

1. Mount Google Drive

▼ Lab8 - Codificação de Áudio com ffmpeg

Objetivos:

- Conhecer as opções de codificação do ffmpeg
- Verificar várias taxas mpeg 1, camada 3 e conversão entre codecs
- Variação de tempo

```
from google.colab import drive
drive.mount('/content/drive')
```



Drive already mounted at /content/drive; to attempt to forcibly remount, call

Arquivo .wav full

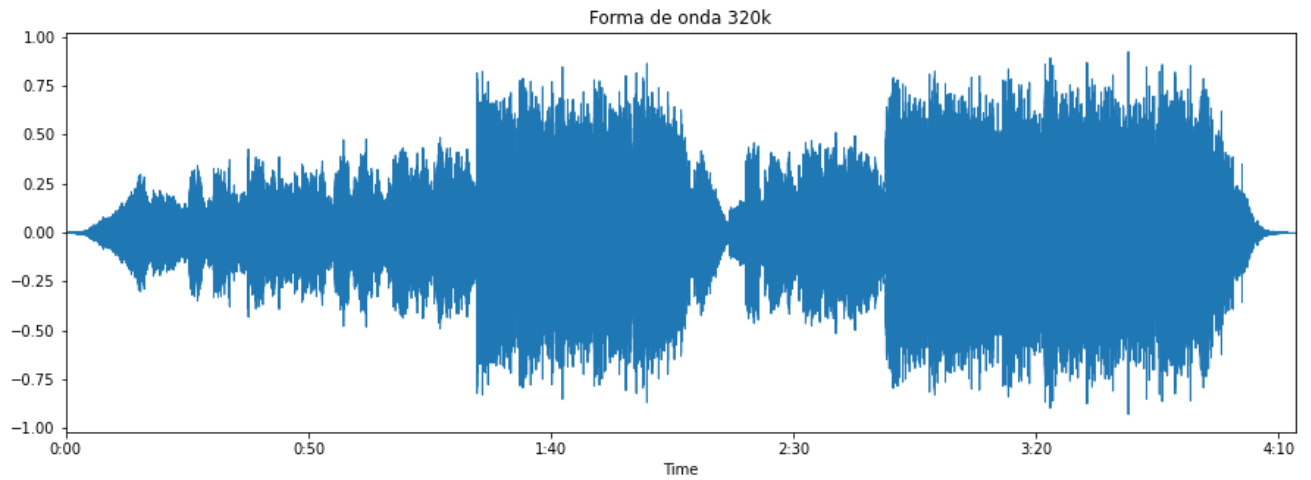
```
import numpy as np
from scipy import signal
import matplotlib.pyplot as plt
import librosa
import librosa.display
import IPython.display
import math
```

```
abba320k = '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/ABBA_Fernando.wav'
fer320 , sr320 = librosa.load(abba320k)
print(type(fer320), type(sr320))
print(fer320.shape, sr320)
# Player será aberto! AGUARDE até abrir!
IPython.display.Audio(data=fer320, rate=sr320)
```

```
pics = '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/pics'
```

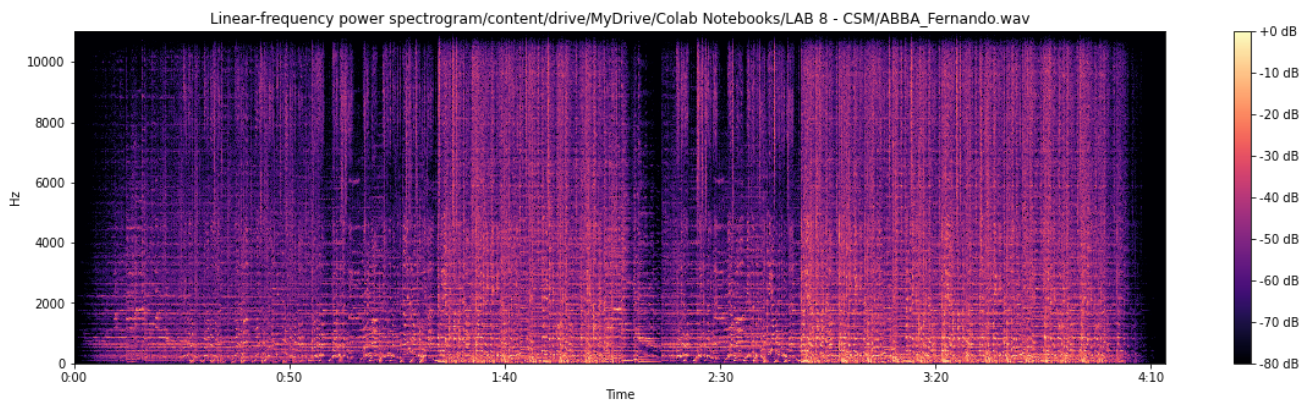
```
plt.figure()
fig, ax = plt.subplots(figsize=(15, 5))
librosa.display.waveplot(fer320, sr=sr320)
plt.title('Forma de onda 320k')
name = 'shapeABBA_320k.png'
plt.savefig(f'{pics}/{name}')
```

<Figure size 432x288 with 0 Axes>



Espectrograma Original

```
plt.figure(figsize=(20, 5))
D = librosa.amplitude_to_db(np.abs(librosa.stft(fer320)), ref=np.max)
librosa.display.specshow(D, x_axis='time', y_axis='linear')
plt.colorbar(format='%+2.0f dB')
plt.title('Linear-frequency power spectrogram'+ abba320k)
name = 'EspectroABBA_320k.png'
plt.savefig(f'{pics}/{name}')
```



Verificar os codecs

▼ Opções do ffmpeg

Conversão wav em mp3, 128kbps

```
!ffmpeg -h
```

```
!ffmpeg -codecs
```

DEA..S s302m	SMPTE 302M
D.A.L. sdx2_dpcm	DPCM Squareroot-Delta-Exact
D.A..S shorten	Shorten
D.A.L. sipr	RealAudio SIPR / ACELP.NET
D.A.L. smackaudio	Smacker audio (decoders: smackaud)
..A.L. smv	SMV (Selectable Mode Vocoder)
D.A.L. sol_dpcm	DPCM Sol
DEA... sonic	Sonic
.EA... sonicls	Sonic lossless
DEA.L. speex	Speex (decoders: libspeex) (encoders: libspeex)
D.AI.S tak	TAK (Tom's lossless Audio Kompressor)
DEA..S truehd	TrueHD
D.A.L. truespeech	DSP Group TrueSpeech
DEAI.S tta	TTA (True Audio)
D.A.L. twinvq	VQF TwinVQ
D.A.L. vmdaudio	Sierra VMD audio
DEA.L. vorbis	Vorbis (decoders: vorbis libvorbis) (encoders: v
..A.L. voxware	Voxware RT29 Metasound
D.A... wavesynth	Wave synthesis pseudo-codec
DEAILS wavpack	WavPack (encoders: wavpack libwavpack)
D.A.L. westwood_snd1	Westwood Audio (SND1) (decoders: ws_snd1)
D.A..S wmalossless	Windows Media Audio Lossless
D.A.L. wmapro	Windows Media Audio 9 Professional
DEA.L. wma1	Windows Media Audio 1
DEA.L. wma2	Windows Media Audio 2
D.A.L. wmavoice	Windows Media Audio Voice
D.A.L. xan_dpcm	DPCM Xan
D.A.L. xma1	Xbox Media Audio 1
D.A.L. xma2	Xbox Media Audio 2
..D... bin_data	binary data
..D... dvd_nav_packet	DVD Nav packet
..D... klv	SMPTE 336M Key-Length-Value (KLV) metadata
..D... otf	OpenType font
..D... scte_35	SCTE 35 Message Queue
..D... timed_id3	timed ID3 metadata
..D... ttf	TrueType font
DES... ass	ASS (Advanced SSA) subtitle (decoders: ssa ass)
DES... dvb_subtitle	DVB subtitles (decoders: dvbsub) (encoders: dvbs
D.S... dvb_teletext	DVB teletext (decoders: libzvbi_teletextdec)
DES... dvd_subtitle	DVD subtitles (decoders: dvdsb) (encoders: dvds
D.S... eia_608	EIA-608 closed captions (decoders: cc_dec)
D.S... hdmv_pgs_subtitle	HDMV Presentation Graphic Stream subtitles (decod
..S... hdmv_text_subtitle	HDMV Text subtitle
D.S... jacosub	JACosub subtitle
D.S... microdvd	MicroDVD subtitle
DES... mov_text	MOV text
D.S... mpl2	MPL2 subtitle
D.S... pjs	PJS (Phoenix Japanimation Society) subtitle
D.S... realtext	RealText subtitle
D.S... sami	SAMI subtitle
..S... srt	SubRip subtitle with embedded timing

..S... ssa	SSA (SubStation Alpha) subtitle
D.S... stl	Spruce subtitle format
DES... subrip	SubRip subtitle (decoders: srt subrip) (encoders: srt subrip)
D.S... subviewer	SubViewer subtitle
D.S... subviewer1	SubViewer v1 subtitle
DES... text	raw UTF-8 text
D.S... vplayer	VPlayer subtitle

```
!ffmpeg -i '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/ABBA_Fernando.wav' -
```

```
ffmpeg version 3.4.8-0ubuntu0.2 Copyright (c) 2000-2020 the FFmpeg developers
built with gcc 7 (Ubuntu 7.5.0-3ubuntu1~18.04)
configuration: --prefix=/usr --extra-version=0ubuntu0.2 --toolchain=hardened
libavutil      55. 78.100 / 55. 78.100
libavcodec      57.107.100 / 57.107.100
libavformat     57. 83.100 / 57. 83.100
libavdevice     57. 10.100 / 57. 10.100
libavfilter     6.107.100 / 6.107.100
libavresample   3.  7.  0 / 3.  7.  0
libswscale      4.  8.100 / 4.  8.100
libswresample   2.  9.100 / 2.  9.100
libpostproc    54.  7.100 / 54.  7.100
```

Guessed Channel Layout for Input Stream #0.0 : stereo

Input #0, wav, from '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/ABBA_F

Metadata:

```
title           : Fernando
album           : Gold: Greatest Hits
artist          : ABBA
date            : 1993
genre           : Dance
track           : 12
```

Duration: 00:04:13.45, bitrate: 1411 kb/s

Stream #0:0: Audio: pcm_s16le ([1][0][0][0] / 0x0001), 44100 Hz, stereo, s

Stream mapping:

Stream #0:0 -> #0:0 (pcm_s16le (native) -> mp3 (libmp3lame))

Press [q] to stop, [?] for help

Output #0, mp3, to 'abba128k.mp3':

Metadata:

```
TIT2           : Fernando
TALB           : Gold: Greatest Hits
TPE1           : ABBA
TDRC           : 1993
TCON           : Dance
TRCK           : 12
TSSE           : Lavf57.83.100
```

Stream #0:0: Audio: mp3 (libmp3lame), 44100 Hz, stereo, s16p, 128 kb/s

Metadata:

```
encoder        : Lavc57.107.100 libmp3lame
```

size= 3961kB time=00:04:13.46 bitrate= 128.0kbits/s speed=36.6x

video:0kB audio:3961kB subtitle:0kB other streams:0kB global headers:0kB muxir

Áudio e Espectrograma do arquivo codificado

```
abba128k = './abba128k.mp3'
print(abba128k)
fer128 , sr128 = librosa.load(abba128k)
print(type(fer128), type(sr128))
```

```

print(fer128.shape, sr128)
# Player será aberto! AGUARDE até abrir!
IPython.display.Audio(data=fer128, rate=sr128)

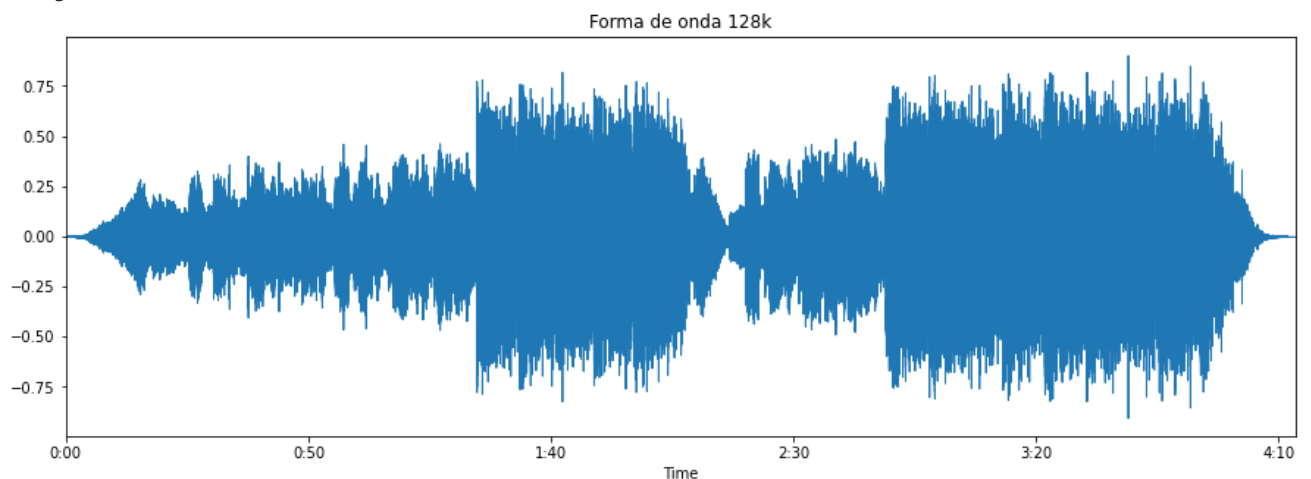
```

```

plt.figure()
fig, ax = plt.subplots(figsize=(15, 5))
librosa.display.waveplot(fer128, sr=sr128)
plt.title('Forma de onda 128k')
name = 'shapeABBA_128k.png'
plt.savefig(f'{pics}/{name}')

```

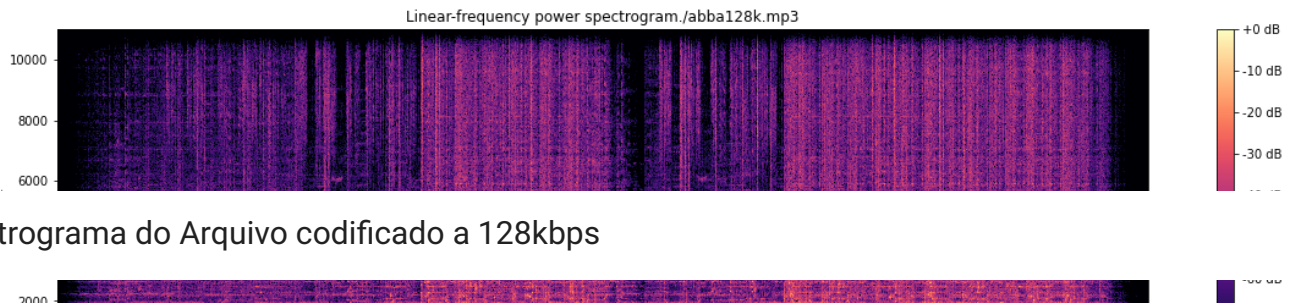
<Figure size 432x288 with 0 Axes>



```

plt.figure(figsize=(20, 5))
D = librosa.amplitude_to_db(np.abs(librosa.stft(fer128)), ref=np.max)
librosa.display.specshow(D, x_axis='time', y_axis='linear')
plt.colorbar(format='%+2.0f dB')
plt.title('Linear-frequency power spectrogram'+ abba128k)
name = 'EspectroABBA_128k.png'
plt.savefig(f'{pics}/{name}')

```



Codificando a 64kbps

```
!ffmpeg -i '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/ABBA_Fernando.wav' -
```

```
ffmpeg version 3.4.8-0ubuntu0.2 Copyright (c) 2000-2020 the FFmpeg developers
built with gcc 7 (Ubuntu 7.5.0-3ubuntu1~18.04)
configuration: --prefix=/usr --extra-version=0ubuntu0.2 --toolchain=hardened
libavutil      55. 78.100 / 55. 78.100
libavcodec     57.107.100 / 57.107.100
libavformat    57. 83.100 / 57. 83.100
libavdevice    57. 10.100 / 57. 10.100
libavfilter     6.107.100 /  6.107.100
libavresample   3.  7.  0 /  3.  7.  0
libswscale     4.  8.100 /  4.  8.100
libswresample   2.  9.100 /  2.  9.100
libpostproc    54.  7.100 / 54.  7.100
```

Guessed Channel Layout for Input Stream #0.0 : stereo

Input #0, wav, from '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/ABBA_F

Metadata:

```
title           : Fernando
album           : Gold: Greatest Hits
artist          : ABBA
date            : 1993
genre           : Dance
track           : 12
```

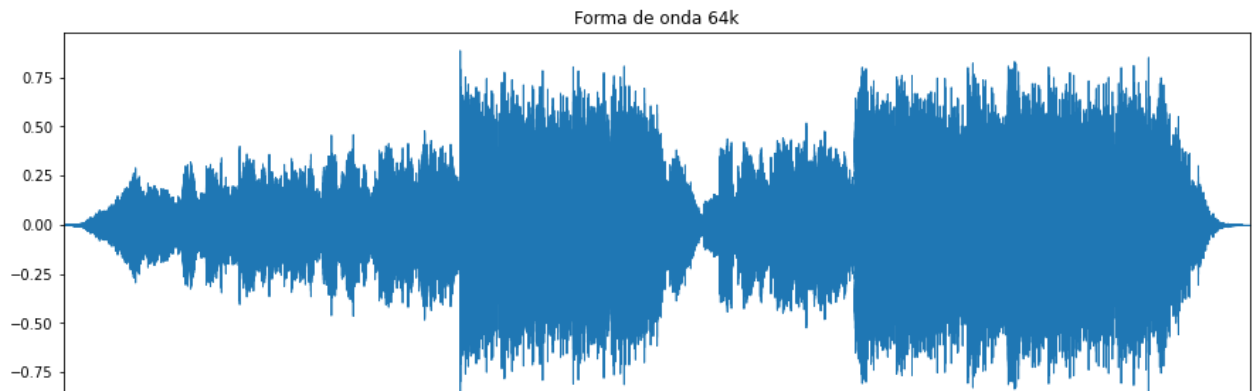
Duration: 00:04:13.45, bitrate: 1411 kb/s

Stream #0:0: Audio: pcm_s16le ([1][0][0][0] / 0x0001), 44100 Hz, stereo, s

File 'abba64k.mp3' already exists. Overwrite ? [y/N] ^C

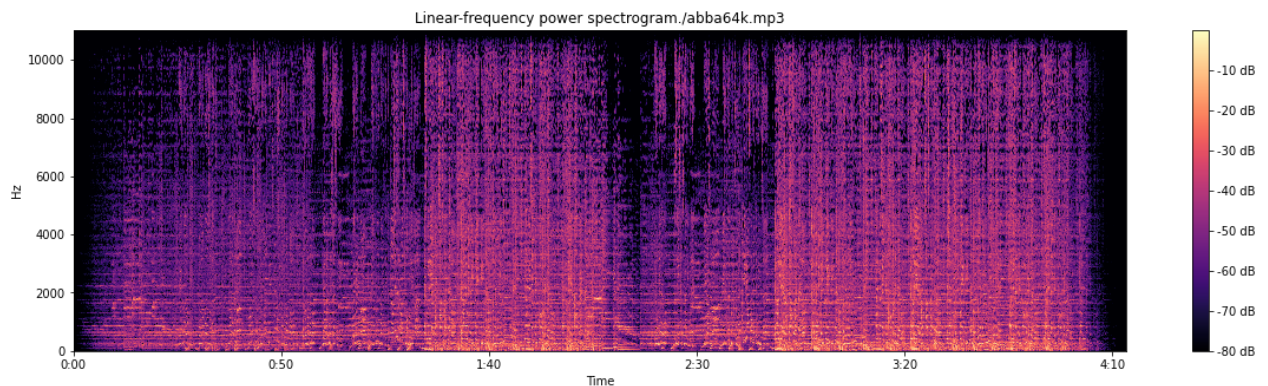
```
abba64k = './abba64k.mp3'
print(abba64k)
fer64 , sr64 = librosa.load(abba64k)
plt.figure()
fig, ax = plt.subplots(figsize=(15, 5))
librosa.display.waveplot(fer64, sr=sr64)
plt.title('Forma de onda 64k')
name = 'shapeABBA_64k.png'
plt.savefig(f'{pics}/{name}')
```

```
./abba64k.mp3
/usr/local/lib/python3.7/dist-packages/librosa/core/audio.py:165: UserWarning: PySoundFile failed. Trying audioread instead.
  warnings.warn("PySoundFile failed. Trying audioread instead.")
<Figure size 432x288 with 0 Axes>
```



Espectrograma do arquivo a 64kbps

```
plt.figure(figsize=(20, 5))
D = librosa.amplitude_to_db(np.abs(librosa.stft(fer64)), ref=np.max)
librosa.display.specshow(D, x_axis='time', y_axis='linear')
plt.colorbar(format='%+2.0f dB')
plt.title('Linear-frequency power spectrogram'+ abba64k)
name = 'EspectroABBA_64k.png'
plt.savefig(f'{pics}/{name}')
```



Audio do arquivo

```
IPython.display.Audio(data=fer64, rate=sr64)
```

Conversão de Codecs


```
!ffmpeg -i abba128k.mp3 -c:a libvorbis abba128k.ogg
```

```
ffmpeg version 3.4.8-0ubuntu0.2 Copyright (c) 2000-2020 the FFmpeg developers
built with gcc 7 (Ubuntu 7.5.0-3ubuntu1~18.04)
configuration: --prefix=/usr --extra-version=0ubuntu0.2 --toolchain=hardened
libavutil      55. 78.100 / 55. 78.100
libavcodec      57.107.100 / 57.107.100
libavformat     57. 83.100 / 57. 83.100
libavdevice     57. 10.100 / 57. 10.100
libavfilter     6.107.100 / 6.107.100
libavresample   3.  7.  0 / 3.  7.  0
libswscale      4.  8.100 / 4.  8.100
libswresample   2.  9.100 / 2.  9.100
libpostproc    54.  7.100 / 54.  7.100
```

```
Input #0, mp3, from 'abba128k.mp3':
```

```
Metadata:
```

```
title           : Fernando
album           : Gold: Greatest Hits
artist          : ABBA
date            : 1993
genre           : Dance
track           : 12
encoder         : Lavf57.83.100
```

```
Duration: 00:04:13.49, start: 0.025057, bitrate: 128 kb/s
```

```
Stream #0:0: Audio: mp3, 44100 Hz, stereo, s16p, 128 kb/s
```

```
Metadata:
```

```
encoder         : Lavc57.10
```

```
Stream mapping:
```

```
Stream #0:0 -> #0:0 (mp3 (native) -> vorbis (libvorbis))
```

```
Press [q] to stop, [?] for help
```

```
Output #0, ogg, to 'abba128k.ogg':
```

```
Metadata:
```

```
title           : Fernando
album           : Gold: Greatest Hits
artist          : ABBA
date            : 1993
genre           : Dance
track           : 12
encoder         : Lavf57.83.100
```

```
Stream #0:0: Audio: vorbis (libvorbis), 44100 Hz, stereo, fltp
```

```
Metadata:
```

```
encoder         : Lavc57.107.100 libvorbis
title           : Fernando
album           : Gold: Greatest Hits
artist          : ABBA
date            : 1993
genre           : Dance
TRACKNUMBER     : 12
```

```
size=      3267kB time=00:04:13.43 bitrate= 105.6kb/s speed=43.1x
```

```
video:0kB audio:3234kB subtitle:0kB other streams:0kB global headers:4kB muxing
```

```
!ls -l
```

```
total 20124
```

```
-rw-r--r-- 1 root root 4056451 Dec 18 20:00 abba128k.mp3
-rw-r--r-- 1 root root 3345162 Dec 18 21:47 abba128k.ogg
-rw-r--r-- 1 root root 8110864 Dec 18 20:50 abba128k_vh.mp3
-rw-r--r-- 1 root root 1014336 Dec 18 20:43 abba16k.mp3
```



```
-rw-r--r-- 1 root root 1014336 Dec 18 20:42 abba32k.mp3
-rw-r--r-- 1 root root 2028304 Dec 18 20:11 abba64k.mp3
-rw-r--r-- 1 root root 1014336 Dec 18 20:47 abba8k.mp3
drwx----- 5 root root 4096 Dec 18 19:44 drive
drwxr-xr-x 1 root root 4096 Dec 3 14:33 sample_data
```

Faça o procedimento para mais três taxas, 32k, 16k e 8kbps.

```
!ffmpeg -i '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/ABBA_Fernando.wav' -
```

```
ffmpeg version 3.4.8-0ubuntu0.2 Copyright (c) 2000-2020 the FFmpeg developers
built with gcc 7 (Ubuntu 7.5.0-3ubuntu1~18.04)
configuration: --prefix=/usr --extra-version=0ubuntu0.2 --toolchain=hardened
libavutil      55. 78.100 / 55. 78.100
libavcodec     57.107.100 / 57.107.100
libavformat    57. 83.100 / 57. 83.100
libavdevice    57. 10.100 / 57. 10.100
libavfilter    6.107.100 / 6.107.100
libavresample  3.  7.  0 / 3.  7.  0
libswscale     4.  8.100 / 4.  8.100
libswresample  2.  9.100 / 2.  9.100
libpostproc   54.  7.100 / 54.  7.100
```

Guessed Channel Layout for Input Stream #0.0 : stereo

```
Input #0, wav, from '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/ABBA_F
```

Metadata:

```
title           : Fernando
album           : Gold: Greatest Hits
artist          : ABBA
date            : 1993
genre           : Dance
track           : 12
```

Duration: 00:04:13.45, bitrate: 1411 kb/s

```
Stream #0:0: Audio: pcm_s16le ([1][0][0][0] / 0x0001), 44100 Hz, stereo, s
File 'abba32k.mp3' already exists. Overwrite ? [y/N] y
```

Stream mapping:

```
Stream #0:0 -> #0:0 (pcm_s16le (native) -> mp3 (libmp3lame))
```

Press [q] to stop, [?] for help

```
Output #0, mp3, to 'abba32k.mp3':
```

Metadata:

```
TIT2            : Fernando
TALB            : Gold: Greatest Hits
TPE1            : ABBA
TDRC            : 1993
TCON            : Dance
TRCK            : 12
TSSE            : Lavf57.83.100
```

```
Stream #0:0: Audio: mp3 (libmp3lame), 44100 Hz, stereo, s16p, 32 kb/s
```

Metadata:

```
encoder         : Lavc57.107.100 libmp3lame
```

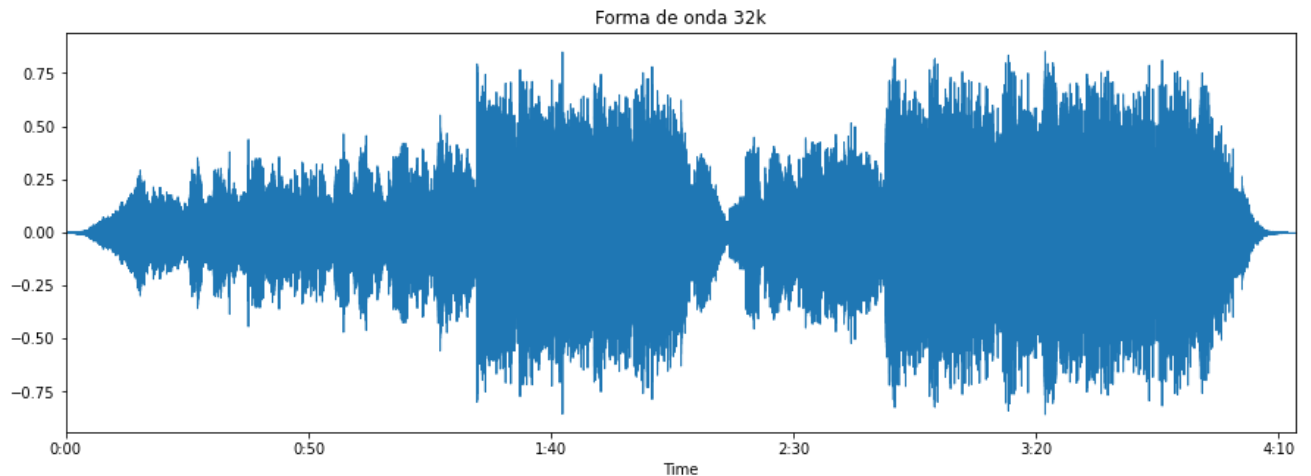
```
size=          991kB time=00:04:13.46 bitrate= 32.0kb/s speed=50.1x
```

```
video:0kB audio:990kB subtitle:0kB other streams:0kB global headers:0kB muxing
```

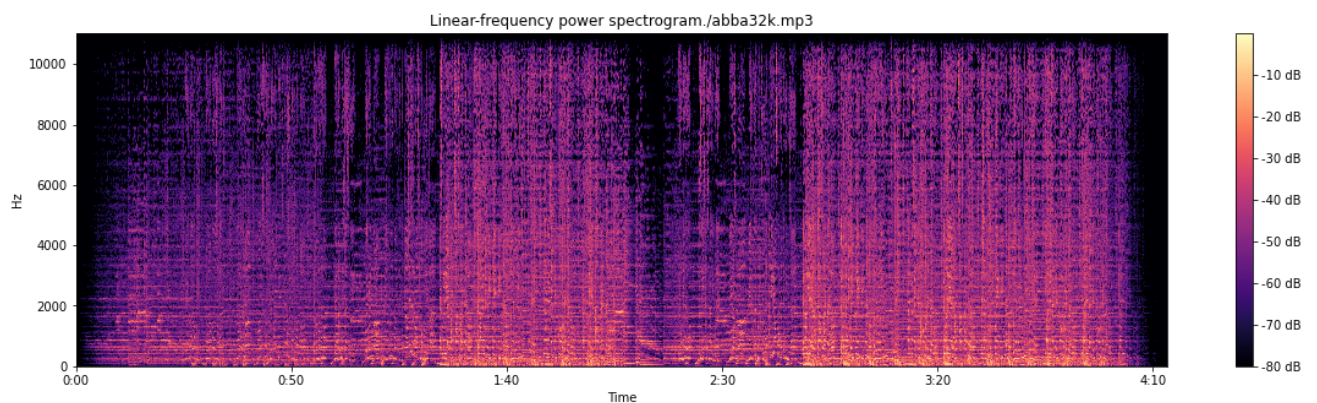
```
abba32k = './abba32k.mp3'
print(abba32k)
fer32 , sr32 = librosa.load(abba32k)
plt.figure()
```

```
fig, ax = plt.subplots(figsize=(15, 5))
librosa.display.waveplot(fer32, sr=sr32)
plt.title('Forma de onda 32k')
name = 'shapeABBA_32k.png'
plt.savefig(f'{pics}/{name}')

./abba32k.mp3
/usr/local/lib/python3.7/dist-packages/librosa/core/audio.py:165: UserWarning:
  warnings.warn("PySoundFile failed. Trying audioread instead.")
<Figure size 432x288 with 0 Axes>
```



```
plt.figure(figsize=(20, 5))
D = librosa.amplitude_to_db(np.abs(librosa.stft(fer64)), ref=np.max)
librosa.display.specshow(D, x_axis='time', y_axis='linear')
plt.colorbar(format='%+2.0f dB')
plt.title('Linear-frequency power spectrogram'+ abba32k)
name = 'EspectroABBA_32k.png'
plt.savefig(f'{pics}/{name}')
```



```
IPython.display.Audio(data=fer32, rate=sr32)
```

```
!ffmpeg -i '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/ABBA_Fernando.wav' -
```

```
ffmpeg version 3.4.8-0ubuntu0.2 Copyright (c) 2000-2020 the FFmpeg developers
built with gcc 7 (Ubuntu 7.5.0-3ubuntu1~18.04)
configuration: --prefix=/usr --extra-version=0ubuntu0.2 --toolchain=hardened
libavutil      55. 78.100 / 55. 78.100
libavcodec     57.107.100 / 57.107.100
libavformat    57. 83.100 / 57. 83.100
libavdevice    57. 10.100 / 57. 10.100
libavfilter    6.107.100 / 6.107.100
libavresample  3.  7.  0 / 3.  7.  0
libswscale     4.  8.100 / 4.  8.100
libswresample  2.  9.100 / 2.  9.100
libpostproc   54.  7.100 / 54.  7.100
```

Guessed Channel Layout for Input Stream #0.0 : stereo

Input #0, wav, from '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/ABBA_F

Metadata:

```
title           : Fernando
album           : Gold: Greatest Hits
artist          : ABBA
date            : 1993
genre           : Dance
track           : 12
```

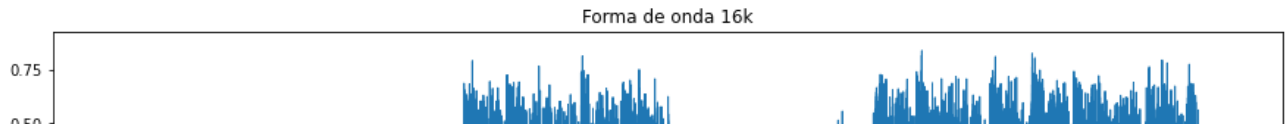
Duration: 00:04:13.45, bitrate: 1411 kb/s

Stream #0:0: Audio: pcm_s16le ([1][0][0][0] / 0x0001), 44100 Hz, stereo, s

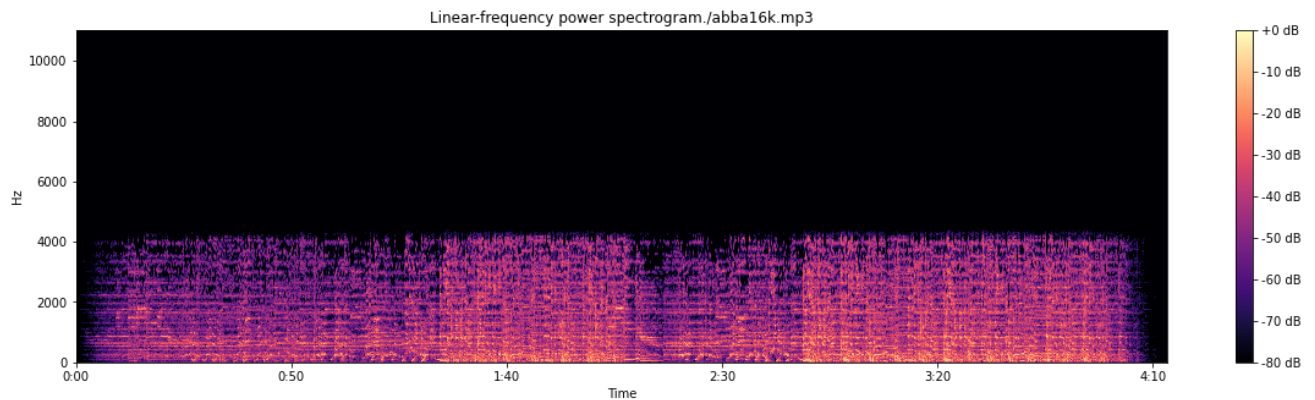
File 'abba16k.mp3' already exists. Overwrite ? [y/N] ^C

```
abba16k = './abba16k.mp3'
print(abba16k)
fer16 , sr16 = librosa.load(abba16k)
plt.figure()
fig, ax = plt.subplots(figsize=(15, 5))
librosa.display.waveplot(fer16, sr=sr16)
plt.title('Forma de onda 16k')
name = 'shapeABBA_16k.png'
plt.savefig(f'{pics}/{name}')
```

```
./abba16k.mp3
/usr/local/lib/python3.7/dist-packages/librosa/core/audio.py:165: UserWarning:
  warnings.warn("PySoundFile failed. Trying audioread instead.")
<Figure size 432x288 with 0 Axes>
```



```
plt.figure(figsize=(20, 5))
D = librosa.amplitude_to_db(np.abs(librosa.stft(fer16)), ref=np.max)
librosa.display.specshow(D, x_axis='time', y_axis='linear')
plt.colorbar(format='%+2.0f dB')
plt.title('Linear-frequency power spectrogram'+ abba16k)
name = 'EspectroABBA_16k.png'
plt.savefig(f'{pics}/{name}')
```



```
IPython.display.Audio(data=fer16, rate=sr16)
```

```
!ffmpeg -i '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/ABBA_Fernando.wav' -
```

```
ffmpeg version 3.4.8-0ubuntu0.2 Copyright (c) 2000-2020 the FFmpeg developers
  built with gcc 7 (Ubuntu 7.5.0-3ubuntu1~18.04)
  configuration: --prefix=/usr --extra-version=0ubuntu0.2 --toolchain=hardened
  libavutil      55. 78.100 / 55. 78.100
  libavcodec     57.107.100 / 57.107.100
  libavformat    57. 83.100 / 57. 83.100
  libavdevice    57. 10.100 / 57. 10.100
  libavfilter    6.107.100 / 6.107.100
  libavresample  3.  7.  0 / 3.  7.  0
  libswscale     4.  8.100 / 4.  8.100
  libswresample  2.  9.100 / 2.  9.100
  libpostproc   54.  7.100 / 54.  7.100
```

Guessed Channel Layout for Input Stream #0.0 : stereo

```
Input #0, wav, from '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/ABBA_F
Metadata:
```

```

title           : Fernando
album           : Gold: Greatest Hits
artist          : ABBA
date            : 1993
genre           : Dance
track           : 12
Duration: 00:04:13.45, bitrate: 1411 kb/s
  Stream #0:0: Audio: pcm_s16le ([1][0][0][0] / 0x0001), 44100 Hz, stereo, s
Stream mapping:
  Stream #0:0 -> #0:0 (pcm_s16le (native) -> mp3 (libmp3lame))
Press [q] to stop, [?] for help
Output #0, mp3, to 'abba8k.mp3':
  Metadata:
    TIT2           : Fernando
    TALB           : Gold: Greatest Hits
    TPE1           : ABBA
    TDRC           : 1993
    TCON           : Dance
    TRCK           : 12
    TSSE           : Lavf57.83.100
  Stream #0:0: Audio: mp3 (libmp3lame), 44100 Hz, stereo, s16p, 8 kb/s
  Metadata:
    encoder        : Lavc57.107.100 libmp3lame
size=      991kB time=00:04:13.46 bitrate= 32.0kbits/s speed=47.1x
video:0kB audio:990kB subtitle:0kB other streams:0kB global headers:0kB muxing

```

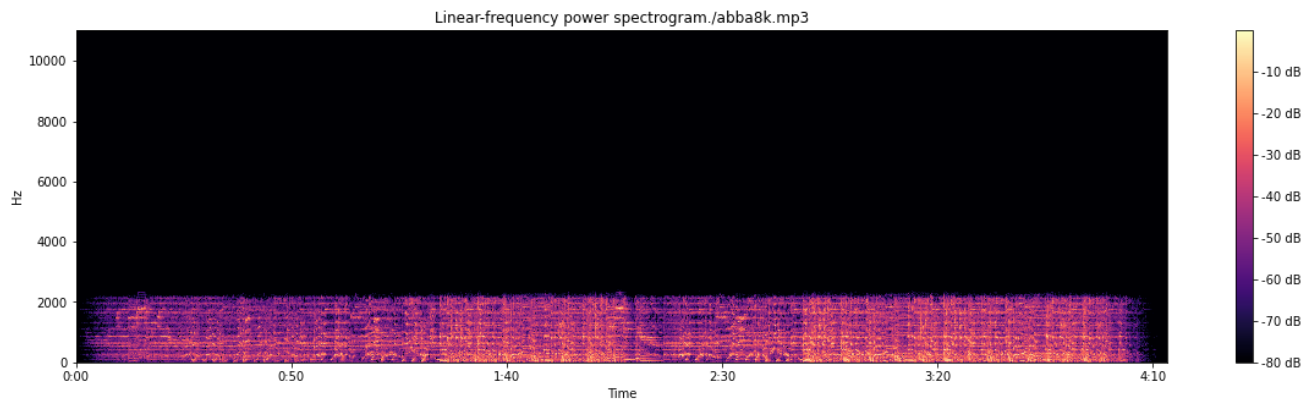
```

abba8k = './abba8k.mp3'
print(abba8k)
fer8 , sr8 = librosa.load(abba8k)
plt.figure()
fig, ax = plt.subplots(figsize=(15, 5))
librosa.display.waveplot(fer8, sr=sr8)
plt.title('Forma de onda 8k')
name = 'shapeABBA_8k.png'
plt.savefig(f'{pics}/{name}')

```

[illegible]

```
plt.figure(figsize=(20, 5))
D = librosa.amplitude_to_db(np.abs(librosa.stft(fer8)), ref=np.max)
librosa.display.specshow(D, x_axis='time', y_axis='linear')
plt.colorbar(format='%+2.0f dB')
plt.title('Linear-frequency power spectrogram'+ abba8k)
name = 'EspectroABBA_8k.png'
plt.savefig(f'{pics}/{name}')
```



```
IPython.display.Audio(data=fer8, rate=sr8)
```

Alterando a velocidade

```
!ffmpeg -i abba128k.mp3 -filter:a "atempo=2.0" -vn abba128k v2.mp3
```

Audio v2

```
abba128k_2 = './abba128k_v2.mp3'
print(abba128k_2)
fer128_2 , sr128_2 = librosa.load(abba128k_2)
IPython.display.Audio(data=fer128_2, rate=sr128_2)
```

Mude para um tempo mais lento

```
!ffmpeg -i abba128k.mp3 -filter:a "atempo=0.5" -vn abba128k_vh.mp3
```

```
ffmpeg version 3.4.8-0ubuntu0.2 Copyright (c) 2000-2020 the FFmpeg developers
  built with gcc 7 (Ubuntu 7.5.0-3ubuntu1~18.04)
```

```

configuration: --prefix=/usr --extra-version=0ubuntu0.2 --toolchain=hardenec
libavutil      55. 78.100 / 55. 78.100
libavcodec     57.107.100 / 57.107.100
libavformat    57. 83.100 / 57. 83.100
libavdevice    57. 10.100 / 57. 10.100
libavfilter    6.107.100 / 6.107.100
libavresample  3.  7.  0 / 3.  7.  0
libswscale     4.  8.100 / 4.  8.100
libswresample  2.  9.100 / 2.  9.100
libpostproc    54.  7.100 / 54.  7.100
Input #0, mp3, from 'abba128k.mp3':
  Metadata:
    title           : Fernando
    album           : Gold: Greatest Hits
    artist          : ABBA
    date            : 1993
    genre           : Dance
    track           : 12
    encoder         : Lavf57.83.100
  Duration: 00:04:13.49, start: 0.025057, bitrate: 128 kb/s
    Stream #0:0: Audio: mp3, 44100 Hz, stereo, s16p, 128 kb/s
      Metadata:
        encoder      : Lavc57.10
Stream mapping:
  Stream #0:0 -> #0:0 (mp3 (native) -> mp3 (libmp3lame))
Press [q] to stop, [?] for help
Output #0, mp3, to 'abba128k_vh.mp3':
  Metadata:
    TIT2           : Fernando
    TALB           : Gold: Greatest Hits
    TPE1           : ABBA
    TDRC           : 1993
    TCON           : Dance
    TRCK           : 12
    TSSE           : Lavf57.83.100
  Stream #0:0: Audio: mp3 (libmp3lame), 44100 Hz, stereo, s16p
    Metadata:
      encoder       : Lavc57.107.100 libmp3lame
size=      7921kB time=00:08:26.88 bitrate= 128.0kbits/s speed= 25x
video:0kB audio:7920kB subtitle:0kB other streams:0kB global headers:0kB muxir

```

```

abba128k_h = './abba128k_vh.mp3'
print(abba128k_h)
fer128_h , sr128_h = librosa.load(abba128k_h)
IPython.display.Audio(data=fer128_h, rate=sr128_h)

```

Repita todos os passos para um arquivo de áudio original de preferência de um CD, mas ao invés da opção codificação com taxa fixa (-b:a), use -q:a N (com N um número de 0 a 9), de 320 a 65kbps.

▼ Relatório

- Comparar os espectrogramas e as taxas de compressão nos arquivos gravados (façam tabelas e as analisem!).
- Apresentar as formas de onda e os espectrogramas para os arquivos com tempo modificados.
- Apresentar arquivos de áudio diferentes para cada integrante do grupo: sugerimos escolherem músicas com instrumental e ritmos diferentes.

```
import numpy as np
from scipy import signal
import matplotlib.pyplot as plt
import librosa
import librosa.display
import IPython.display
import math
```

```
!ffmpeg -h
!ffmpeg -codecs
```

```
!ffmpeg -i '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/Mazurca_Chopin_opus7
chopin320k = './chopin320k.mp3'
chp320 , sr320 = librosa.load(chopin320k)
print(type(chp320), type(sr320))
print(chp320.shape, sr320)
# Player será aberto! AGUARDE até abrir!
IPython.display.Audio(data=chp320, rate=sr320)
```

```
y
yfor N in range(10):
    !ffmpeg -i "/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/Mazurca_Chopin_opu
    chopin320k = './chopin320k.mp3'
    print(chopin320k)
    chp320 , sr320 = librosa.load(chopin320k)
    print(type(chp320), type(sr320))
    print(chp320.shape, sr320)
    IPython.display.Audio(data=chp320, rate=sr320)
    plt.figure(figsize=(20, 5))
    D = librosa.amplitude_to_db(np.abs(librosa.stft(fer320)), ref=np.max)
    librosa.display.specshow(D, x_axis='time', y_axis='linear')
    plt.colorbar(format='%+2.0f dB')
    plt.title('Linear-frequency power spectrogram'+ chopin320k)
```

File "<ipython-input-95-6bd5a8ba7ed6>", line 3

```
chopin{N} = './chopin320k.mp3'
```

^

SyntaxError: invalid syntax

```
!ffmpeg -i '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/01. Green Light.mp3'
green320k = './green320k.mp3'
fer320 , sr320 = librosa.load(green320k)
print(type(fer320), type(sr320))
print(fer320.shape, sr320)
# Player será aberto! AGUARDE até abrir!
IPython.display.Audio(data=fer320, rate=sr320)
```

```
for N in range(10):
    !ffmpeg -i '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/01. Green Light.mp3'
    green320k = './green320k.mp3'
    print(green320k)
    fer320 , sr320 = librosa.load(green320k)
    print(type(fer320), type(sr320))
    print(fer320.shape, sr320)
    # Player será aberto! AGUARDE até abrir!
    IPython.display.Audio(data=fer320, rate=sr320)
    plt.figure(figsize=(20, 5))
    D = librosa.amplitude_to_db(np.abs(librosa.stft(fer320)), ref=np.max)
    librosa.display.specshow(D, x_axis='time', y_axis='linear')
    plt.colorbar(format='%+2.0f dB')
    plt.title('Linear-frequency power spectrogram'+ green320k)
```

```
!ffmpeg -i '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/02. Youth.mp3' -code
youth320k = './youth320k.mp3'
fer320 , sr320 = librosa.load(youth320k)
print(type(fer320), type(sr320))
print(fer320.shape, sr320)
# Player será aberto! AGUARDE até abrir!
IPython.display.Audio(data=fer320, rate=sr320)
```

```
for N in range(10):
    !ffmpeg -i '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/02. Youth.mp3' -co
    youth320k = './youth320k.mp3'
    print(youth320k)
    fer320 , sr320 = librosa.load(youth320k)
    print(type(fer320), type(sr320))
    print(fer320.shape, sr320)
    # Player será aberto! AGUARDE até abrir!
    IPython.display.Audio(data=fer320, rate=sr320)
    plt.figure(figsize=(20, 5))
    D = librosa.amplitude_to_db(np.abs(librosa.stft(fer320)), ref=np.max)
    librosa.display.specshow(D, x_axis='time', y_axis='linear')
```

```
plt.colorbar(format='%+2.0f dB')
plt.title('Linear-frequency power spectrogram'+ youth320k)
```

```
ffmpeg version 3.4.8-0ubuntu0.2 Copyright (c) 2000-2020 the FFmpeg developers
built with gcc 7 (Ubuntu 7.5.0-3ubuntu1~18.04)
configuration: --prefix=/usr --extra-version=0ubuntu0.2 --toolchain=hardened
libavutil      55. 78.100 / 55. 78.100
libavcodec     57.107.100 / 57.107.100
libavformat    57. 83.100 / 57. 83.100
libavdevice    57. 10.100 / 57. 10.100
libavfilter    6.107.100 / 6.107.100
libavresample  3.  7.  0 / 3.  7.  0
libswscale     4.  8.100 / 4.  8.100
libswresample  2.  9.100 / 2.  9.100
libpostproc    54.  7.100 / 54.  7.100
```

Input #0, mp3, from '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/02. Yc

Metadata:

```
track          : 02
title          : Youth
artist         : Glass Animals
album          : How To Be A Human Being
genre          : Electronic
date           : 2016
```

Duration: 00:03:48.81, start: 0.025056, bitrate: 322 kb/s

Stream #0:0: Audio: mp3, 44100 Hz, stereo, s16p, 320 kb/s

Metadata:

```
encoder        : LAME3.99r
```

Side data:

```
replaygain: track gain - -7.600000, track peak - unknown, album gain - u
```

Stream #0:1: Video: mjpeg, yuvj444p(pc, bt470bg/unknown/unknown), 600x600

Metadata:

```
title          : Glass Animals
comment        : Cover (front)
```

File 'youth320k.mp3' already exists. Overwrite ? [y/N] y

Stream mapping:

```
Stream #0:1 -> #0:0 (mjpeg (native) -> png (native))
```

```
Stream #0:0 -> #0:1 (mp3 (native) -> mp3 (libmp3lame))
```

Press [q] to stop, [?] for help

[swscaler @ 0x55a0632de000] deprecated pixel format used, make sure you did see [mp3 @ 0x55a06328a600] Frame rate very high for a muxer not efficiently support Please consider specifying a lower framerate, a different muxer or -vsync 2

Output #0, mp3, to 'youth320k.mp3':

Metadata:

```
TRCK           : 02
TIT2           : Youth
TPE1           : Glass Animals
TALB           : How To Be A Human Being
TCON           : Electronic
TDRC           : 2016
TSSE           : Lavf57.83.100
```

Stream #0:0: Video: png, rgb24(progressive), 600x600 [SAR 1:1 DAR 1:1], q=

Metadata:

```
title          : Glass Animals
comment        : Cover (front)
encoder        : Lavc57.107.100 png
```

Stream #0:1: Audio: mp3 (libmp3lame), 44100 Hz, stereo, s16p

Metadata:

```
encoder        : Lavc57.107.100 libmp3lame
```

Side data:

```
replaygain: track gain - -7.600000, track peak - unknown, album gain - u
```

[mp3 @ 0x55a0632b9900] Header missing

```
!ffmpeg -i '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/01. Green Light.mp3'
push320k = './push320k.mp3'
fer320 , sr320 = librosa.load(push320k)
print(type(fer320), type(sr320))
print(fer320.shape, sr320)
# Player será aberto! AGUARDE até abrir!
IPython.display.Audio(data=fer320, rate=sr320)

for N in range(10):
    !ffmpeg -i '/content/drive/MyDrive/Colab Notebooks/LAB 8 - CSM/09 - Push.mp3' -co
    push320k = './push320k.mp3'
    print(push320k)
    fer320 , sr320 = librosa.load(push320k)
    print(type(fer320), type(sr320))
    print(fer320.shape, sr320)
    # Player será aberto! AGUARDE até abrir!
    IPython.display.Audio(data=fer320, rate=sr320)
    plt.figure(figsize=(20, 5))
    D = librosa.amplitude_to_db(np.abs(librosa.stft(fer320)), ref=np.max)
    librosa.display.specshow(D, x_axis='time', y_axis='linear')
    plt.colorbar(format='%+2.0f dB')
    plt.title('Linear-frequency power spectrogram'+ push320k)
```

Conclusão:

Como pudemos perceber as representações no espectômetro são bem semelhantes enquanto variamos o N. Se levarmos em conta a variação da taxa de compressão, cada vez que ela diminui os valores de frequência também diminuem tornando as ondas menos largas em relação ao eixo y tornando assim o som um pouco mais grave mesmo que no geral a variação seja quase imperceptível.

