

IN PRAISE OF LOUDNESS

SAMUEL FISCHMANN

Se Spar Se



In Praise of Loudness

Sam Fischmann





Who is this guy?

Co-Four 20+ yea Longtim Dog per

CRYSTAL . CLEAR. STUPID LOUIN

- Co-Founder of Musik Hack
- 20+ years software development
- Longtime musician, composer, and cook
- Dog person / allergic to cats



What is this talk about?

How is it measured? What are the standards related to it? What are the myths?

What is loudness and how is it achieved?





The Basics of Digital Loudness **CRASH COURSE!**







NOTDYNAMICENOUGH

tOodyna C C MUSIK





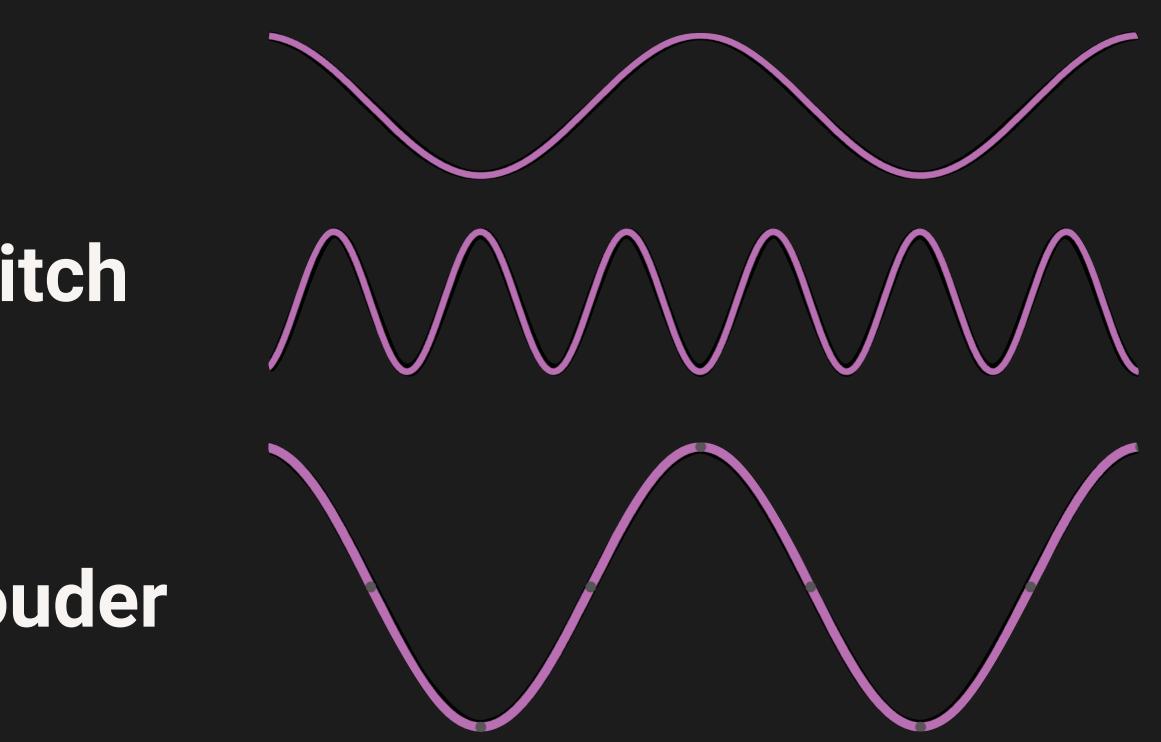


Loudness is Movement

Sound is **movement**

Faster movement is higher pitch

More peak to peak travel is louder



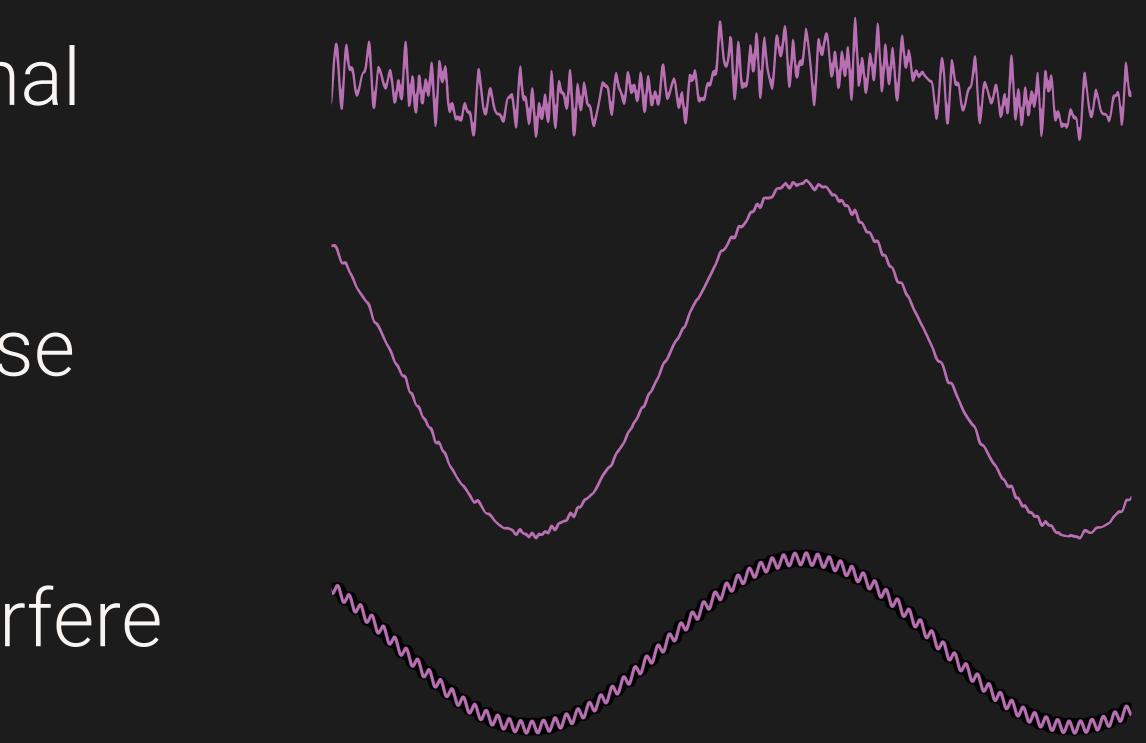


Signal-to-Noise Ratio (SNR)

Noise can overpower signal

Signal can overpower noise

But they don't always interfere





Digital Movement and Bit Depth

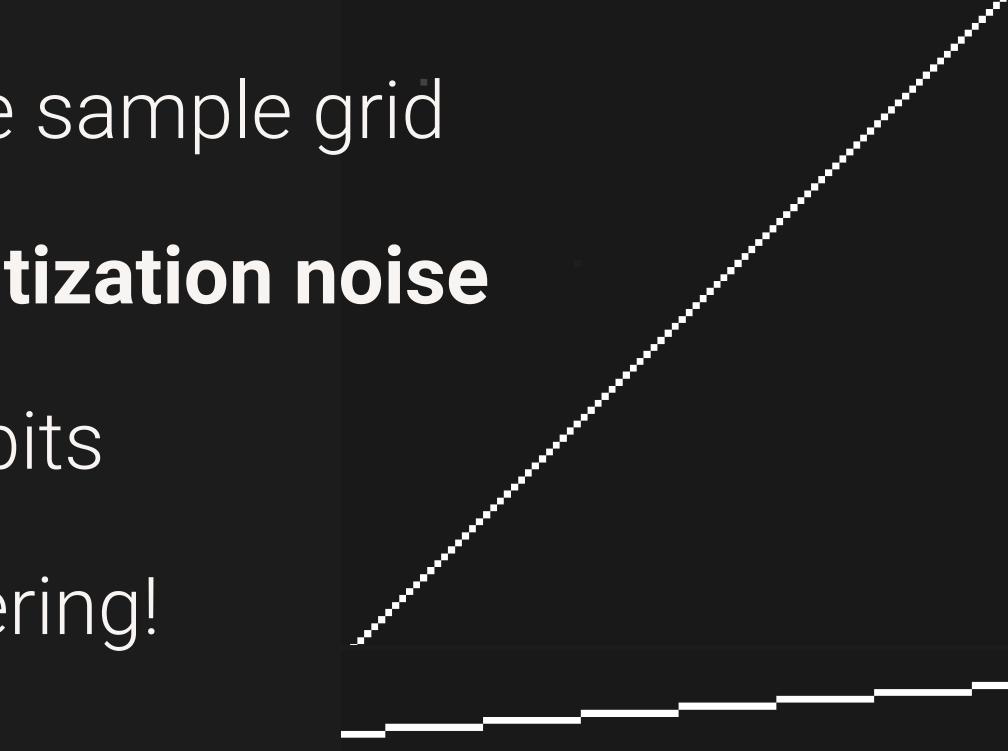
DAW Range: -1.0 to 1.0 (floating point, 32- or 64-bit) Asset Range: 16- or 24-bit integers Higher bit depth == more precise samples

- Loudness is movement, so quiet means less resolution



Quiet Considerations: Quantization

All digital audio "snaps" to the sample grid Diff from real to snap is **quantization noise** This is why we mix in 32/64 bits This is why we invented dithering!





Loudness vs. Dynamic Range

Loudness describes one window of sound in time Dynamic range compares at least two windows ... or, in the case of bit depth the quietest and loudest signals possible



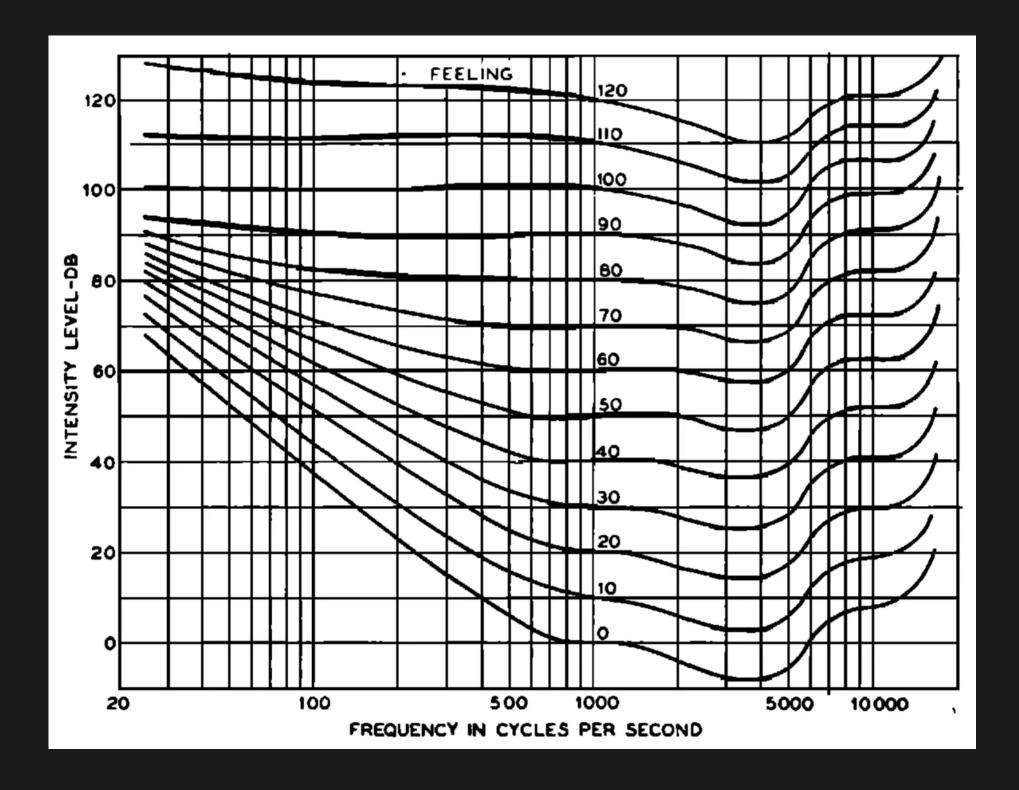
Loudness vs. Dynamic Range

Decibels is the measurement for loudness Relative unit of measurement, logarithmic, 10 dB \sim 2x dBFS relative to 0 as max peak to peak movement 16-bit: 96 dB, 24-bit 144 dB theoretical, ~130 max real Dynamic range in vinyl is 50-70dB, usually closer to 60 on a decent system



Human Concerns: Contours, Pink Noise, etc.

Different frequencies are perceived differently Pink noise vs. white noise Fletcher Munson ISO 226

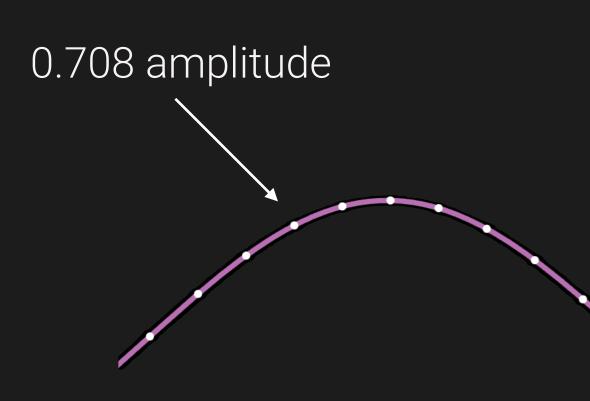




Measurements The language of the "thief of joy"



Amplitude is a raw sample value, decibel is a perceptual log-scale value



To convert to dBFS:

20 * log10(amplitude)

Amplitude vs. Decibel Full-Scale





now on for brevity. But I mean dBFS.

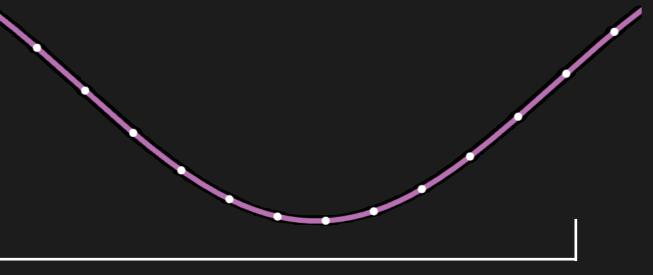
I'm going to use dB instead of dBFS from



1.0, or 0 dB (amplitude)

<aggregation & math> 0.708, or -3 dB (loudness)

Loudness vs. Amplitude





The immediate value of the sample farthest from 0

Peak

-3 dB (peak)

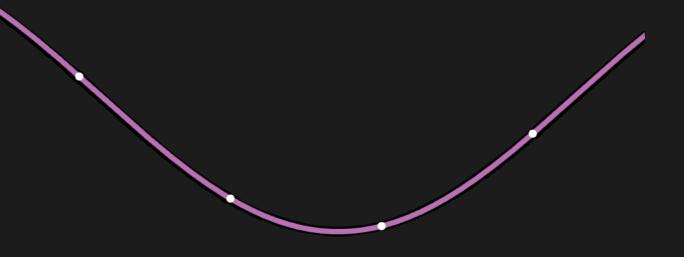




Intersample Peak

The theoretical actual peak amplitude after reconstruction

0 dB (peak)





Standardized intersample peak Defined in ITU-R BS.1770

"True" Peak

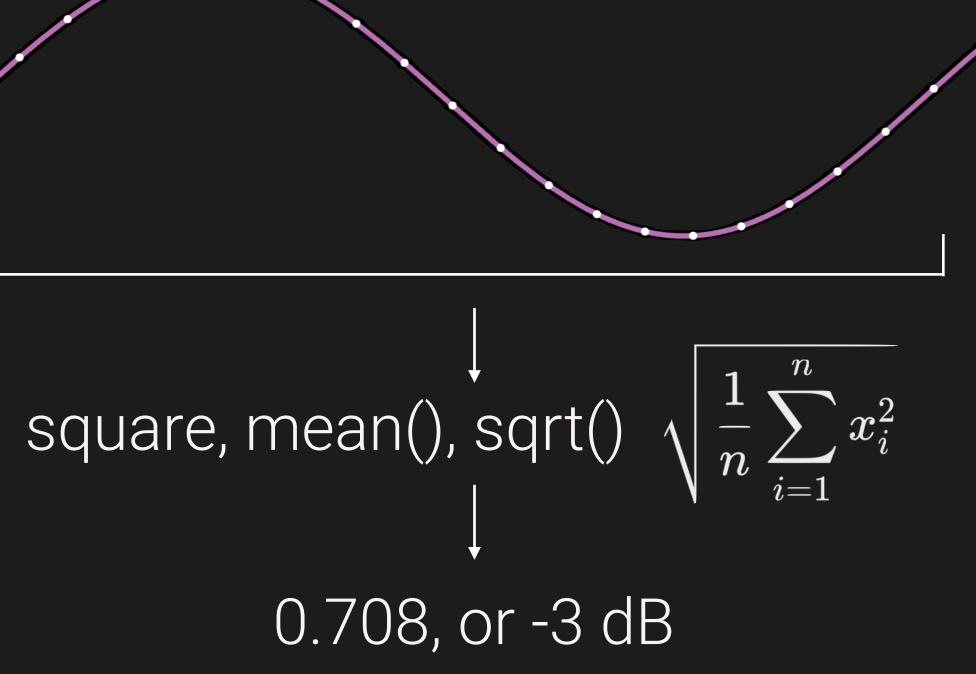
- 4x oversampling w/ specified filter or better
- I did a talk about oversampling: <u>https://www.youtube.com/watch?v=yJrJi-4SDUQ</u>



RMS: Root Mean Squared

The classic loudness measurement

300 ms window (VU meter)





Ignores very quiet or silent material Extended to multichannel loudness*

*but Atmos object-based mixing must be rendered out to 5.1 before measurement, or for bed-based, you've got some calculations to do depending on the output system, defeating the purpose of the media format to begin with: https://www.itu.int/dms_pubrec/itu-r/rec/bs/R-REC-BS.2051-3-202205-I!!PDF-E.pdf

LUFS

- Most popular modern loudness measurement Perception-based (frequency loudness curves)



LUFS / LKFS

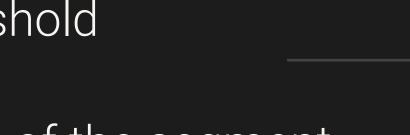
- "K" filter: high shelf ~1.2 kHz, low cut ~100 Hz 1.
- Break into 75% overlapping blocks of 400 ms 2.
- Square, mean, convert to LUFS for block 3.
- Discard blocks under absolute threshold of -70 4.
- Avg. remaining blocks, subtract 10 for relative threshold 5.
- Discard all blocks under the relative threshold 6.
- Avg. remaining blocks for the LUFS value of the segment 7.

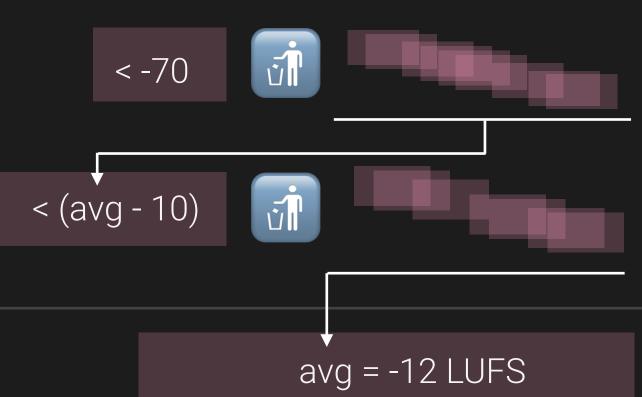














LUFS Variants

LUFS-M / Momentary: a single 400 ms block LUFS-S / Short-term: over 3 seconds (usually) LUFS-I / Integrated: over a whole piece of media



Loudness Range (LRA)

The macro-level dynamics of a signal Calculate the LUFS difference between them!

- After discarding blocks below thresholds, find:
 - •The loudest block in the quietest 10th percentile
 - •The quietest block in the loudest 95th percentile



Peak to Loudness Ratio (PLR) & Crest Factor

The micro-level dynamics of a signal (punctuation) PLR: True peak / Integrated LUFS Crest: Peak / RMS (usually)



Limiters, Clippers, Compressors oh my!



Characteristics of Loudeners

Do transients get brighter or darker? Does it "pump" or "fart" when pushed? Does it re-balance the mix?

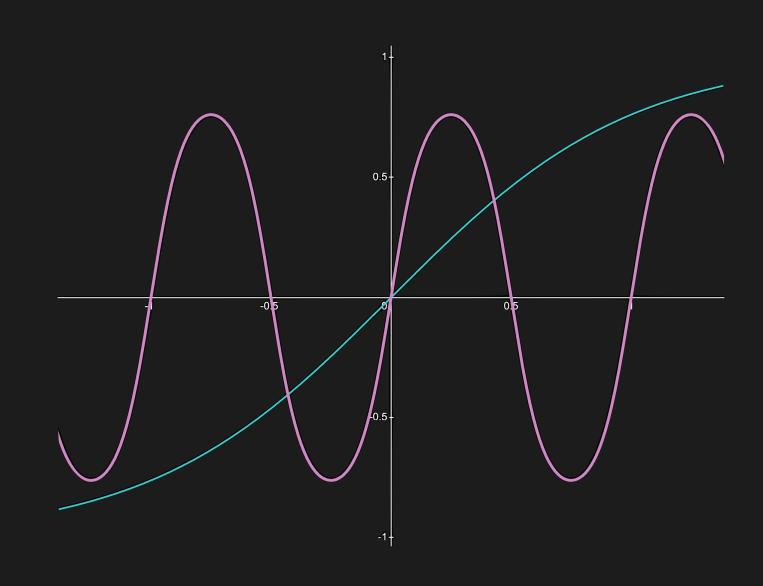
- Does it add saturation, when, and how much?



Transfer function Adds distortion Odd-order harmonics are cleaner (3rd)* Nonlinear within normal range/aliasing

*See my talk for the <u>DSP Online conference</u>

Saturator



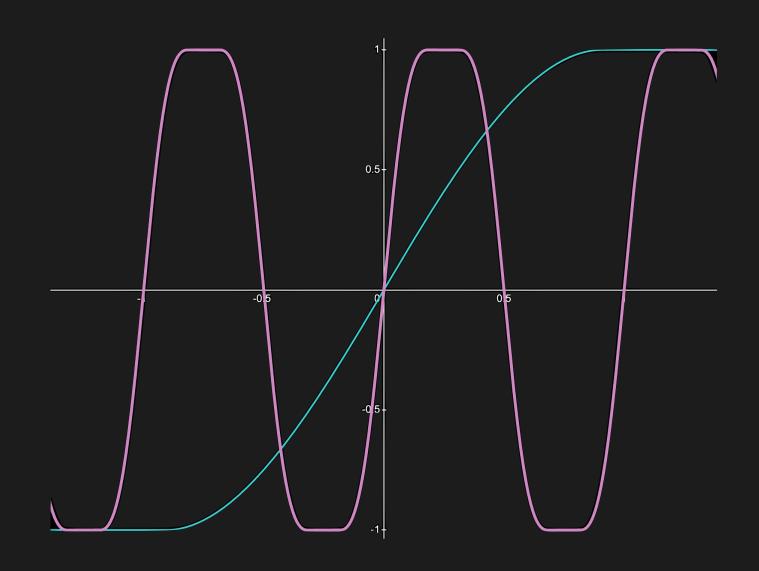






Transfer function Similar to a saturator, hard limit Nonlinear within normal range/aliasing

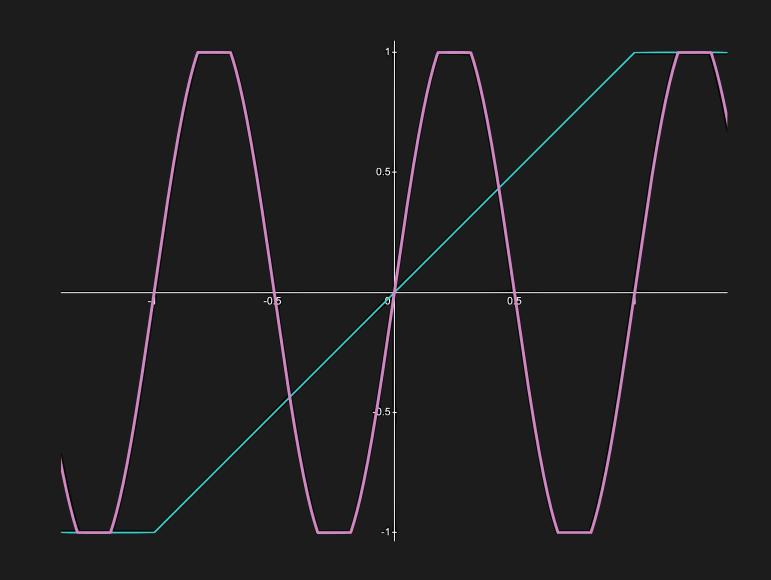
Soft Clipper





Hard Clipper

Transfer function Linear within allowable range (-1,1) Sudden distortion burst on overs Bursts can "hide" inside transients





Linear within threshold, non-linear outside The longer outside the threshold, the more intense Lots of decisions with wide impacts on sound:

Log or linear domain Peak vs RMS Knee shape and type

ballistics smoother type attack and release times

Compressor

- Ballistics section gradually applies/releases gain reduction

 - feed forward or backward

lookahead amount max dB reduction every curve tunable



All design decisions from compressor slide Extensive use of lookahead and fast attack times Release tuning is the really exciting part

Limiter

- Basically, a compressor that prevents clipping (but still flatlines)





Multiband Compressor

Compress each band, then rejoin Alters the character of the mix when it gets loud Another variation on the idea: dynamic EQ

- Linkwitz Reily (or similar) crossover to split into bands









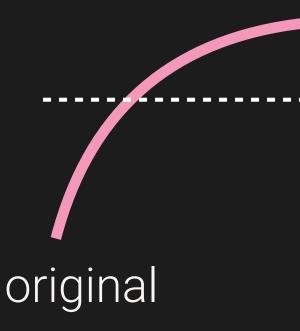
There are lots of ways to make sound louder They do not all sound the same Some are more transparent, some are more creative All of them are **TOOLS**

All this to say...



Limiters "duck" volume change Zero attack either hard or soft clips (hard or soft knee)

hard knee



About Lookahead

- Lookahead allows ducking **before**, retaining some shape





The Loudness Wars If you can't beat 'em, look ahead



Mastering/Finalizing audio for the format (cassette, vinyl) Limitations on volume for device tolerance

Pre-CD



Perfect replication for distribution Peak-normalization paradigm for compatible delivery Less variability on playback (until amplification stage)

- Well-defined dynamic range and frequency characteristics



Lookahead Limiters

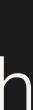
Peak-normalization's only limitation is peak below 0 Engineers start to push boundaries for radio dominance

- Lookahead meant peaks could be ducked without poke through
- But also: radio is already compressed, engineers want control











Sreedhar / IEEE Spectrum / "The Future of Music"

Criticized modern music production, not able to "breathe" Focused on a flat waveform; visual indication in micro dynamics Acknowledges listening environment as drive for loudness Perplexingly mentions higher-bit-depth formats as a solution?*

*"Audiophiles looking to the future for relief from overcompression see a cloudy picture. DVD-Audio and Super Audio Compact Disc (SACD) are two high-fidelity formats that were thought to be solutions to the loudness war. Both formats offer not only a greater dynamic range than CD but also higher sampling rates." - Suhas Sreedhar, <u>The Future of Music</u>



Sentiments in "The Future of Music"

"Even though you love this album, you can't listen to it anymore. You shut it off, tired, puzzled, and confused."

"... it could be responsible for halting technological advances in sound quality for years to come."

"Not only is all impact lost, but the constant level of the sound is fatiguing to the ear"



Deruty / Sound on Sound

Uses LRA instead, criticizes Sreedhar for focusing on crest Says macro dynamics has not changed Extensive research on old tracks and new





lan Shepard / Bob Katz

Suggests using PLR instead of LRA Started "Dynamic Range Day"

Deruty is wrong; we are explicitly talking about micro dynamics

They're right about Deruty's argument, but not necessarily right



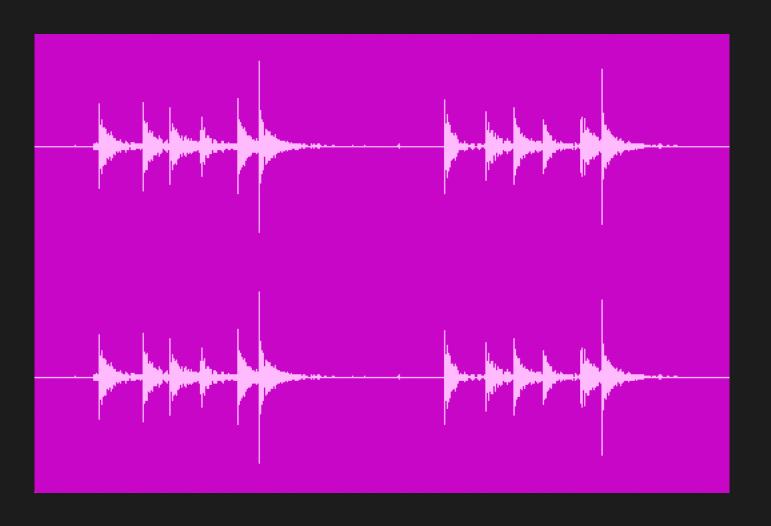
THEN WHO IS RIGHT?

You can't use measurements to insist on what's better Just look at John Atkinson in Stereophile reviews! Remember Sreedhar's thought about listening environments?

- Also, loudness is a creative decision, like harmony or dissonance



Original Sample



https://www.cambridge-mt.com/ms/mtk/



+6 dB, gain matched

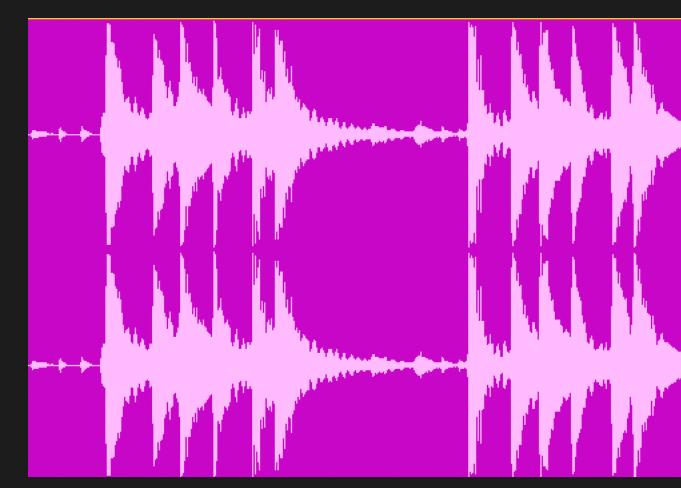




+12 dB, gain matched

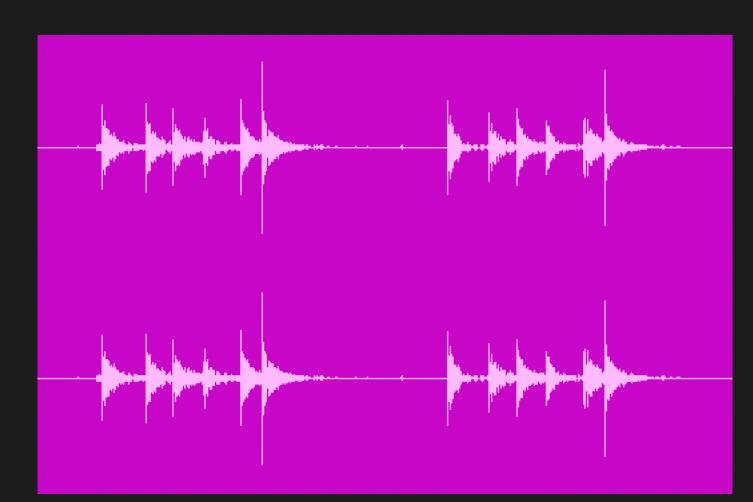






+12 dB@rail@nanatyched







I never did anything because it had to be louder. Ever. Everything I did was always because 'it's not feeling right'.

-Andrew Scheps

https://www.musicradar.com/news/andrew-scheps-mixing-metallica-adele-chili-peppers



The Modern Era: Broadcast Standards EBU-R128, ITU-R BS.1770, AES-R7, AND SUPPLEMENTS





Who Are These Agencies?

EBU - European Broadcast Union (Europe) AES - Audio Engineering Society (Private / US) ITU - International Telecommunications Union (United Nations)



Inconsistent loudness in program material annoys listeners Applies to broadcasters of all types:

- TV
- Radio
- Streaming

The Problem





The Problem as Stated in EBU R 128

a) that peak normalization of audio signals has led to considerable loudness differences between programs...

b) that the resulting loudness inconsistencies... are the cause of the most viewer/listener complaints;



The Solution per EBU 128 R

Push quiet material up (with exceptions) Push loud material down

- Normalize broadcasted audio to -23 LUFS (+/-1) Limit broadcasted audio to -1dB True Peak (+/- 0.3)



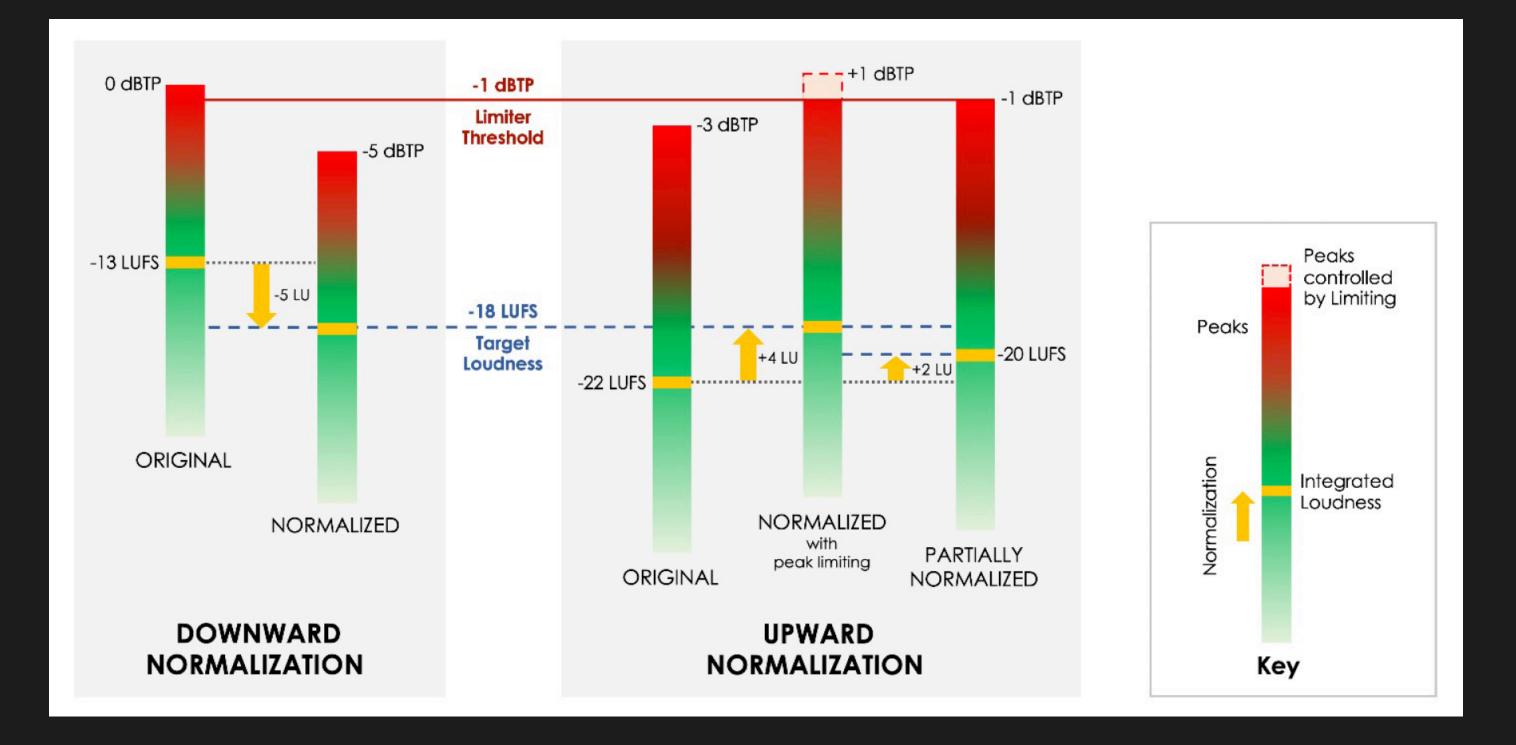
Why -23?

One number to rule them all (simplicity)

Support pre-existing and varied material without normalizing up!



Why is Normalizing up Bad?



AES77-2023, Processes for Downward and Upward Loudness Normalization



I Still Don't See -14 LUFS

EBU R 128 s2: Streaming Boogaloo Personal music players: low gain, low SNR Listening environments: high background noise Aim for -23 LUFS when players have better amplifiers

- Allow -20 to -16 LUFS, TP limiter for upward normalization



I STILL DON'T SEE -14 LUFS!

AESTD1008 & AES77 ARE YOU HAPPY NOW?

- Track vs. Album Normalization





Service	Target
Spotify	-14
Spotify Loud	-11
Apple Music	-16
YouTube	-14

Adaptation

Upwards?	Documented
TP -1 dB no limiter	Yes
Yes, limiter	Yes
TP -1 dB no limiter?	No
???	No



The Hunt for Artistic Relevance

"Dynamic compression is again an artistic tool and not a loudness weapon — the audio quality increases!"

"Loudness variation is an artistic tool, and the concept of loudness normalisation according to R 128 actually encourages more dynamic mixing!"

"...the actual technical change of the audio signal level through active normalisation to -23 LUFS has direct influence on the artistic process — and in a positive way! The production side is thus relieved from fighting the 'loudness war' – an unfortunate result of the peak-normalisation paradigm."

-EBU Tech 3343 GUIDELINES FOR PRODUCTION OF PROGRAMMES IN ACCORDANCE WITH EBU R 128







k Lotus Audio

💋 Ryan Schwabe

🔞 Morningdew Media

🔇 Klangfreund

💴 Logic Studio Training

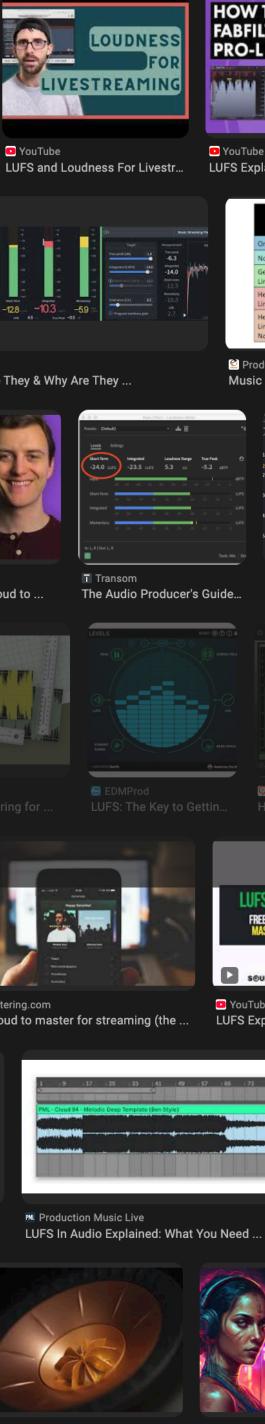
🗠 www.loudnesspenalty.com

🐼 Audio Audit

w Wix.com

iZotope

🗐 Worship Sound Guy



Sound Me

Where we are Today

They contain true peaks > 0 dB

One research example: <u>https://www.izotope.com/en/learn/mastering-trends.html</u>

- Billboard chart tracks (cross-genre) are -11 to -6 LUFS
- They sound good (dare I say great), plenty of impact
- The advice doesn't stick artistically or commercially



... but why did it get here?

These rules were designed to solve a **broadcasting** problem Scope creep and hubris did not help anyone Marketing & product mgmt. targeted an ill-defined problem There was a failure to remember that...



Measure Does not Define Experience NUMBERS ALONE ARE MEANINGLESS



"Reason is, and ought only to be the slave of the passions, and can never pretend to any other office than to serve and obey them."

- David Hume, A Treatise of Human Nature

Allan Ramsay, Portrait of David Hume, 1754



Fix Complaints from the Real World

...and they have every right to their own passions; don't impose them as gospel

- The public complained about *inconsistent loudness!* Did the public complain about **poorly produced art**? ...about *impure transients*? ...about *limp impact*? No, these were the passion of a (still) vocal minority!



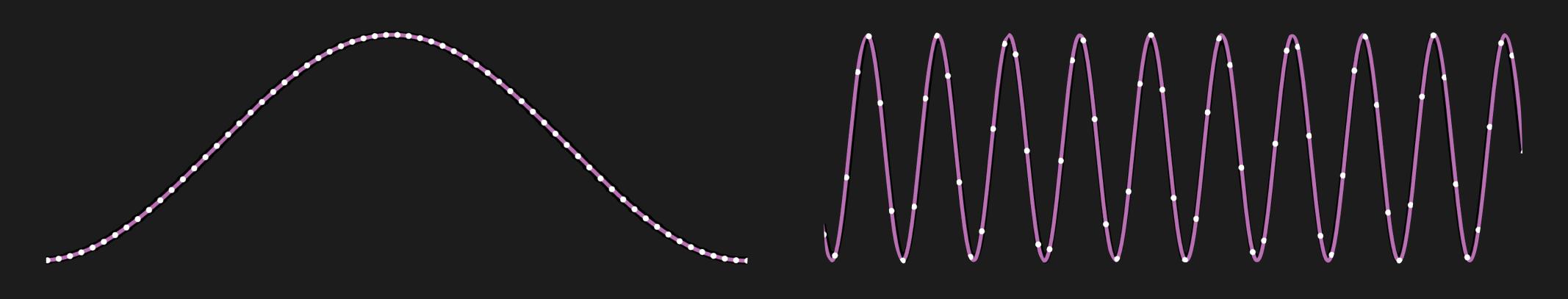
Accuracy and "Realism" != Better Production

Productions with this goal *never* fought the loudness war, anyway.



True Peaks: The Untold Stories

AES R-7: True peaks only really matter for highs/transients (Transients are short noise bursts, so already mask peaks heavily)





True Peaks: The Untold Stories contd.

AES R-7:

TP meters still have a max under-read up to 0.6 dB AES 77-2023-F:

"...true peak limiting more than about 1dB may produce more audible artifacts than simply letting audio material clip"



True Peaks: Codecs

Lossy codecs don't clip on encode Lossy codecs don't inherently clip on decode You can always listen... https://www.apple.com/apple-music/apple-digital-masters/

- Volume normalization with FP decoder = HEADROOM
- Are occasional clips more audible than the "lossiness"?



In Conclusion, Listen The best advice is trite



Dynamic Range, PLR

Dynamic range by itself is not a virtue

- Not all dynamic range compression is the same
- Perceived dynamics/impact > measured dynamics



Loudness is Robust & Predictable

- Sounds great; does not necessarily lack impact Better listenability in noisy environments No upwards normalization/standards safe
- Less extreme peaks, lower amplification required
- Lower amplification == less analog nonlinearity
- More predictable & listenable in real world systems



No man is wise at all times, or is without his blind side.

-Desiderius Erasmus, <u>In Praise of Folly</u>





THANK YOU!

