

Loudness Range:  
A descriptor to supplement  
loudness normalisation  
in accordance with EBU R 128



**Supplementary information for R 128**

Geneva  
December 2010



## Contents

1.	Introduction .....	5
2.	<i>Loudness Range</i> .....	5
3.	Algorithm Description .....	5
3.1	Algorithm Definition .....	6
4.	Minimum requirements, compliance test.....	7
5.	MATLAB implementation.....	8
6.	References.....	9
7.	Further reading .....	9



## Loudness Range: A descriptor to supplement Loudness normalisation in accordance with EBU R 128

<i>EBU Committee</i>	<i>First Issued</i>	<i>Revised</i>	<i>Re-issued</i>
Technical Committee	2010		

**Keywords:** Loudness, normalisation, dynamic range, statistics

### 1. Introduction

The EBU has studied the needs of audio signal levels in production, distribution and transmission of broadcast programmes. It is of the opinion that an audio-levelling paradigm is needed based on **loudness** measurement. This is described in EBU Technical Recommendation R 128 [1]. In addition to the average loudness of a programme (*'Programme Loudness'*) the EBU recommends that the descriptors *'Loudness Range'* and *'Maximum True Peak Level'* be used for the normalisation of audio signals and to comply with the technical limits of the complete signal chain as well as the aesthetic needs of each programme/station depending on the genre(s) and the target audience.

In this document the descriptor *'Loudness Range'* and the algorithm for its computation will be introduced and explained in detail.

The algorithm was kindly provided by the company TC Electronic.

### 2. Loudness Range

*Loudness Range* (abbreviated 'LRA') quantifies the variation in a time-varying loudness measurement. *Loudness Range* is supplementary to the main audio descriptor, *Programme Loudness*, of EBU R 128. *Loudness Range* measures the variation of loudness on a macroscopic time-scale, in units of LU (Loudness Units). The computation of *Loudness Range* is based on a measurement of loudness level as specified in ITU-R BS.1770 [2]. *Loudness Range* should not be confused with other measures of dynamic range or crest factor, etc.

### 3. Algorithm Description

The computation of *Loudness Range* is based on the statistical distribution of measured loudness. Thus, a short but very loud event would not affect the *Loudness Range* of a longer segment. Similarly the fade-out at the end of a music track, for example, would not increase *Loudness Range* noticeably. Specifically, the range of the distribution of loudness levels is determined by estimating the difference between a low and a high percentile of the distribution. This method is analogous to the *Interquartile Range (IQR)*, used in the field of descriptive statistics to obtain a robust estimate of the spread of a data sample.

*Loudness Range* furthermore employs a cascaded gating method. Certain types of programme may be, overall, very consistent in loudness, but have some sections with very low loudness, for

example only containing background noise (e.g. like atmosphere). If *Loudness Range* did not use the gating, such programmes would (incorrectly) get quite a high *Loudness Range* measurement, due to the relatively large difference in loudness between the regions of background noise and those of normal (foreground) loudness.

The *Loudness Range* algorithm is independent of the sample rate and format of the input signal.

### 3.1 Algorithm Definition

The input to the algorithm is a vector of loudness levels, computed as specified in ITU-R BS.1770 [2], using a *sliding analysis-window* of length **3 seconds** for integration. An overlap between consecutive analysis-windows must be used in order to prevent loss of precision in the measurement of shorter programmes. A minimum block overlap of **66%** (i.e. a minimum 2 s of overlap) between consecutive analysis windows is required; the exact amount of overlap is implementation-dependent.

A cascaded gating scheme is employed which uses an absolute threshold of very low level, in combination with a relative threshold of higher, signal-dependent, level.

The purpose of the relative-threshold gating is to gate out any periods of silence or background noise, using a method that is independent of any level-normalisation of the input signal. The lower edge of *Loudness Range* should not be defined by the noise floor (which may be inaudible), but should instead correspond to the weakest 'real' signal. The relative threshold is set to a level of **-20 LU** relative to the absolute-gated loudness level. The purpose of the absolute-threshold gate is to make the conversion from the relative threshold to an absolute level robust against longer periods of silence or low-level background noise. The absolute threshold is set to **-70 LUFS**, because no relevant signals are generally found below this loudness level.

It is noted that measurement of very short programmes, where leading or trailing silence is included, or of programmes consisting, for example, of isolated utterances, could result in misleadingly high values of LRA.

The application of the cascaded gating leaves only the loudness levels of the sliding-window blocks that contain foreground and (medium-level) background sounds, eliminating low-level signals, background noise, and silence. The width of the distribution of these loudness levels is then quantified using a *percentile range*. Percentiles belong to *non-parametric statistics* and are employed in the computation of *Loudness Range* because the loudness levels cannot in general be assumed to belong to a particular statistical distribution.

LRA is defined as the difference between the estimates of the 10<sup>th</sup> and the 95<sup>th</sup> percentiles of the distribution. The lower percentile of **10%**, can, for example, prevent the fade-out of a music track from dominating *Loudness Range*. The upper percentile of **95%** ensures that a single unusually loud sound, such as a gunshot in a movie, cannot by itself be responsible for a large *Loudness Range*.

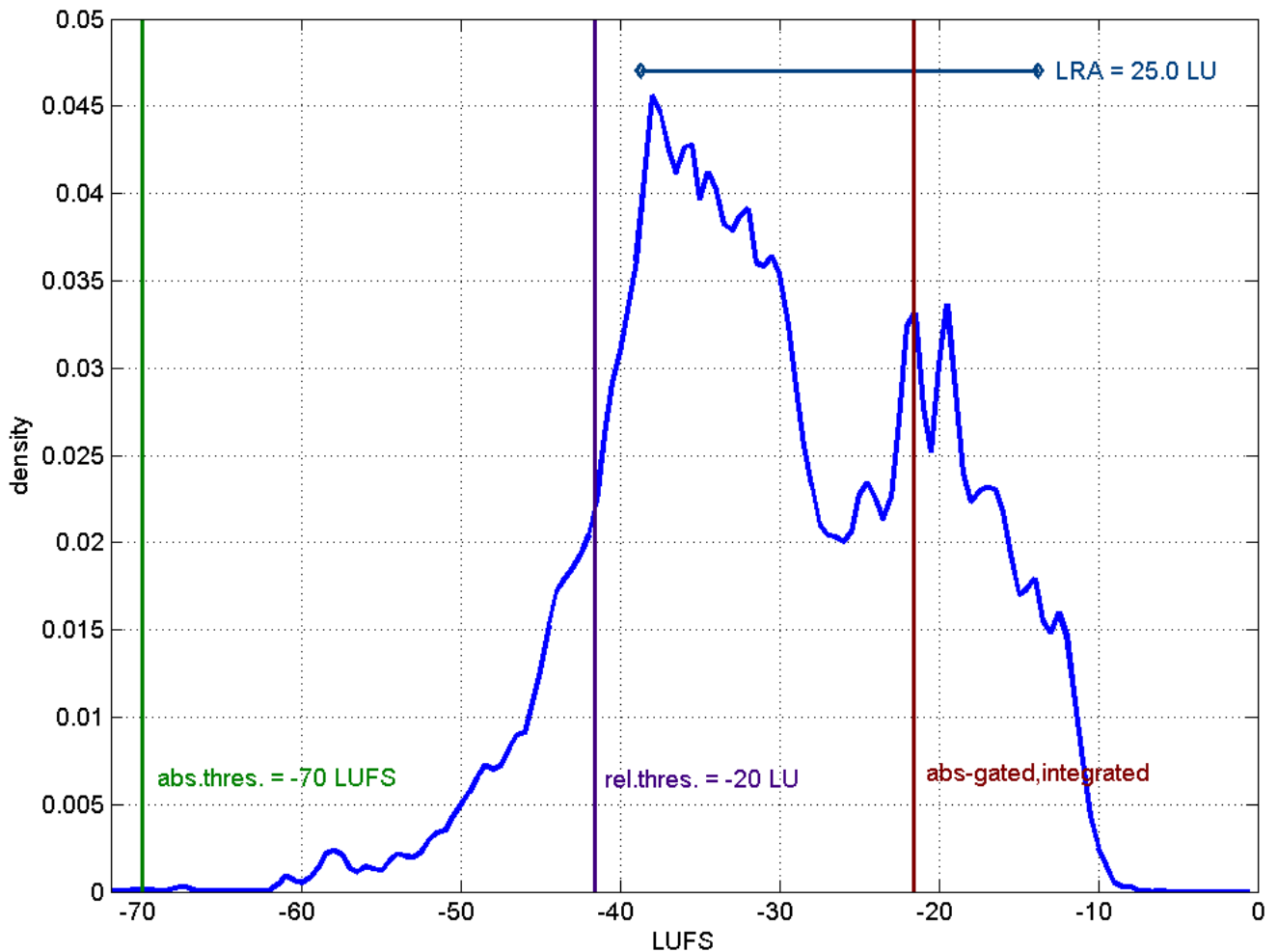


Figure 1: Loudness distribution, with gating thresholds and *Loudness Range* for the film 'The Matrix' (DVD version). Adopted from Skovenborg & Lund (2009) 'Loudness Descriptors to Characterize Wide Loudness-Range Material', 127th AES Conv.

In Figure 1, the absolute threshold is marked at -70 LUFS. The absolute-gated loudness level from that is -21.6 LUFS (marked as *abs-gated,integrated*). The relative threshold is shown 20 LU below that at -41.6 LUFS. The resulting *Loudness Range* (LRA = 25.0 LU) is shown between the 10<sup>th</sup> and 95<sup>th</sup> percentiles of the distribution of loudness levels above the relative threshold.

#### 4. Minimum requirements, compliance test

The *Loudness Range* descriptor is a part of an EBU Mode loudness meter, as defined in EBU Tech Doc 3341 [3]. In the following, a set of Minimum Requirements for the *Loudness Range* computation is provided, in the form of 'minimum requirements test signals' with corresponding expected response and accepted tolerances.

If a loudness meter, offering EBU Mode, does *not* pass these 'minimum requirements' tests, there is a considerable risk that the meter is *not* compliant with EBU Mode. If, on the other hand, a meter does pass the 'minimum requirements' tests this does *not* imply that the meter is sufficiently accurate in all respects of its implementation.

Table 1: Minimum requirements test signals

Test case	Test signal	Expected response and accepted tolerances
1	Stereo sine wave, 1000 Hz, -20.0 dBFS (per-channel peak level); signal applied in phase to both channels simultaneous, 20 s duration; followed immediately by the same signal at -30.0 dBFS (i.e. the tones are 10 dB apart)	LRA = 10 ±1 LU
2	As #1, with the 2 tones at -20.0 dBFS and -15.0 dBFS, respectively	LRA = 5 ±1 LU
3	As #1, with the 2 tones at -40.0 dBFS and -20.0 dBFS, respectively	LRA = 20 ±1 LU
4	As #1, but with 5 tone-segments at -50.0 dBFS, -35.0 dBFS, -20.0 dBFS, -35.0 dBFS, and -50.0 dBFS, respectively; 20 s duration each tone	LRA = 15 ±1 LU
5	Authentic programme 1, stereo, narrow <i>Loudness Range</i> (NLR) programme segment; similar in genre to a commercial/promo	LRA = 5 ±1 LU
6	Authentic programme 2, stereo, wide <i>Loudness Range</i> (WLR) programme segment; similar in genre to a movie/drama	LRA = 15 ±1 LU

In all the above test cases, the expected response is unchanged if the test signal is repeated one or more times in its full length. The loudness meter shall be reset before each measurement.

These 'minimum requirements test signals' [4] are available for download from the EBU Technical website.

## 5. MATLAB implementation

An algorithm for computing *Loudness Range* is provided below using the MATLAB® language (no MATLAB toolbox functions are used). This MATLAB implementation is intended to complement the textual definition of the LRA algorithm. However, other implementations would be equally valid provided that the measurements stay within the permitted tolerance, and even though they might yield slightly different LRA measurements for some input signals.

```
% A MATLAB FUNCTION TO COMPUTE LOUDNESS RANGE
% -----
function LRA = LoudnessRange( ShortTermLoudness )

% Input: ShortTermLoudness is a vector of loudness levels, computed
% as specified in ITU-R BS.1770, using a sliding analysis-window
% of length 3 s, overlap >= 2 s

% Constants
ABS_THRES = -70; % LUFS (= absolute measure)
REL_THRES = -20; % LU (= relative measure)
PRC_LOW = 10; % lower percentile
PRC_HIGH = 95; % upper percentile

% Apply the absolute-threshold gating
abs_gate_vec = (ShortTermLoudness >= ABS_THRES);
% abs_gate_vec is indices of loudness levels above absolute threshold
stl_absgated_vec = ShortTermLoudness(abs_gate_vec);
% only include loudness levels that are above gate threshold
```



```

% Apply the relative-threshold gating (non-recursive definition)
n = length(stl_absgated_vec);
stl_power = sum(10.^(stl_absgated_vec./10))/n;    % undo 10log10, and calculate mean
stl_integrated = 10*log10(stl_power);            % LUFS
rel_gate_vec = (stl_absgated_vec >= stl_integrated + REL_THRES);
% rel_gate_vec is indices of loudness levels above relative threshold
stl_relgated_vec = stl_absgated_vec( rel_gate_vec );
% only include loudness levels that are above gate threshold

% Compute the high and low percentiles of the distribution of
% values in stl_relgated_vec
n = length(stl_relgated_vec);
stl_sorted_vec = sort(stl_relgated_vec);
% sort elements in ascending order
stl_perc_low = stl_sorted_vec(round((n-1)*PRC_LOW/100 + 1));
stl_perc_high = stl_sorted_vec(round((n-1)*PRC_HIGH/100 + 1));

% Compute the Loudness Range descriptor
LRA = stl_perc_high - stl_perc_low;    % in LU

```

## 6. References

- [1] EBU Technical Recommendation R 128 'Loudness normalisation and permitted maximum level of audio signals'
- [2] Recommendation ITU-R BS.1770 'Algorithms to measure audio programme loudness and true-peak audio level'
- [3] EBU Tech Doc 3341 'Loudness Metering: 'EBU Mode' metering to supplement Loudness normalisation in accordance with EBU R 128'
- [4] Minimum requirements test signals for EBU Mode loudness meters are available from the EBU at <http://tech.ebu.ch/loudness>.

## 7. Further reading

EBU Tech Doc 3343 'Practical Guidelines for Production and Implementation in accordance with EBU R 128'

EBU Tech Doc 3344 'Practical Guidelines for Distribution of Programmes in accordance with EBU R 128'