center_frequency # Ending at twice the center
       frequency of the filter
115 fm_signal = generate_fm_signal(sampling_frequency, duration, start_freq, end_freq)
117 # Filtering the FM signal
filtered_fm_signal = Octave_filter.apply_filter(fm_signal)
# Filtering the FM signal
120 ourFiltered_fm_signal = ourOctave_filter.apply_filter(fm_signal)
122 # Sample rate and base frequency
123 base_freq = 32.703*1 # Starting frequency of the first octave
124 # Calculate octave ranges
125 num_octaves = 6
126 octave_ranges = calculate_octave_ranges(base_freq, num_octaves, 8000)
127
128
129 # Create a set of filters
130 N = 1000
131 filters = [FIRFilter(N, fmin=fmin, fmax=fmax, padding_factor=9, fs=8000) for fmin,
      fmax in octave_ranges]
132 num_filters = len(filters)
133 #filters = [FIRFilter(N, fmin=fmin*N, fmax=fmax*N, padding_factor=9) for fmin, fmax
       in octave_ranges]
134 print(f"octave ranges: {octave_ranges}")
135 # Calculate grid size
136 total_subplots = num_filters
rows, cols = calculate_grid_dimensions(total_subplots)
138 print(f"rows: {rows}")
print(f"cols: {cols}")
140 # Create a figure for plotting
141 fig = plt.figure(figsize=(22, 16))
mids = [(b+a)/2 \text{ for a, b in octave\_ranges}]
```

```
144 print (mids)
# Plot each filter's response
146 for i, filter in enumerate(filters):
      # Calculate position
147
      position = i * 2 + 1
                             # Position for frequency response plot
148
       x = mids[i]
149
       filter.plot_filter(fig, num_filters+1, 1, i+1)
150
       plt.axvline(x=x, ymin=0, ymax=1, color='r', linestyle='--')
151
       plt.legend(['octave window', 'center'])
       plt.xticks([x])
       ax = plt.gca()
155
       ax.set_xlim([0, filter.fs/2])
156
plt.tight_layout()
plt.show(block=False)
plt.savefig("freq_data_2.png", dpi=300, transparent=False)
160 plt.close()
161 num_octaves = 6
162 octave_ranges = calculate_octave_ranges(base_freq, num_octaves, 1000)
163 filters = [FIRFilter(N, fmin=fmin, fmax=fmax, padding_factor=9, fs=8000) for fmin,
      fmax in octave_ranges]
164 num_filters = len(filters)
165 # Generate the signal 5.2
166 \text{ fs} = 8000
t1 = np.linspace(0, 0.25, int(fs\star0.25), endpoint=False)
t2 = np.linspace(0.3, 0.55, int(fs\star0.25), endpoint=False)
t3 = np.linspace(0.6, 0.85, int(fs\star0.25), endpoint=False)
t_end = np.linspace(0.85, 1, int(fs*0.15), endpoint=False)
171
x1 = np.cos(2*np.pi*220*t1)
x2 = np.cos(2*np.pi*440*t2)
x3 = np.cos(2*np.pi*880*t3) + np.cos(2*np.pi*1760*t3)
zero_padding = np.zeros(int(fs*0.05))
zeros_end = np.zeros(int(fs*0.15))
x = \text{np.concatenate((x1, zero_padding, x2, zero_padding, x3, zeros_end))}
179 # Filter and plot the signal
180 fig = plt.figure(figsize=(20, 20))
181 plt.title('Filtered Signal with Filters', fontsize=fontsize+10, fontweight='bold')
182 plt.subplot(len(filters)+1, 1, 1)
183 plt.plot(x)
184 plt.ylabel('Input Signal x', fontsize=fontsize, fontweight='bold')
185
186 ax.set_xlim([0, fs])
187 for i, filter in enumerate(filters):
      filtered_x = filter.process(x)
188
       filtered_sig = (filtered_x)
189
      plt.subplot(len(filters)+1, 1, i+2)
190
       plt.plot(np.real(filtered_sig))
191
192
       ax = plt.gca()
       ax.set_ylim([-1.100, 1.100])
193
       ax.set_xlim([0, fs])
194
       plt.tight_layout()
       plt.ylabel('Amplitude', fontsize=fontsize, fontweight='bold') if i == (len(
196
       filters)//2) else None
197
199 plt.xlabel('Sample t[s]', fontsize=fontsize, fontweight='bold')
200 plt.show(block=False)
plt.savefig("time_data.png", dpi=300, transparent=False)
202 plt.close()
```

```
204
205 fig = plt.figure(figsize=(22, 16))
207 \text{ mids} = [(b+a)/2 \text{ for a, b in octave\_ranges}]
208 x_{fftd} = np.fft.fft(x)
209
210 # Plot each filter's response
211 for i, filter in enumerate(filters):
       # Calculate position
       position = i * 2 + 1 # Position for frequency response plot
214
       x = mids[i]
215
       # Determine the width of the rectangle from the octave range
216
       fmin, fmax = octave_ranges[i]
217
       rect_width = fmax - fmin
218
219
       filter.plot_filter(fig, num_filters+1, 1, i+1)
220
221
       plt.axvline(x=x, ymin=0, ymax=1, color='r', linestyle='--')
222
       # Adding the shaded rectangle
223
       plt.axvspan(x - rect_width/2, x + rect_width/2, ymin=0, ymax=1, alpha=0.3,
       color='blue')
225
226
      plt.plot(x_fftd)
       plt.legend(['octave window', 'center', 'filter pass band', 'signal_fft'], loc='
227
      upper right', bbox_to_anchor=(1, 1)) if i == 1 else None
      plt.xticks([x])
228
       ax = plt.gca()
229
       ax.set_xlim([0, filter.fs/2])
230
       plt.tight_layout()
232
234 plt.show(block=False)
plt.savefig("fft_freqdata.png", dpi=300, transparent=False)
236 input()
```

0.4 Python FIR Filter Class

```
import numpy as np
2 from numpy import zeros, append
3 from numpy.fft import fftshift, fft
4 import matplotlib.pyplot as plt
  class FIRFilter:
      def __init__(self, N=10000, fmin=3, fmax=7, padding_factor=9, fs=8000):
           self.N = N
9
          self.padding_factor = padding_factor
1.0
          self.fs = fs # Sampling rate
          self.H = zeros(N)
          self.w = zeros(N)
1.3
          self.pos = np.arange(N)
14
          self.fmin = fmin*self.N/self.fs
15
          self.fmax = fmax*self.N/self.fs
16
17
          self.h = None
18
          self.h_pad = None
19
          self.H_pad = None
20
          self.w_pad = None
21
          self.h_ham = None
22
          self.H_ham_pad = None
23
```

```
24
25
          self.create_filter()
           self.apply_padding()
26
          self.apply_hamming_window()
27
28
29
      def create_filter(self):
30
31
           k = np.arange(-int(self.N/2), int(self.N/2))
           self.w = k * self.fs / self.N # Adjusted to use actual frequency values
33
           self.H = np.where((np.abs(k) >= self.fmin) & (np.abs(k) <= self.fmax), 1,
      0)
34
           self.h = fftshift(fft(fftshift(self.H)))
3.5
      def apply_padding(self):
36
          NP = self.N + self.padding_factor * self.N
37
          self.h_pad = append(self.h, zeros(self.padding_factor * self.N))
38
          self.H_pad = fftshift(fft(self.h_pad)) / self.N
39
40
           k = np.arange(-NP/2, NP/2)
          self.w_pad = k * self.fs / NP
41
42
      def apply_hamming_window(self):
43
          self.h_ham = self.h * 0.5 * (1 + np.cos(2 * np.pi * (self.pos - self.N / 2))
44
       / self.N))
45
           self.h_ham_pad = append(self.h_ham, zeros(self.padding_factor * self.N))
46
           self.H_ham_pad = fftshift(fft(self.h_ham_pad)) / self.N
47
      def process(self, input_data):
48
          return np.convolve(input_data, self.h_ham)/self.N
49
      def plot_filter(self, fig, row, col, pos):
51
           # Frequency Response Plot
52
          ax1 = fig.add_subplot(row, col, pos)
53
           # ax1.scatter(self.w, self.H.real, c='b', s=150)
54
           # ax1.plot(self.w_pad, abs(self.H_pad), 'r')
55
          ax1.plot(self.w_pad, abs(self.H_ham_pad), 'black')
56
           \#ax1.set_xlim(0, .5)
57
          ax1.set_title(f'Frequency Response of Octave Filter {pos}', fontsize=5,
58
      fontweight='bold')
          if pos == row-1:
59
               ax1.legend(['Hamming'], prop={'size': 5})
               ax1.set_xlabel('Frequency (Ratio of Fs)', fontsize=5, fontweight='bold'
61
               ax1.set_ylabel('Magnitude', fontsize=5, fontweight='bold')
62
63
          ax1.grid(True)
64
      def plot_filter1(self):
65
           # MatPlotLib plotting
66
67
          fig = plt.figure(figsize=(22, 16))
68
69
           # Frequency Response Plot
           ax1 = fig.add_subplot(211)
           ax1.scatter(self.w, self.H.real, c='b', s=150)
           ax1.plot(self.w_pad, abs(self.H_pad), 'r')
           ax1.plot(self.w_pad, abs(self.H_ham_pad), 'black')
73
           ax1.set_xlabel('Frequency (Hz)', fontsize=15, fontweight='bold')
74
          ax1.set_ylabel('Magnitude', fontsize=15, fontweight='bold')
          ax1.set_title('Frequency Response of FIR Filter', fontsize=15, fontweight='
76
      bold')
77
          ax1.legend(['Ideal', 'Actual', 'Hamming'], prop={'size': 15})
78
          ax1.tick_params(axis='both', labelsize=15)
79
          ax1.grid(True)
```

```
# Time Domain Plot
81
          ax2 = fig.add_subplot(212)
82
          ax2.vlines(self.pos, 0, self.h.real, 'b')
83
          # ax2.vlines(self.pos, 0, self.h.imag, 'r')
84
          ax2.scatter(self.pos, self.h.real, c='b', s=150)
85
          \# ax2.scatter(self.pos, self.h.imag, c='r', s=150)
86
87
          ax2.set_xlabel('Position', fontsize=15, fontweight='bold')
88
          ax2.set_ylabel('Value (Unscaled)', fontsize=15, fontweight='bold')
          ax2.set_title('Time Domain FIR Filter', fontsize=15, fontweight='bold')
          # ax2.legend(['Real', 'Imag'], prop={'size': 15})
          ax2.tick_params(axis='both', labelsize=15)
          ax2.grid(True)
92
93
          plt.show(block=False)
94
```