# SPECTROGRAM ANALYSIS FOR INSTRUMENT PITCH COMPARISON

# ANALYSED INSTRUMENTS

# TWO INSTRUMENTS FOR TWO "FAMILIES"



Sound waves are produced by a **bow** rubbing strings.

Brass family (wind)

Different pitches are produced using slides, keys and crooks.

**Trumpet** and **violin** are expected to produce *higher* sound with respect to **cello** and **trombone**, whose pitch is considerable *lower*. So similarity may be detected both between instruments belonging the same *family* rather than ones closer in terms of sound *height*.

When plotting spectrograms, each column will correspond to an instrument: (1) Cello, (2) Violin, (3) Trombone, (4) Trumpet.



### Cello

Largest violin after contrabassoon. Played on the ground. Low sound.





### **Trumpet**

Small brass instrument equipped with keys. Sound range is quite high..





### Violin

The smallest of its family. It's played on the shoulder. Sound is high..





### Trombone

Equipped with a large sliding *coulisse*. Several sizes exist, with generally middle-low sound.



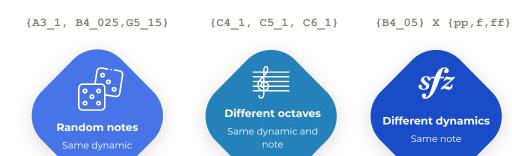
# CHOSEN **NOTE SETS**

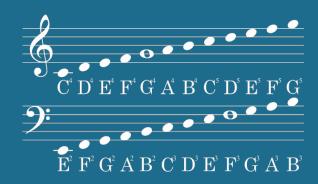
# **ISOLATE VARIATING FACTORS (DYNAMIC & HEIGHT)**

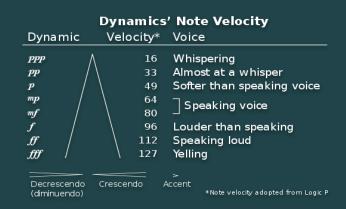
A huge constraint was the <u>non strict regularity in the dataset</u>: basing on note ids, not every dynamic was present for each note.

For arc instruments, only "arco normal" played notes were picked to avoid introducing another source of complexity.

Each set aims to expose how different aspects impacts on waves and spectrum.







# ABOUT **METHOD** AND **CODE**





Librosa and matplotlib are the hinges of this experimentation.

<u>Documentation examples</u> were followed to implement a the <u>various plotting</u> functions considerable useful for the analysis.

A global high order function was implemented to display subplots in a grid and make observation easier.

Harmonics isolation was also implemented, as an option, within this function, allowing visualize difference with respect to the full spectrum.

The full notebook can be viewed just by <u>clicking on the Colab Icon</u> in this slide.

```
# Wrapping plotting function to display all images closer
nnotes = 3
ninstruments = 4

def grid_plot(plotting_func, h_only=False, save_feature=True):
    fig, axarr = plt.subplots(nrows=nnotes, ncols=ninstruments)
    fig.set_size_inches(10*ninstruments, 6*nnotes)

# Dislpay plots having notes on rows and instruments on columns
for i, sound in enumerate(notes):
    row = i % nnotes
    col = i // nnotes
    name, note = sound

# Analyzing only harmonics can be useful
    if(h_only):
    note = (ilbrosa.effects.hpss(note[0])[0], note[1])

#Execute transformation, features are saved for further usage.
feature, feature_name = plotting_func(note[0], note[1], name, (fig, axarr[row, col] ))
#Harmonics != full
    if(h_only):
    feature_name += '_harmonics'

# Saving computed stuff for further usage
    if(save_feature):
    features[name][feature_name] = feature
    plt.tight_layout()
```



### **Preprocessing**

All considered notes were uploaded on a new Google Drive folder and loaded into the Colab Notebook using librosa.load function.

Each extracted time-series was then trimmed, applying again a librosa function to delete "silence" moment (according to a predetermined threshold).

# COMPUTATION: **STFT**

librosa compute spectrograms using the Short Time (Discrete) Fourier Transform.

The STFT represents a signal in the time-frequency domain by computing discrete Fourier transforms (DFT) over short overlapping windows.

The number of rows in the STFT matrix  $\mathbf{D}$  is  $(1 + n_f f f t/2)$ . The default value,  $n_f f t = 2048$  samples, corresponds to a physical duration of 93 milliseconds at a sample rate of 22050 Hz, i.e. the default sample rate in librosa.



### Theoretical foundation

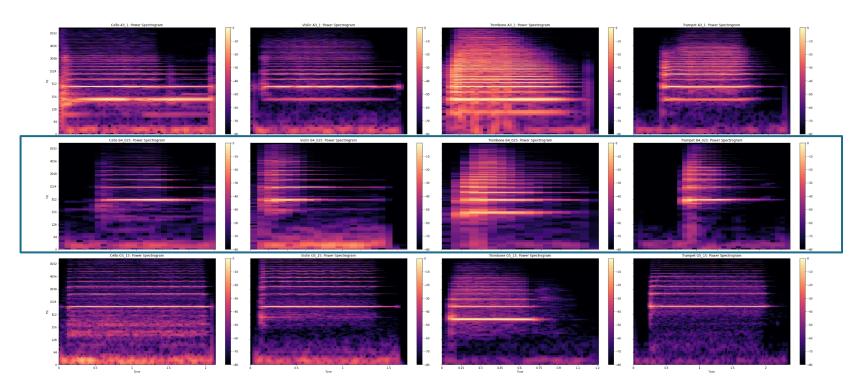
There is no significant information lost if we replace the short term spectrum X(f, t) with its sampled version Xnm by sampling it at the two Nyquist periods 1/F and 1/T.

$$X(f,t) = \int_{-\infty}^{\infty} w(t- au) x( au) e^{j2\pi f au} d au$$

$$W(f) = \int_{-\infty}^{\infty} w(\tau)e^{j2\pi f \tau} d\tau$$

$$X_{nm} = \sum_{k=0}^{T-1} w(nD - k)x(k)e^{j2\pi km/T}$$
  $D = 1/F$ 

# **FULL POWER SPECTROGRAM**



# COMPUTATION: HARMONICS EXTRACTION

Minimization the L2 norm of the power spectrogram gradients allows, formally, to achieve separation.

Median filters operates replacing a given sample in a signal by the median of the signal values in a window around the sample, smoothing both the percussive and harmonics components...

$$J(\mathbf{H},\mathbf{P}) = \frac{1}{2\sigma_H^2} \sum_{h,i} \left( H_{h,i-1} - H_{h,i} \right)^2 + \frac{1}{2\sigma_P^2} \sum_{h,i} \left( P_{h-1,i} - P_{h,i} \right)^2$$

$$y(n)=\mathrm{median}\{x(n-k:n+k),k=(l-1)/2\}$$

$$P_i = \mathcal{M}\left\{S_i, l_{ ext{perc}}
ight\} \ H_i = \mathcal{M}\left\{S_h, l_{harm}
ight\}$$

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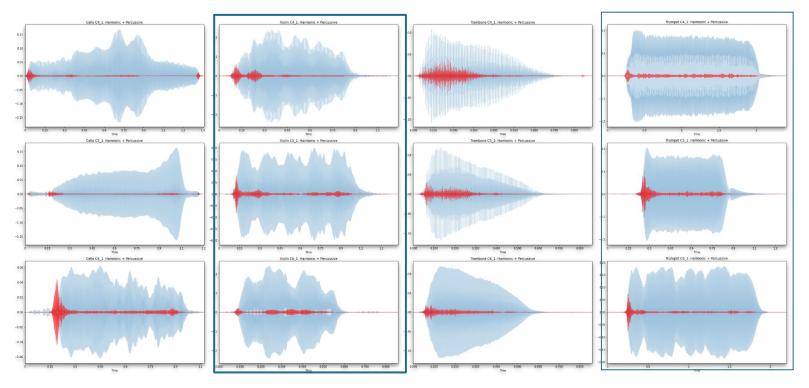


### Intuition behind

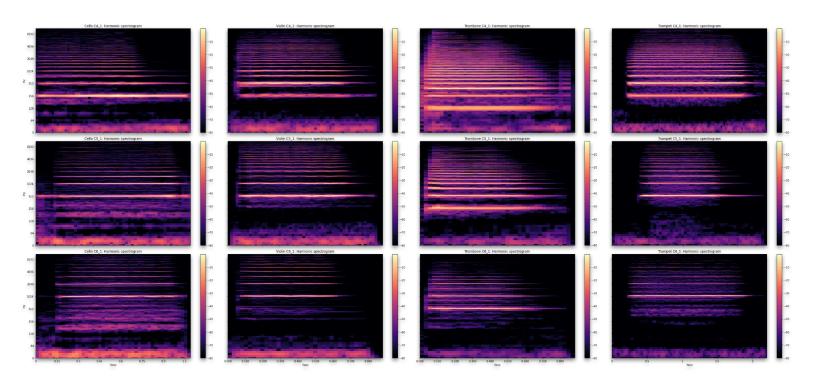
This technique was based on the idea that stable harmonic or stationary components form horizontal ridges on the spectrogram, while percussive components form vertical ridges with a broadband frequency response.

- Fitzgerald, Derry. "Harmonic/percussive separation using median filtering." 13th International Conference on Digital Audio Effects (DAFX10), Graz, Austria, 2010.
- Driedger, Müller, Disch. "Extending harmonic-percussive separation of audio." 15th International Society for Music Information Retrieval Conference (ISMIR 2014), Taipei, Taiwan, 2014.

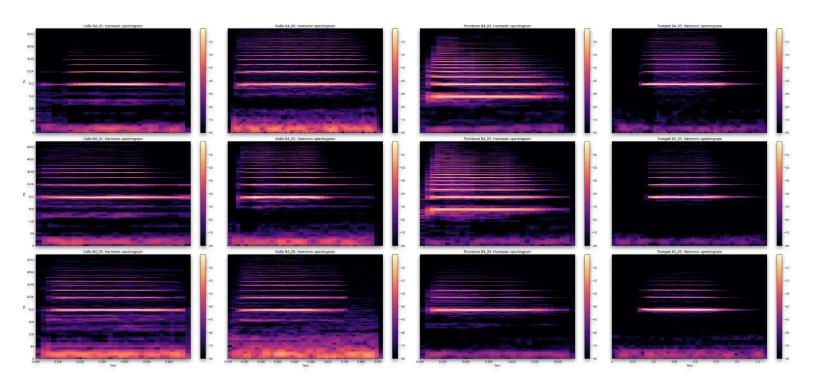
# HARMONICS PERCUSSIVE DECOMPOSITION



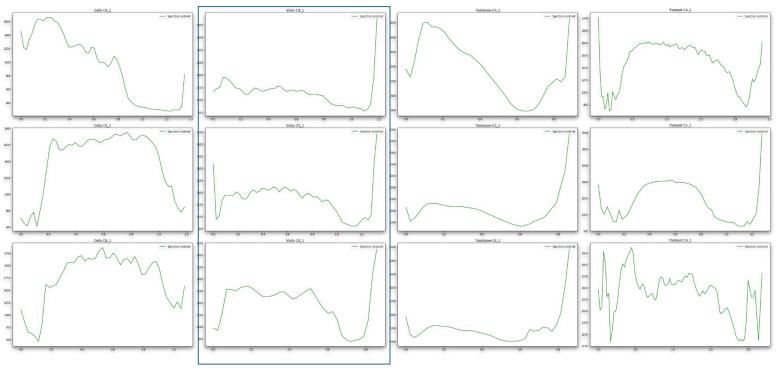
# HARMONICS: SIMILAR SPECTROGRAMS



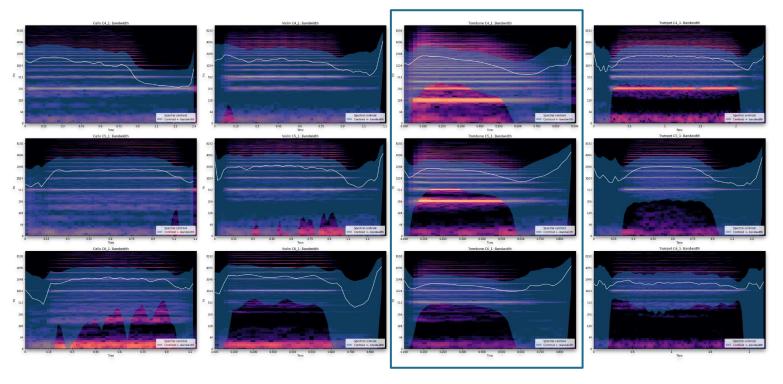
# HARMONICS: SIMILAR SPECTROGRAMS



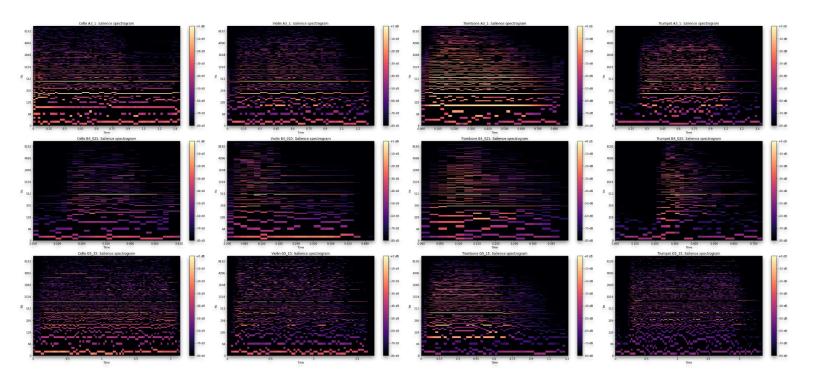
# HARMONICS: SPECTRAL CENTROIDS



# **BANDWIDTH AND CENTROIDS**



# VISUALIZATION: SALIENCE SPECTROGRAM



# FINAL CONSIDERATION

### **ABOUT INSTRUMENT PITCH RECOGNITION**

Each instrument shown a different harmonic profile. Instrument within the same family have closer profiles.

Fixing a note helped a lot reducing complexing, highlighting key similarities and differences.

Octave change produced a vertical translation of the harmonic profile, following the logarithmic scale.

**Dynamic variation** was translated into a (proportional) **decreasement of energy level** for the various harmonics, making spectre of various note similar even if played by different instruments.

# **FURTHER "WEIRD" IMPROVEMENTS**

Several ideas occurred while performing the assignment, like the creation of simplified feature (mean, std, min, max etc..) from computed spectrograms and representation, allowing to computes simple DM algorithms like K-means or even PCA on note-vectors.

Papers on MIR using CNN already exist, maybe they would be useful for the Deep Learning assignment.



### Theoretical deepening

Concept clarified reading papers and experimenting their direct application.



### New skills as a developer

Experience with librosa and matplotlib for time-series and spectrograms.



### Advanced tools explored

STFT, HPPS decomposition, spectral centroids and bandwidth, salience and more...