

Auburn Drumline

Statistical Analysis for Marching Band Drum Core Improvement

Daniel Willis



Bio

- * 15 Year old freshman in High School
- * I'm in the Marching Band and Concert Band
- Playing drums for 5 years
- * I also play rock N Roll Drums with my father
- * I have been to SCALE 5 times and spoke the past 3 years
- Professionally speaking since I was 12
- Spoken at Dockercon, O'Reilly Velocity and multiple Devopsdays

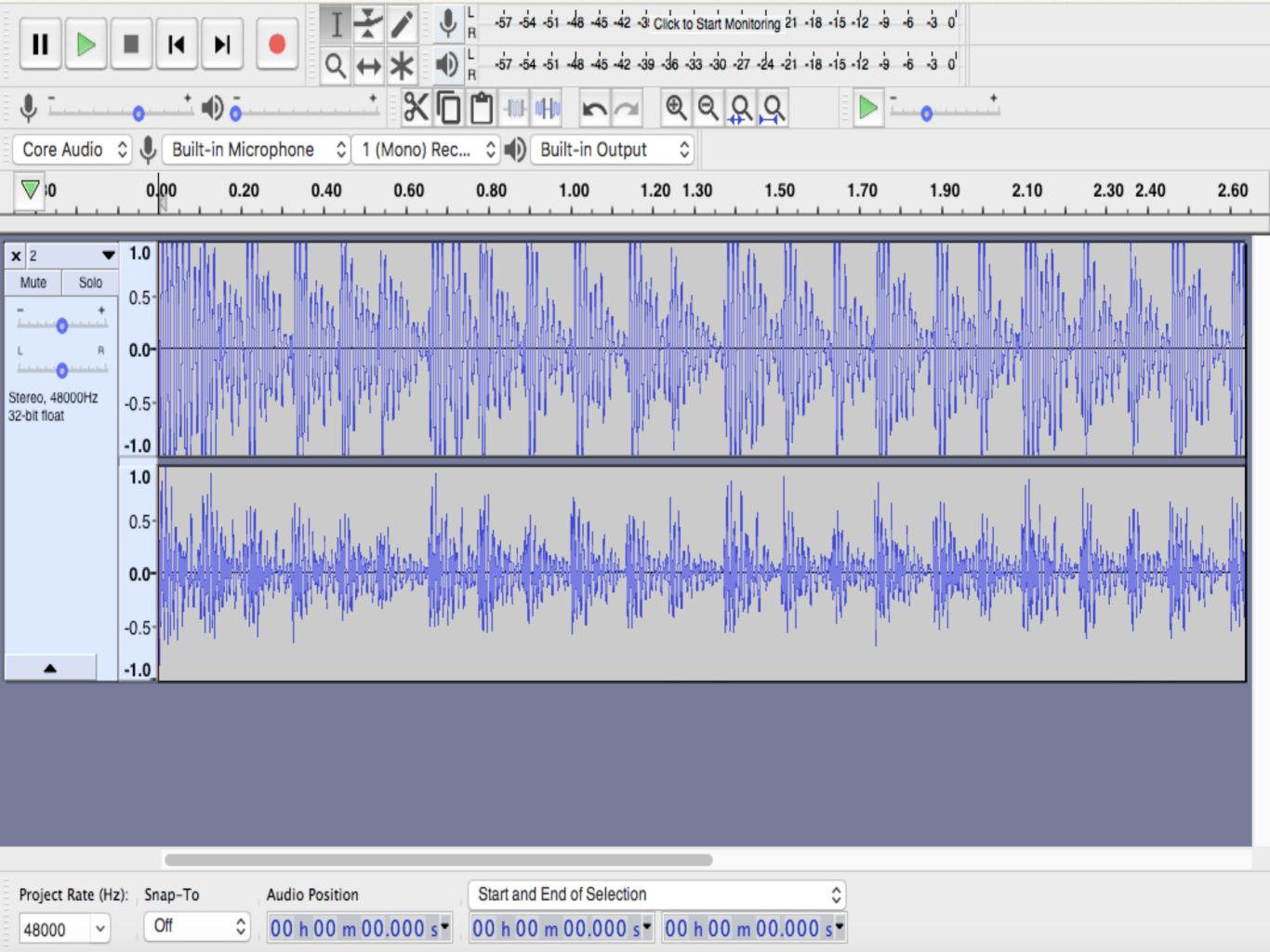
Outline

- * The Idea
- Demonstration (recording a diddle exercise)
- * Audacity
- * History
- Digital Signal Analysis
- Statistical Analysis (Improvement)
- Summary (What I learned)

The Idea

- * Record the same diddle sequence over a three month period.
- * Use audio tools to extract timings, tempo and beat signals.
- * Use statistical tools to look for patterns of improvement over time.

base scale



base scale



History

Joseph Fourier

From Wikipedia, the free encyclopedia

For the French socialist philosopher, see Charles Fourier.

Jean-Baptiste Joseph Fourier (/fʊəriˌeɪ, -iər/;^[1] French: [fuʁje]; 21 March 1768 – 16 May 1830) was a French mathematician and physicist born in Auxerre and best known for initiating the investigation of Fourier series and their applications to problems of heat transfer and vibrations. The Fourier transform and Fourier's law are also named in his honour. Fourier is also generally credited with the discovery of the greenhouse effect.^[2]

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- 1 Biography
- 2 The Analytic Theory of Heat
- 3 Determinate equations
- 4 Discovery of the greenhouse effect
- 5 Works
- 6 See also
- 7 References
- 8 Further reading
- 9 External links

Biography [edit]

Fourier was born at Auxerre (now in the Yonne département of France), the son of a tailor. He was orphaned at age nine. Fourier was recommended to the Bishop of Auxerre and, through this introduction, he was educated by the Benedictine Order of the Convent of St. Mark. The commissions in the scientific corps of the army were reserved for those of good birth, and being thus ineligible, he accepted a military lectureship on mathematics. He took a prominent part in his own district in promoting the French Revolution, serving on the local Revolutionary Committee. He was imprisoned briefly during the Terror but, in 1795, was appointed to the École Normale and subsequently succeeded Joseph-Louis Lagrange at the École Polytechnique.

Fourier accompanied Napoleon Bonaparte on his Egyptian expedition in 1798, as scientific adviser, and was appointed secretary of the Institut d'Égypte. Cut off from France by the British fleet, he organized the workshops on which the French army had to rely for their munitions of war. He also contributed several mathematical papers to the Egyptian Institute (also called the Cairo Institute) which Napoleon founded at Cairo, with a view of weakening British influence in the East. After the British victories and the capitulation of the French under General Menou in 1801, Fourier returned to



Jean-Baptiste Joseph Fourier

Born 21 March 1768

Auxerre, Burgundy, Kingdom of France (now in Yonne, France)

Died 16 May 1830 (aged 62)

Paris, Kingdom of France

Residence France

Nationality French

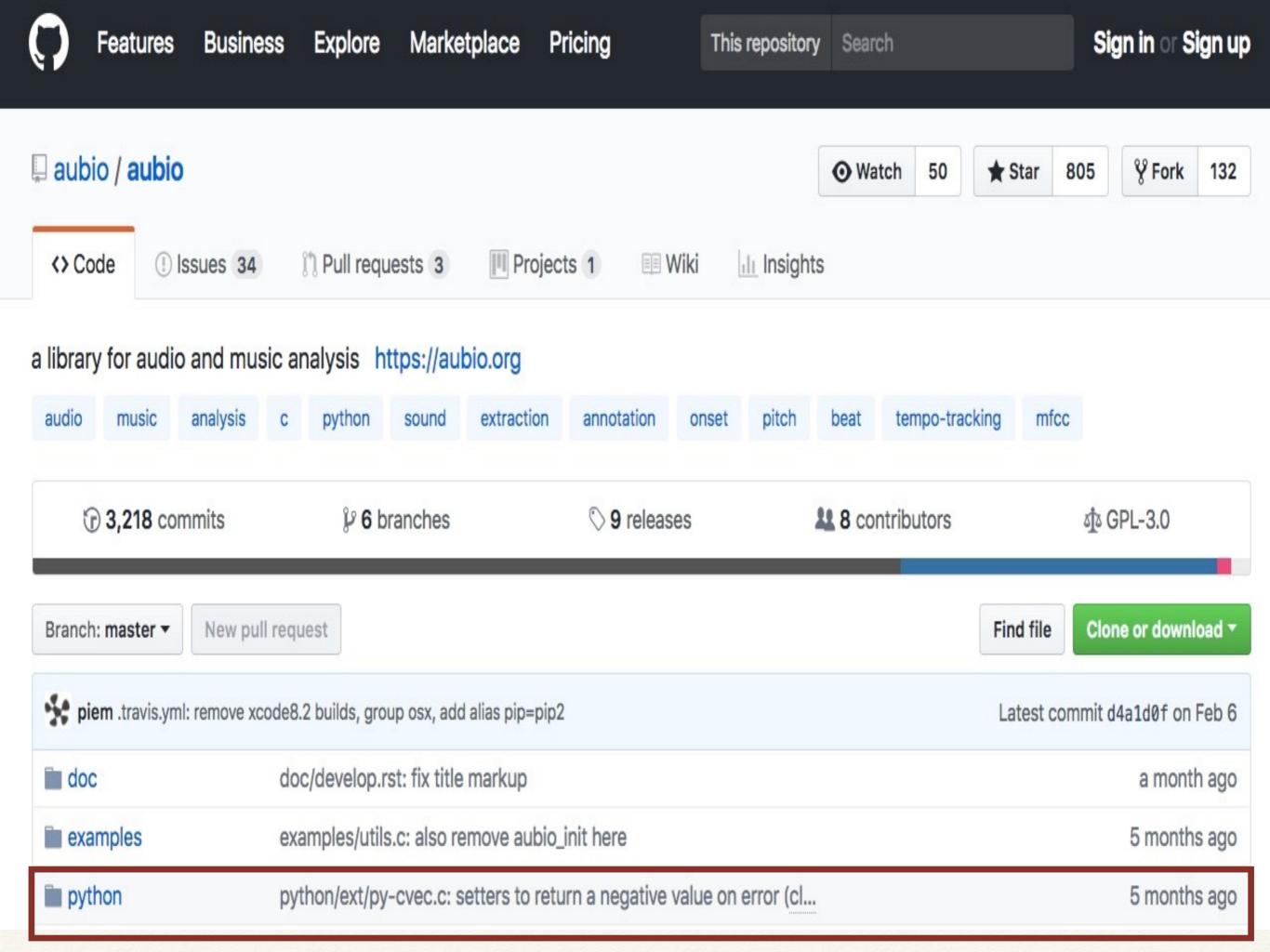
Alma mater École Normale

Known for Fourier series

Fourier transform

Considered the Father of Digital Music

Digital Signal Analysis



```
## Statistical Analysis as a Tool for Marching Band Drum Core
   Practice Improvement
   **Objective**
   I have been recording my snare practice sessions for the past 6 months. I am attempting to
   do statistical analysis of the digital signals of the recordings looking at attributes such
   as amplitude pitch, phase, and tempo. The goal is to be able to identify improvements using
   statistical tools like "R". This is a great presentation for students.
   **Prerequisites**
   Python 2.7.10,
   pip 9.0.1,
   Aubio 0.4.6,
   git version 2.11.0 (Apple Git-81)
12 #### Installation Prep
   Install Aubio:
    'sudo easy_install pip'
    'sudo pip install aubio'
    'python -c "import aubio; print(aubio.version)"
   Clone the Pyhton Samples
    `git clone https://github.com/aubio/aubio.git`
   #### Refference Sites
   https://github.com/aubio/aubio
   http://whatis.techtarget.com/definition/Nyquist-Theorem
   https://www.sweetwater.com/insync/7-things-about-sample-rate/
```

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Statistical Analysis as a Tool for Marching Band Drum Core Practice Improvement

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https://manual.audacityteam.org/man/glossary.html

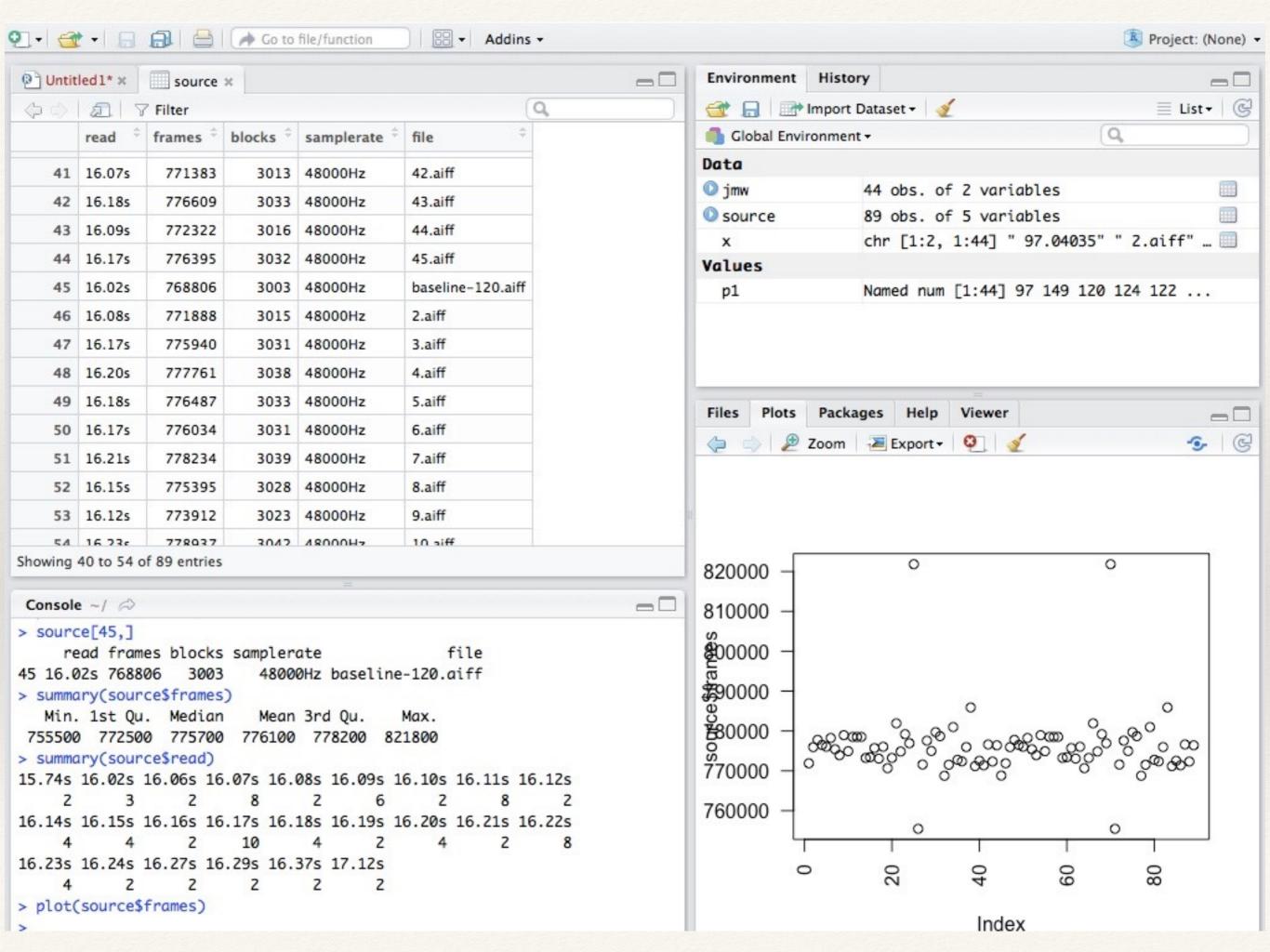
```
import sys
from aubio import source
if __name__ == '__main__':
    if len(sys.argv) < 2:
        print('usage: %s <inputfile> [samplerate] [hop_size]' % sys.argv[0])
        sys.exit(1)
    samplerate = 0
    hop_size = 256
    if len(sys.argv) > 2: samplerate = int(sys.argv[2])
    if len(sys.argv) > 3: hop_size = int(sys.argv[3])
    f = source(sys.argv[1], samplerate, hop_size)
    samplerate = f.samplerate
    total_frames, read = 0, f.hop_size
    while read:
       vec, read = f()
       total_frames += read
       if read < f.hop_size: break
    outstr = "%.2fs" % (total_frames / float(samplerate))
    outstr += ",%d" % total_frames
    outstr += ",%d" % (total_frames // f.hop_size)
    outstr += ",%dHz" % f.samplerate
    outstr += "," + f.uri
    print(outstr)
```

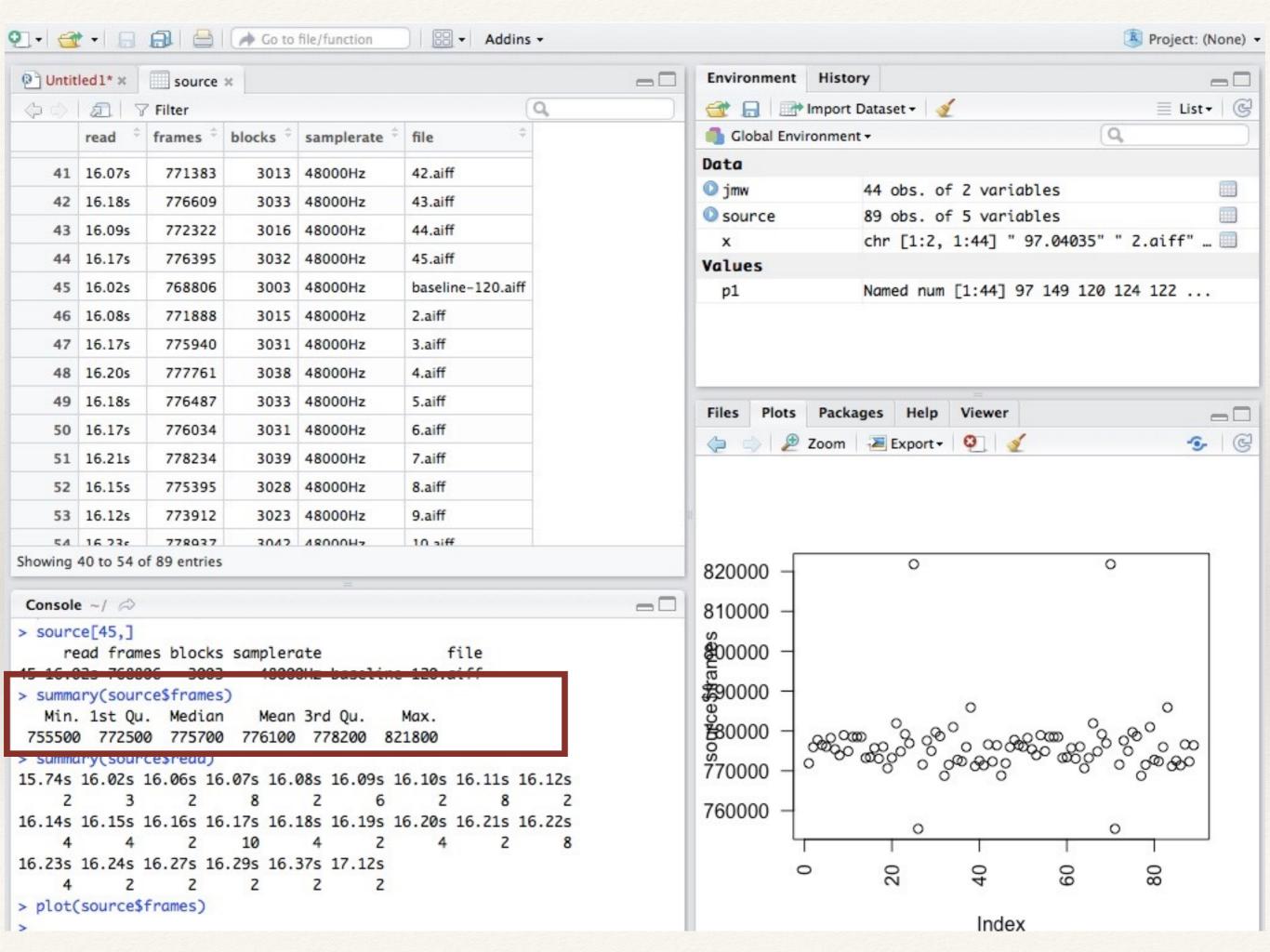
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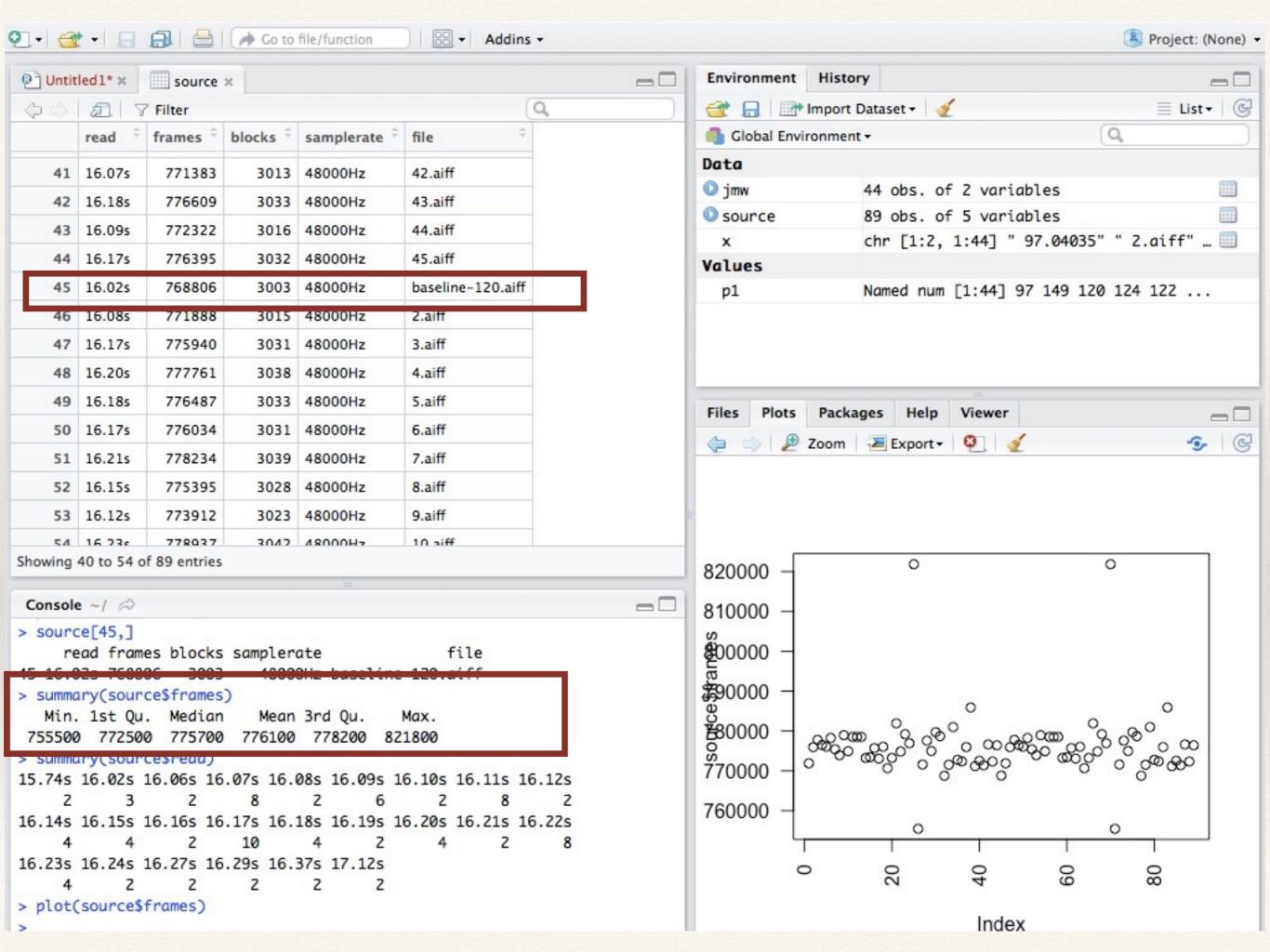
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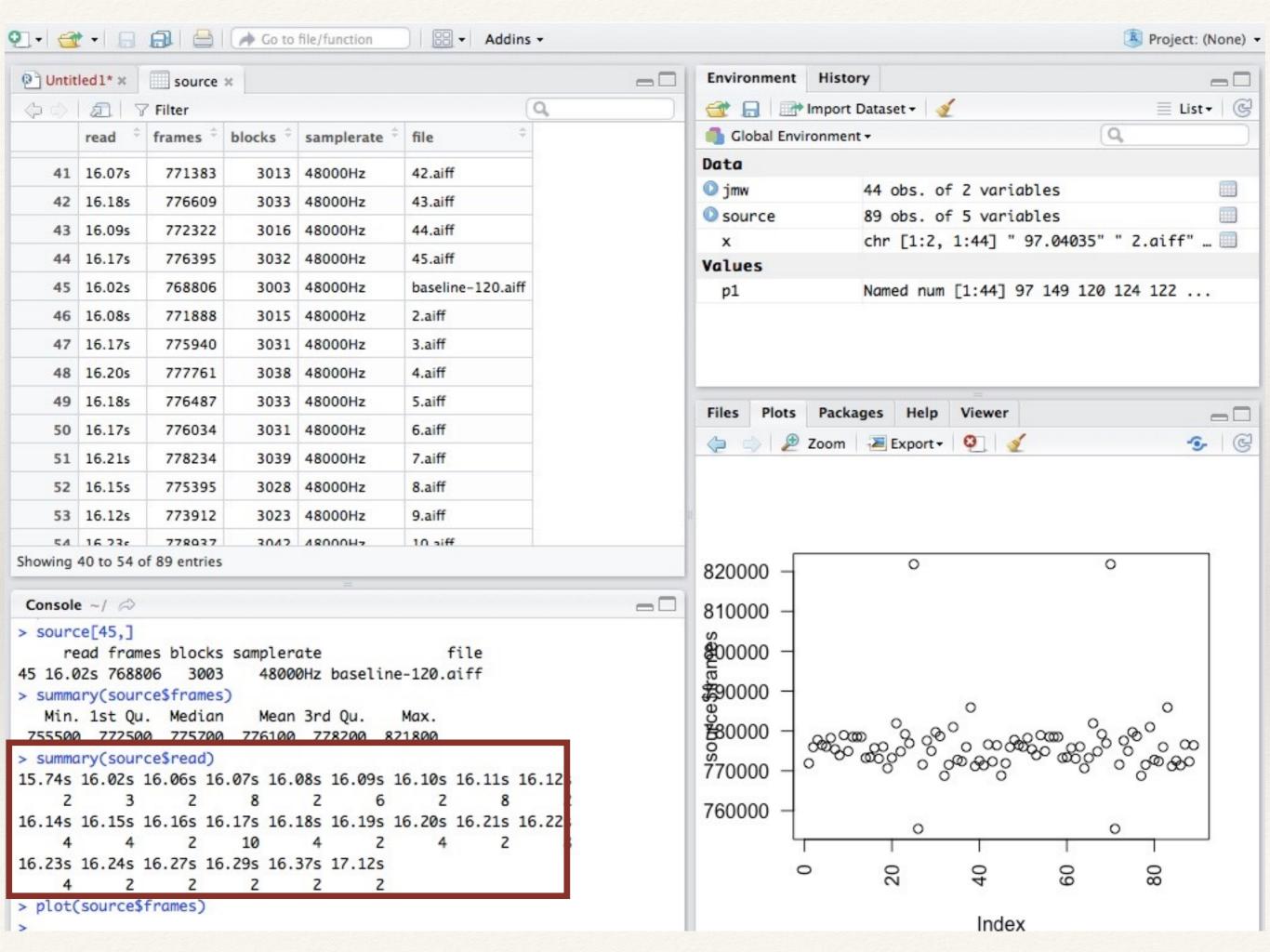
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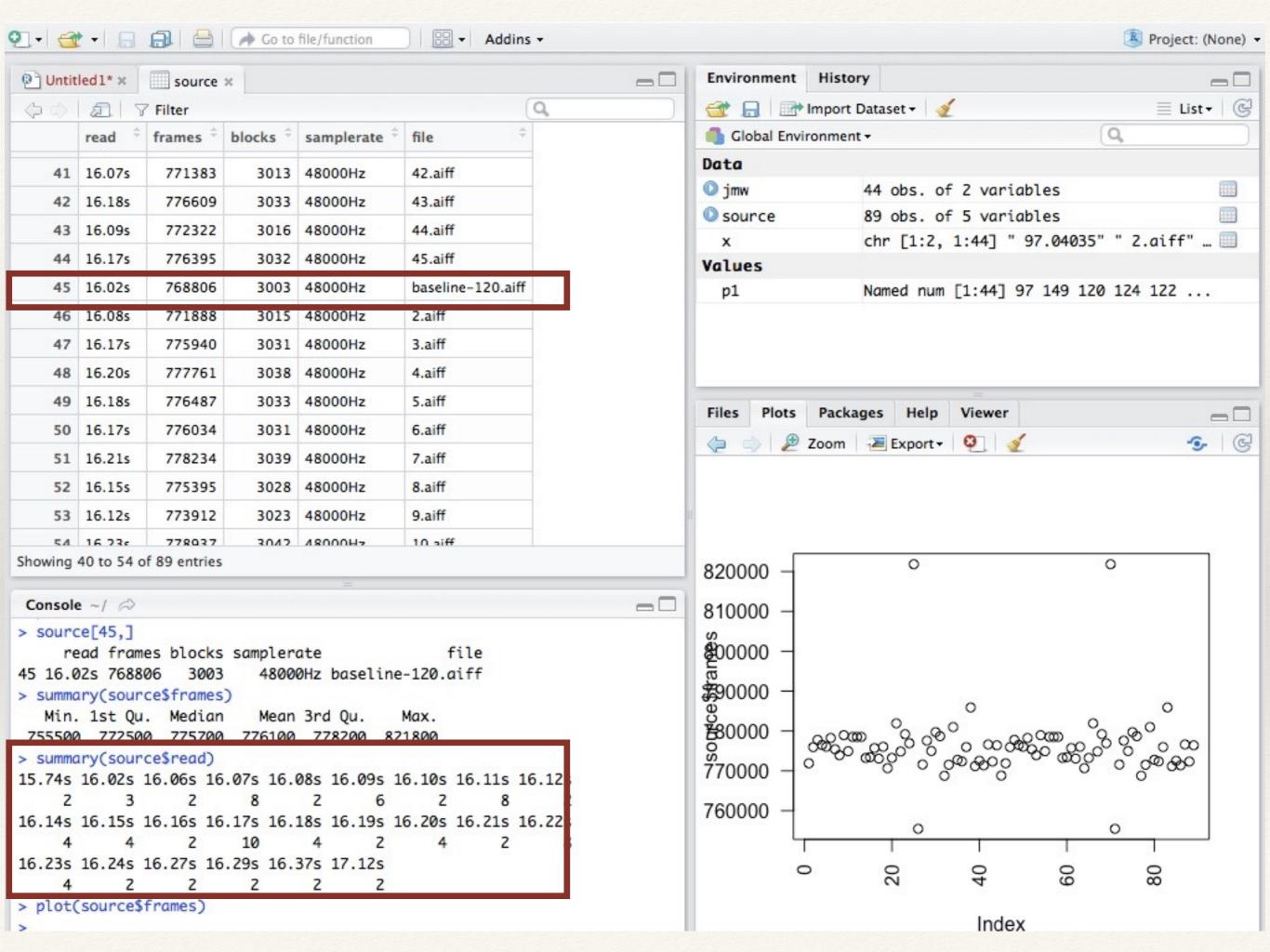
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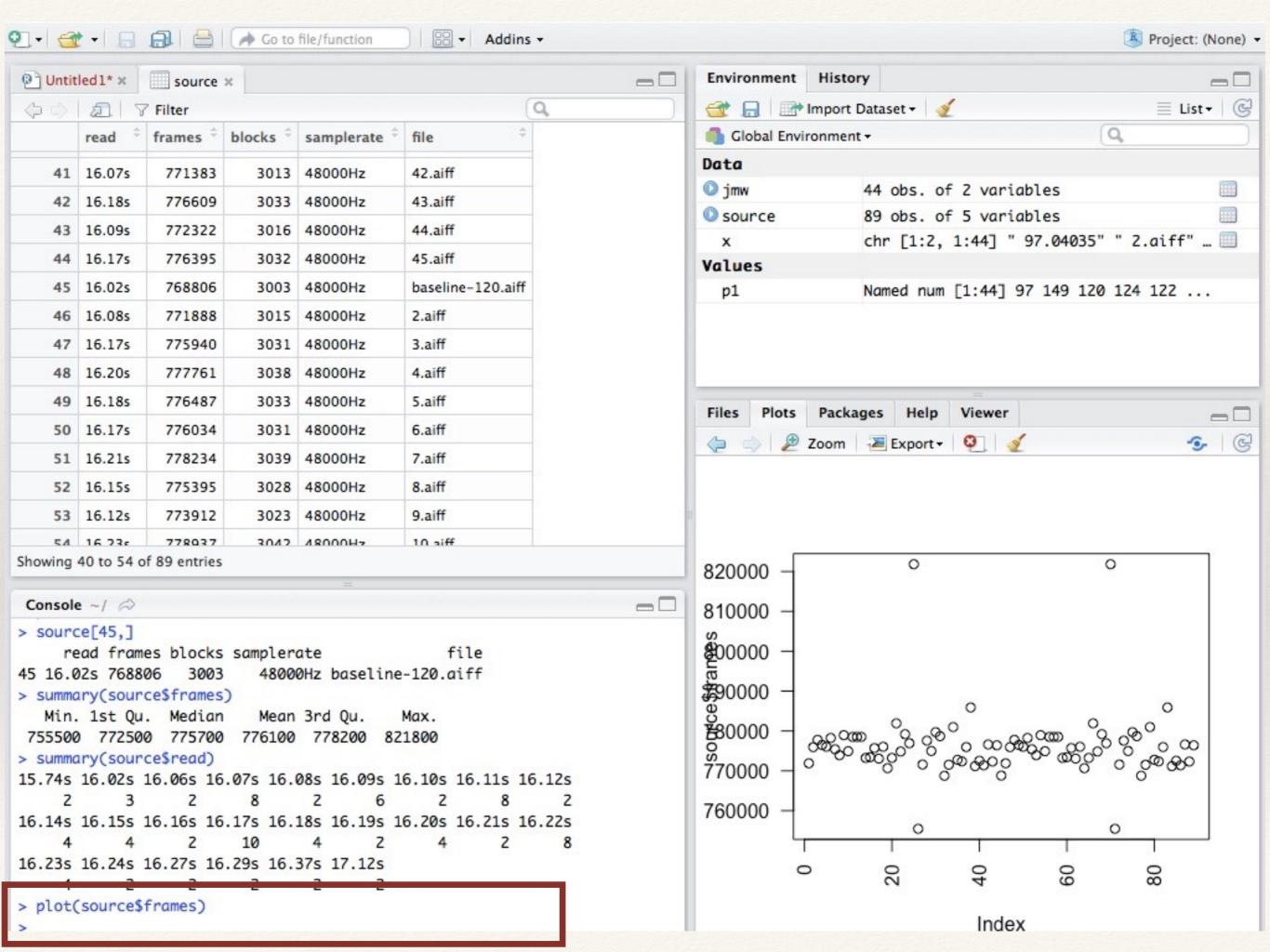


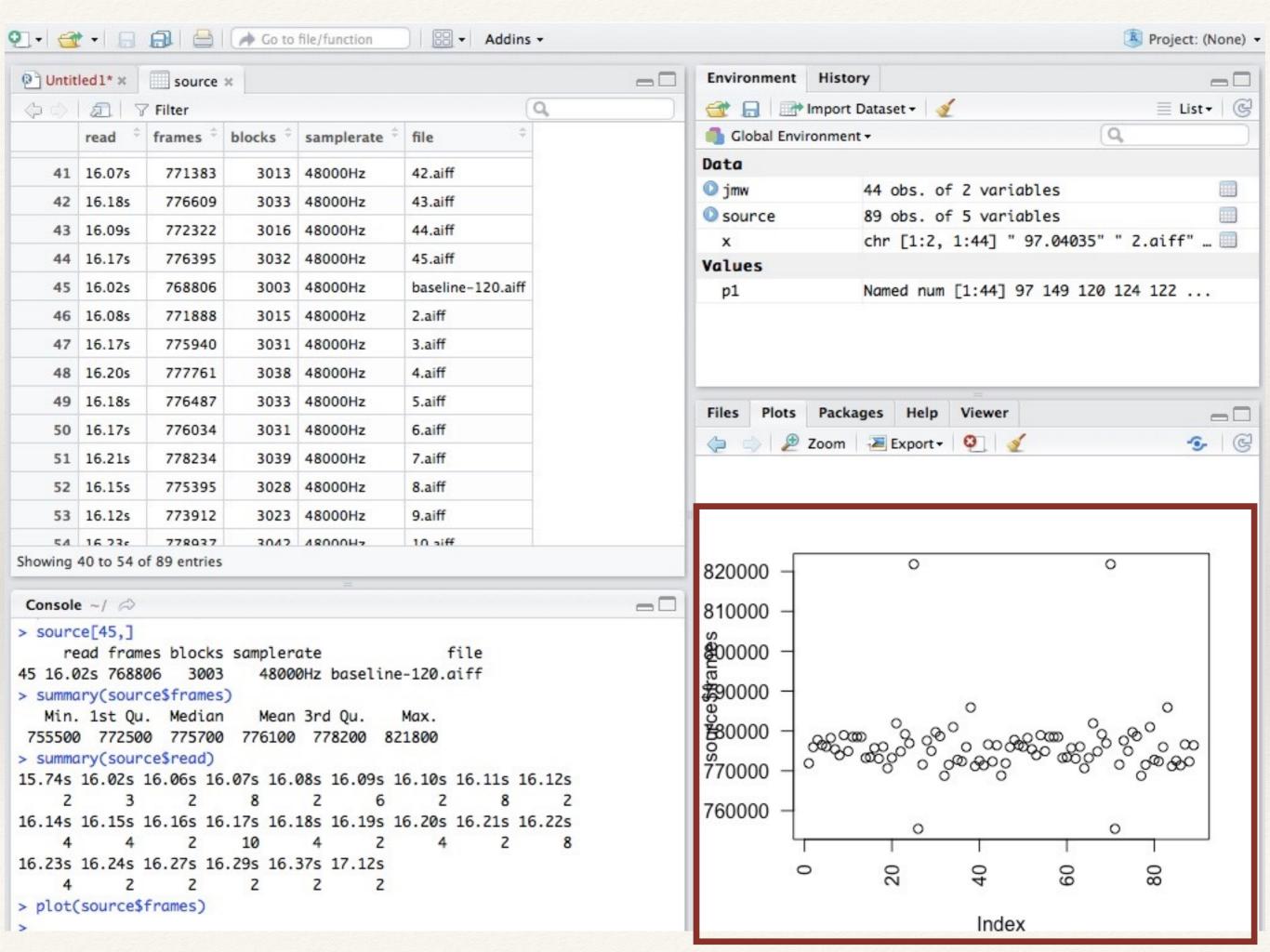












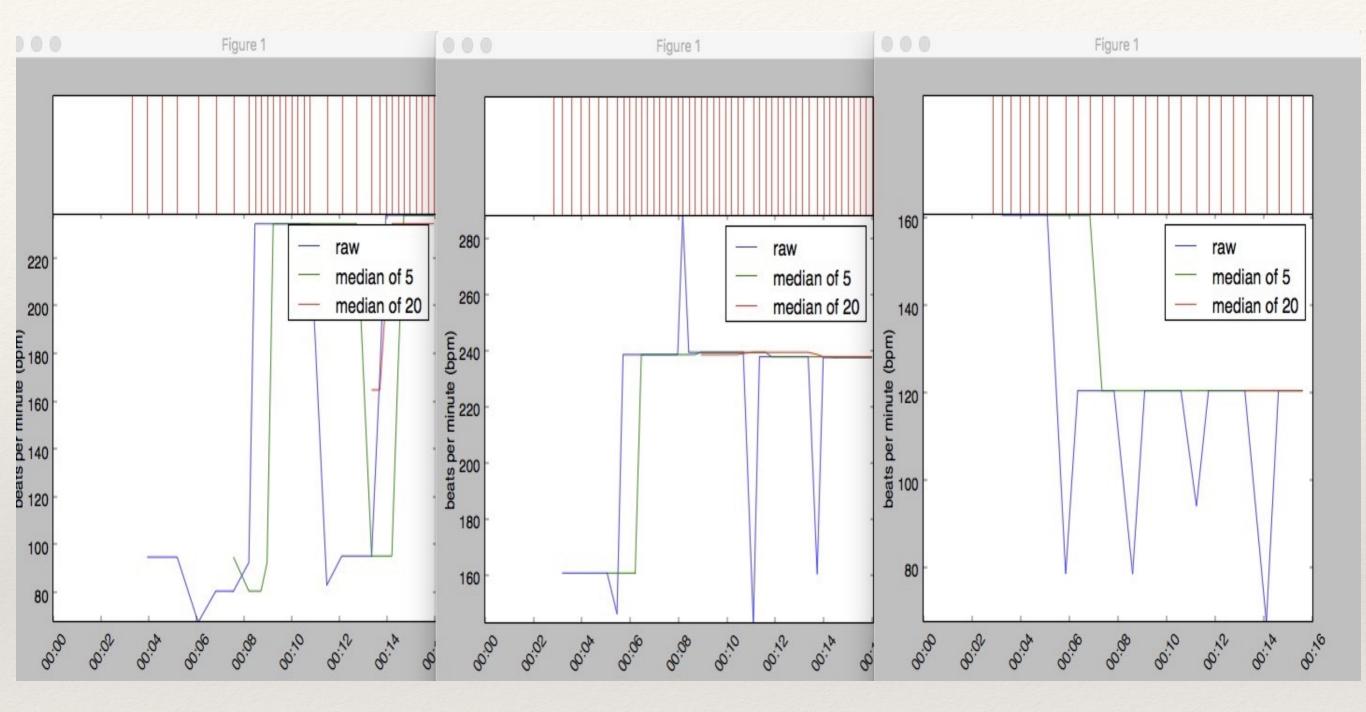
Looking and Temp and Beats

```
#### Analyzing Tempo and Beats
96
97
    Get tempo information from the audio file (dlm_tempo.py)
98
99
100
    #! /usr/bin/env python
101
102
    import sys
103
    from aubio import tempo, source
    from numpy import mean, median, diff
106
                                 # fft size
    win_s = 512
107
    hop_s = win_s // 2
                                 # hop size
108
109
    if len(sys.argv) < 2:
110
        print("Usage: %s <filename> [samplerate]" % sys.argv[0])
111
        sys.exit(1)
113
    filename = sys.argv[1]
114
115
    samplerate = 0
116
    if len( sys.argv ) > 2: samplerate = int(sys.argv[2])
117
118
    s = source(filename, samplerate, hop_s)
    samplerate = s.samplerate
120
    o = tempo("default", win_s, hop_s, samplerate)
121
    # tempo detection delay, in samples
123
    # default to 4 blocks delay to catch up with
124
    delay = 4. * hop_s
125
126
    # list of beats, in samples
    beats = []
128
129
    # total number of frames read
130
    total_frames = 0
131
    while True:
132
         samples, read = s()
133
         print("samples=",samples)
134
         is_beat = o(samples)
135
         if is_beat:
136
             print("is_beat = ",is_beat)
137
             this_beat = int(total_frames - delay + is_beat[0] * hop_s)
138
             print("%f" % (this_beat / float(samplerate)))
139
             beats.append(this_beat)
140
         total_frames += read
```

Analyzing Tempo and Beats

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```



Day 1

Day 54

Baseline

Summary

Summary

- * Although the data didn't show us exactly what we were hoping... it did tell us that we are on the right track.
- * Next time we are going to record digital practices files as opposed to analog files.
- * We are going to look for more improved tools (aubio was great but is very new and appear to be a little buggy)
- * Also Audio as far as we can tell was designed more for music and not drum beats.