What is loudness measurement ? Why should you measure loudness ? Software Wrap up

Open Source Software for loudness measurement

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Welcome!

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- Projects Manager at France Télévisions, the french public television broadcaster.
- Working on loudness since 2011 (E.B.U PLOUD, French Working Group on Delivery Specs, tools specifications, employees training).

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- 2 Why should you measure loudness?
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What is loudness measurement?

- A standardized way to measure perceived loudness of audio content.
- Defined by I.T.U and E.B.U in open standards.
- Uses filtering, weighting and integration to produce the results.
- Produces 2 dynamic meters and 3 program descriptors.

Momentary & Short-Term Loudness

- Dynamic indicators.
- momentary loudness is measured on a 400 ms sliding window.
- short-term loudness is measured on a 3 s sliding window.
- Mostly used in production.

Integrated Loudness

- **integrated loudness** is the average loudness of a complete program.
- It is expressed in LUFS.
- E.B.U R128 target level : -23 LUFS

Loudness Range

- loudness range is a statistical measure of the loudness levels distribution in a program. (Evaluation of the dynamic of the program).
- It is expressed in LU.
- 20 LU is considered as the maximum loudness range fit for TV broadcasting.

True-Peaks

- true-peaks are the intersample audio peaks.
- It is expressed in dBTP.
- true-peaks \neq sample peaks.
- To be taken into account at D/A stage and lossy encoding.

Why should you measure loudness?

- Because of delivery specs or legal constraints.
- To offer a smooth audio experience to your audience.
- To check maximum true-peaks before encoding or broadcasting.

Software usage context

- Realtime or file-based measure.
- Production metering or conformance checking.
- D.A.W plugin or standalone.
- Demux audio data from container.
- Automated workflow.

Software functionalities

- Integrated loudness, loudness range, max true-peaks.
- Logging, plotting, metadata edition/insertion.
- Audio processing (loudness alignment).

libebur128

- Homepage: https://github.com/jiixyj/libebur128
- Licence: Expat (see http://directory.fsf.org/wiki/License:Expat).
- CMake build system.

libebur128, the library

- It's a C library.
- Easy portability.
- Simple API
- Requires libspeexdsp to measure true-peaks.

libebur128, the loudness scanner

- The scanner needs Glib, GTK and taglib. There is input support for gstreamer, libsndfile, libmpg123, FFmpeg and libmpcdec.
- Measure of various multimedia file formats/codecs.
- Integrated loudness, loudness range and max true-peak descriptors.
- Logging of momentary or short-term loudness.
- ReplayGain tagging.

```
🔞 🖨 🗈 manu@mbpro-207647: ~/Sources/libebur128/build
manu@mbpro-207647:~/Sources/libebur128/build$ ./loudness scan -l -p dbtp '/home/
manu/test files/ebu/1kHz Sine -26 LUFS-16bit.wav'
 Loudness, LRA, True peak
-26.0 LUFS, 0.0 LU, -25.9 dBTP, 1kHz Sine -26 LUFS-16bit.wav
-26.0 LUFS, 0.0 LU, -25.9 dBTP
manu@mbpro-207647:~/Sources/libebur128/build$
```

```
manu@mbpro-207647: ~/Sources/libebur128/build
manu@mbpro-207647:~/Sources/libebur128/build$ ./loudness dump -m 0.1 /home/manu/
test files/ebu/seq-3341-3-16bit-v02.wav
-42.0
-39.0
-37.2
-36.0
-36.0
-36.0
-36.0
-36.0
-36.0
-36.0
-36.0
-36.0
-36.0
-36.0
-36.0
-36.0
-36.0
-36.0
-36.0
-36.0
-36.0
```

FFmpeg

- Homepage: http://www.ffmpeg.org/
- Licence : GPL/LGPL (see http://www.ffmpeg.org/legal.html).
- Autotools build system, or thru packages.

FFmpeg, the libavfilter library

- The ebur128 filter is part of libavfilter.
- Part of the libav* libraries.
- Complex API.
- Implemented only for 48 kHz sampling rate. Other input sampling rates must be resampled.

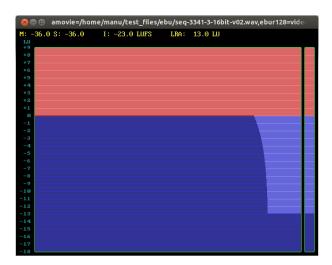
FFmpeg, the ffmpeg executable

- Measure of various multimedia file formats/codecs.
- Integrated loudness and loudness range descriptors.
- Logging of momentary and short-term loudness.
- Real-time short-term loudness plotting and momentary loudness bargraph.

```
🙉 🖨 📵 🛮 manu@mbpro-207647: ~
manu@mbpro-207647:~$ ffmpeq -nostats -i /home/manu/test files/ebu/seq-3341-3-16b
it-v02.way -filter complex ebur128 -f null -
ffmpeg version git-2013-06-21-5d509fb Copyright (c) 2000-2013 the FFmpeg develop
ers
 built on Jun 21 2013 11:12:07 with qcc 4.6 (Ubuntu/Linaro 4.6.3-1ubuntu5)
 configuration: --prefix=/home/manu/ffmpeq build --extra-cflags=-I/home/manu/ff
mpeg_build/include --extra-ldflags=-L/home/manu/ffmpeg_build/lib_--bindir=/home/
manu/bin --extra-libs=-ldl --enable-gpl --enable-nonfree --enable-x11qrab
 libavutil
                52. 37.101 / 52. 37.101
 libavcodec
               55. 16.100 / 55. 16.100
 libavformat
               55. 9.100 / 55. 9.100
 libavdevice 55. 2.100 / 55. 2.100
 libavfilter 3. 77.101 / 3. 77.101
libswscale 2. 3.100 / 2. 3.100
 libswresample 0. 17.102 / 0. 17.102
 libpostproc 52, 3,100 / 52, 3,100
wav @ 0x1a86960] max analyze duration 5000000 reached at 5013333 microseconds
Guessed Channel Layout for Input Stream #0.0 : stereo
Input #0, wav, from '/home/manu/test files/ebu/seq-3341<u>-3-16bit-v02.wav':</u>
 Duration: 00:01:20.00, bitrate: 1536 kb/s
   Stream #0:0: Audio: pcm s16le ([1][0][0] / 0x0001), 48000 Hz, stereo, s16
 1536 kb/s
Output #0. null. to 'pipe:':
 Metadata:
```

```
🔞 🗐 📵 manu@mbpro-207647: ~
    LRA: 13.0 LU
[Parsed ebur128 0 @ 0x1a97740] t: 79.7 M: -36.0 S: -36.0
                                                          I: -23.0 LUFS
    LRA: 13.0 LU
[Parsed ebur128 0 @ 0x1a97740] t: 79.8
                                      M: -36.0 S: -36.0
                                                          I: -23.0 LUFS
    LRA: 13.0 LU
[Parsed ebur128 0 @ 0x1a97740] t: 79.9
                                      M: -36.0 S: -36.0
                                                          I: -23.0 LUFS
    LRA: 13.0 LU
I: -23.0 LUFS
    LRA: 13.0 LU
size=N/A time=00:01:20.00 bitrate=N/A
video:0kB audio:15000kB subtitle:0 global headers:0kB muxing overhead -100.00014
3%
[Parsed_ebur128_0 @ 0x1a97740] Summary:
 Integrated loudness:
           -23.0 LUES
   Threshold: -34.2 LUFS
 Loudness range:
   LRA:
          13.0 LU
   Threshold: -44.0 LUFS
   LRA low: -36.0 LUFS
   LRA high: -23.0 LUFS
manu@mbpro-207647:~$
```

```
🙉 🖨 📵 🛮 manu@mbpro-207647: ~
manu@mbpro-207647:~$ ffplay -f lavfi -i "amovie=/home/manu/test files/ebu/seq-33
41-3-16bit-v02.wav.ebur128=video=1:meter=9 [out0][out1]"
ffplay version git-2013-06-21-5d509fb Copyright (c) 2003-2013 the FFmpeg develop
ers
 built on Jun 21 2013 11:12:07 with qcc 4.6 (Ubuntu/Linaro 4.6.3-1ubuntu5)
 configuration: --prefix=/home/manu/ffmpeg build --extra-cflags=-I/home/manu/ff
mpeg build/include --extra-ldflags=-L/home/manu/ffmpeg build/lib --bindir=/home/
manu/bin --extra-libs=-ldl --enable-gpl --enable-nonfree --enable-x11qrab
 libavutil
                52. 37.101 / 52. 37.101
 libavcodec
               55. 16.100 / 55. 16.100
 libavformat
               55. 9.100 / 55. 9.100
 libavdevice 55. 2.100 / 55. 2.100
 libavfilter 3. 77.101 / 3. 77.101
 libswscale 2. 3.100 / 2. 3.100
 libswresample 0. 17.102 / 0. 17.102
 libpostproc
                52. 3.100 / 52. 3.100
max analyze duration 5000000 reached at 5013333 microsecondsf=0/0
[Parsed amovie 0 @ 0x7f4140003260] Channel lavout is not set in output stream 0
quessed channel layout is 'stereo'
Input #0, lavfi, from 'amovie=/home/manu/test files/ebu/seq-3341-3-16bit-v02.wav
.ebur128=video=1:meter=9 [out0][out1]':
 Duration: N/A, start: 0.000000, bitrate: 6144 kb/s
   Stream #0:0: Video: rawvideo (RGB[24] / 0x18424752). rgb24. 640x480 [SAR 1:1
DAR 4:3], 10 tbr, 48k tbn, 48k tbc
```

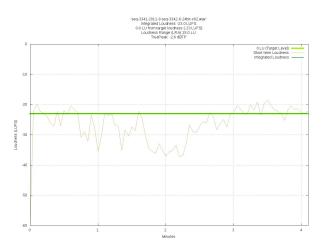


freelcs

- Homepage: http://sourceforge.net/projects/freelcs/
- Licence : GPL.
- Python3 script to install on Ubuntu 12.04.

freelcs

- FreeLCS allows you to build a server that automatically measures and corrects audio loudness according to EBU R128 standard of audio files transferred to it.
- freelcs uses libebur128, gnuplot, sox, media info and (optionally) FFmpeg.
- Drop your files in the HotFolder, freelcs will measure, create a graphic plot, and correct.
- The HotFolder can be shared on the network thru Samba.
- Status monitoring by e-mail and web browser.





Conformance Checking

- 2 sets of audio files for integrated loudness and loudness range conformance checking: E.B.U loudness test set v3 and Compliance material for Recommendation ITU-R BS.1770.
- libebur128 and FFmpeg pass successfully!
- Unfortunately, there's no true-peak conformance test endorsed by I.T.U.

Table

software	IL	LRA	MAX TP	MOM	SHORT	CORR.
libebur128	Х	×	×	Х	Х	-
FFmpeg	×	×	-	×	X	-
freelcs	×	×	X	-	-	X

Performance on a 4'30" .mp4 file with AAC-LC 64 Kbps 48 kHz stereo audio stream.

libebur128 (with true-peak and loudness range) : $\simeq 10$ s.

libebur128 (with loudness range) : \simeq 4 4.

ffmpeg (with momentary and short-term dump) : \simeq 5".

Wrap Up

- There are several open source solutions available for loudness measurement.
- They are mostly fit for file-based measurement.
- To use them in an automated workflow requires some development.
- Plugins ?

Thank You

Questions?

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