

# Mastering, Compression & Codecs



Technology for Performers

# The Music Production Process: An Over(re)view

- Pre-Production
- Recording (Production)
- Post-Production
- **Mastering**

# Mastering

- Arriving at 'final' mixes
  - Before we send our work to be mastered, we need to be sure that it is as good as it can be
- How?
  - Export mixes from Pro Tools
  - Listen critically on different systems (car, different headphones, speakers, laptop)
  - Tweak/refine mixes if necessary
  - Repeat until satisfied

# Mastering

- Once mixes are finalized, they are sent to a mastering engineer to have the finishing touches put on them for release
  - This process happens on a much shorter time-scale than mixing (typically one album per day)
- A good mastering engineer offers a few benefits to a project:
  - Fresh ears and a fresh, unbiased perspective on the project
  - A facility which is optimized for hi-quality stereo monitoring and processing



# Mastering

- ‘Finishing touches’ include:
  - Pacing
  - Denoising
  - Subtle EQ
  - Dynamic Range Processing
  - Adding Metadata

# Mastering

## **Pacing**

- Determining the time between tracks on an album

# Mastering

## **Denoising**

- Removal of hum, hiss, or any other distracting noises in a recording (chair squeaks, coughs, etc.)
- May also be done during the editing phase of the production process instead of at mastering
- Common software for denoising: iZotope RX, Sonnox Restore Suite

# Mastering

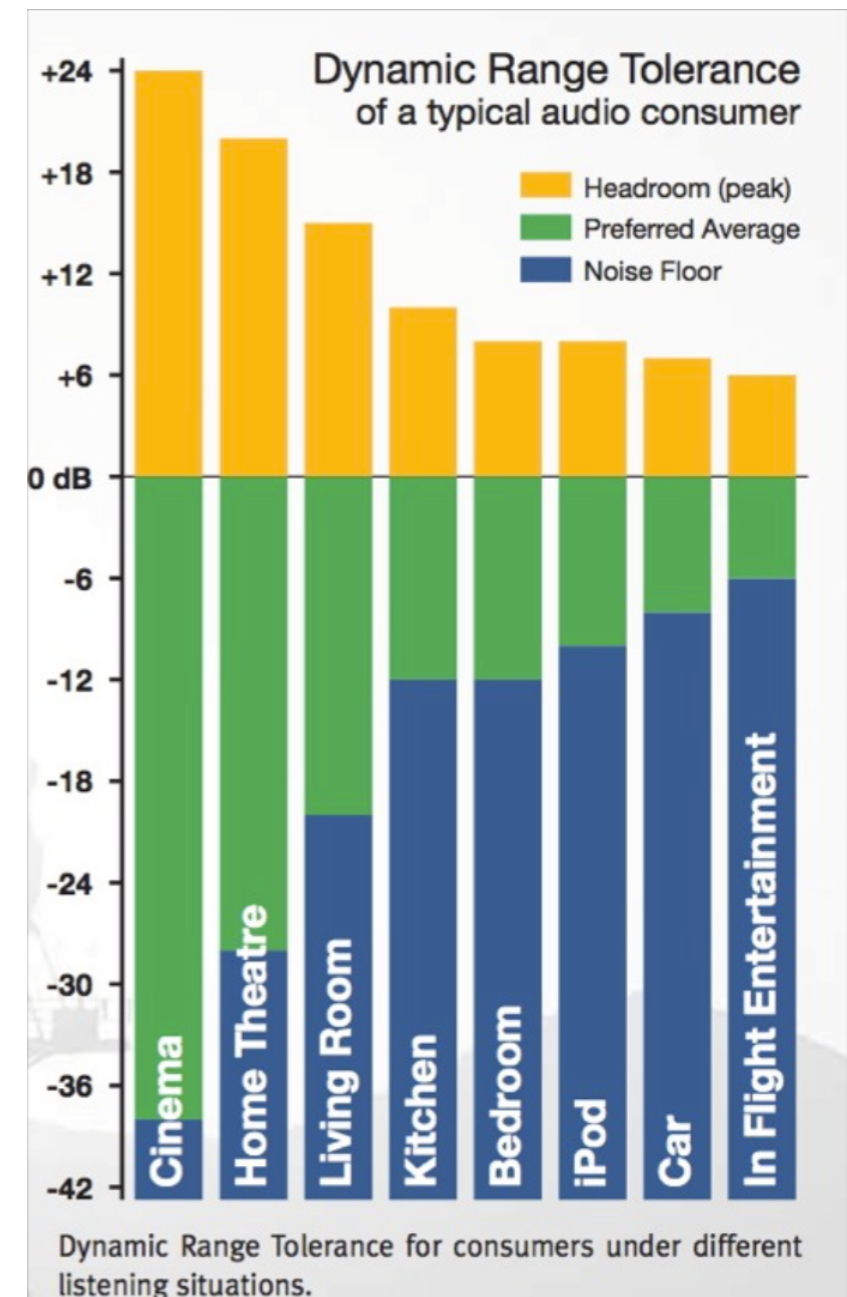
## **EQ**

- For arriving at an appropriate/even spectrum for the program material
- For matching tracks that were mixed in different control rooms or compensating for uneven monitoring environments (i.e. whole album has too much low frequency emphasis)

# Mastering

## Dynamic Range Processing

- Compression/Limiting to get the album to a reasonable loudness/dynamic range for the intended playback medium
- Different for Digital (CD), Vinyl, and specific playback environments
- Compression & Limiting???



Compression Tangent!

# Compression

- A **compressor** is a device that automatically reduces ('compresses') the dynamic range of an input signal
- When the input signal rises above a **threshold**, the device turns down the signal at the output
  - Traditionally used to protect equipment from clipping/overloading
  - Now a common effect used in both practical and creative ways



# Compression

- Compressors typically share a basic feature set:
  - Threshold
  - Ratio
  - Attack
  - Release
  - Makeup Gain

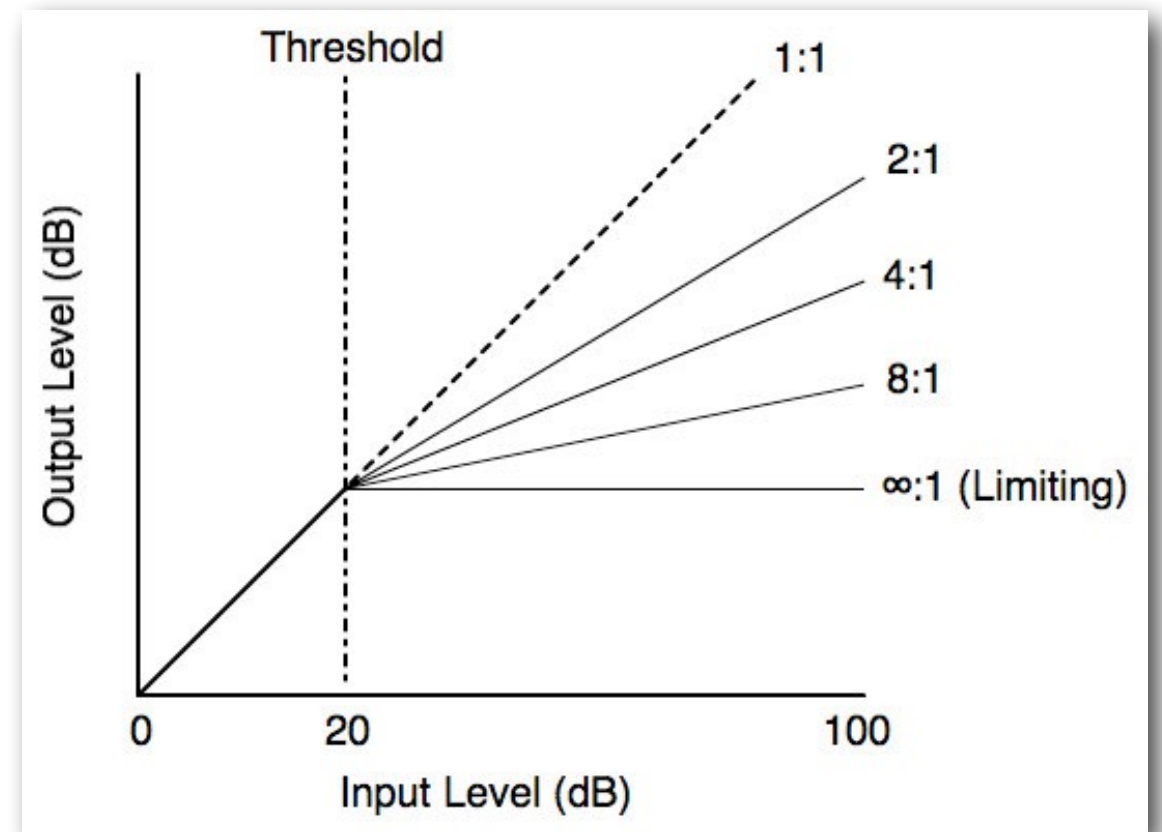
# Compression

## Threshold

- How loud an incoming signal needs to be before the compressor engages

## Ratio

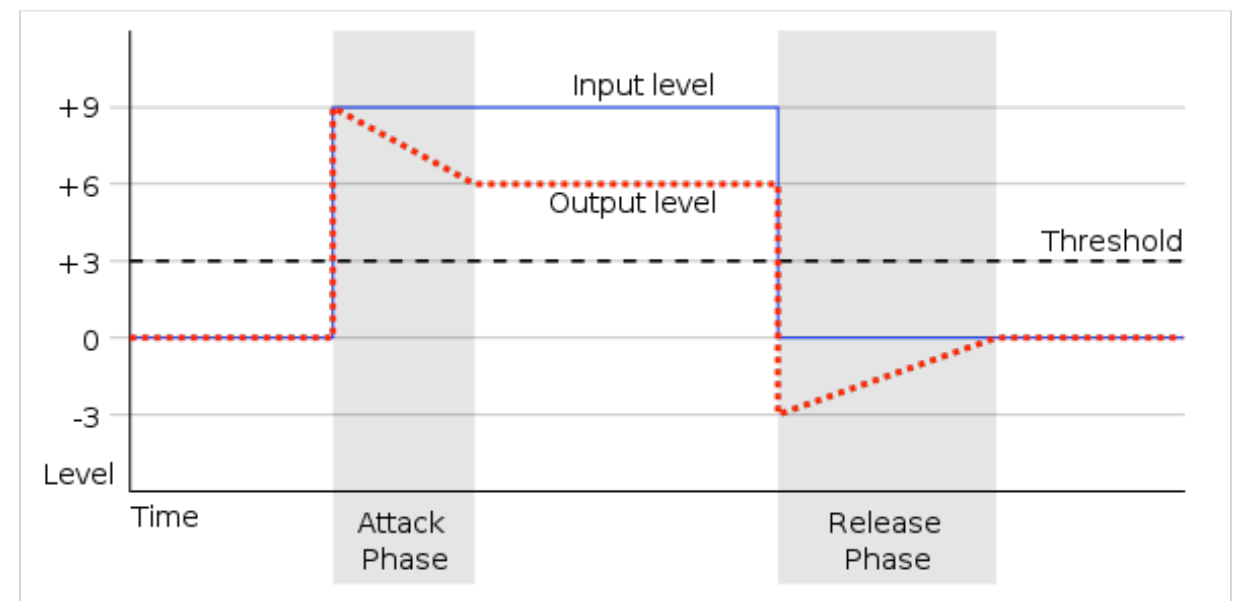
- How much the output signal is turned down after the input signal exceeds the threshold
- Expressed as a ratio of input vs. output (2:1, 3:1, 4:1, 10:1)
- Anything above 10:1 is considered **limiting**



# Compression

## Attack

- How long it takes the compressor to fully compress the signal after it exceeds the threshold
- A 'fast' attack is good for attenuating transients
- A 'slow' attack lets transients through



## Release

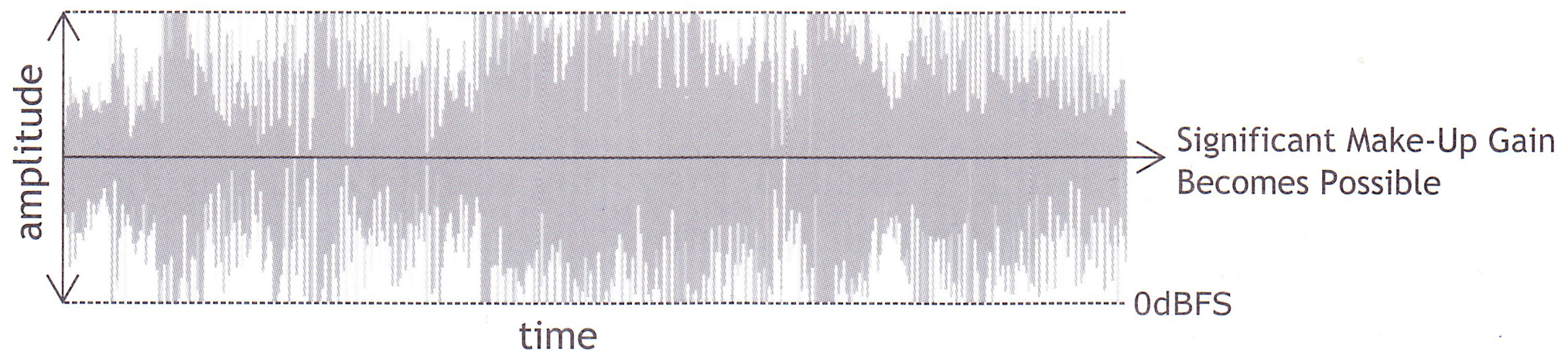
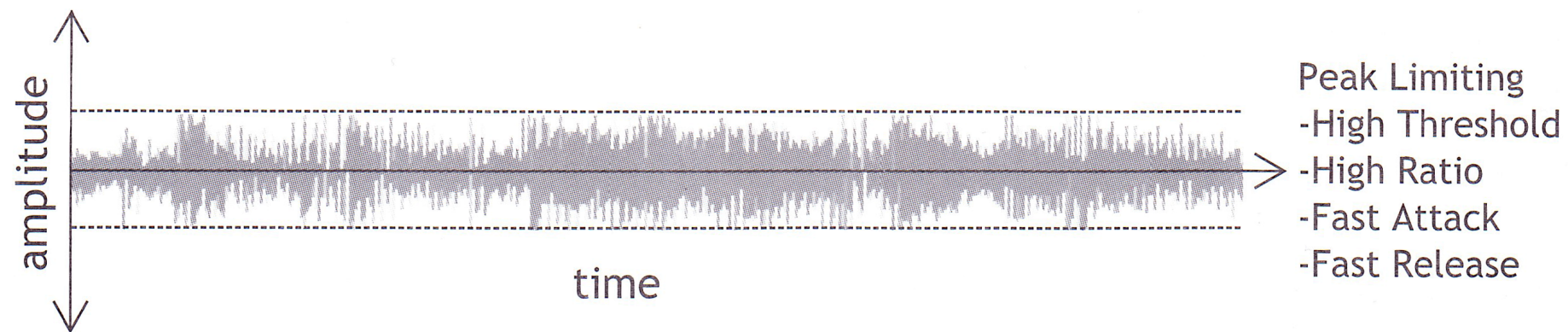
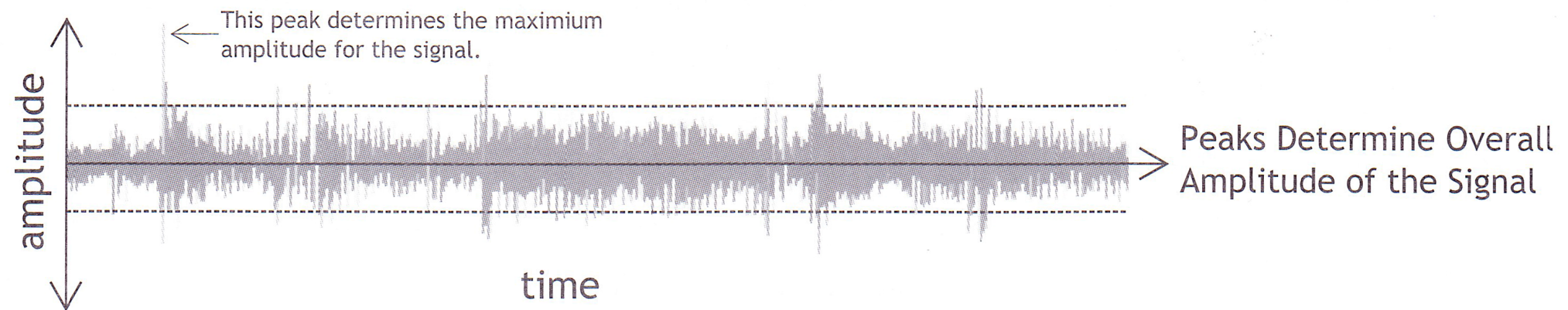
- How long it takes the compressor to return to unity gain after the input signal returns below the threshold
- Closely tied to tempo - release times allow for recovery between notes, phrases, etc.

# Compression

## **Makeup Gain**

- Often just labelled 'Gain'
- Increases the output signal, post-compression
- By 'chopping-off' the loudest part of a signal, we buy ourselves 'headroom' to turn the remaining signal up

# Peak Limiting



# See it in action!

- Let's watch Mastering Engineer Mark Wilder work on a track...
- <https://www.youtube.com/watch?v=pHsRK0In4VQ>

# Mastering

- ‘Finishing touches’ include:
  - Pacing
  - Denoising
  - Subtle EQ
  - Dynamic Range Processing
  - **Adding Metadata**



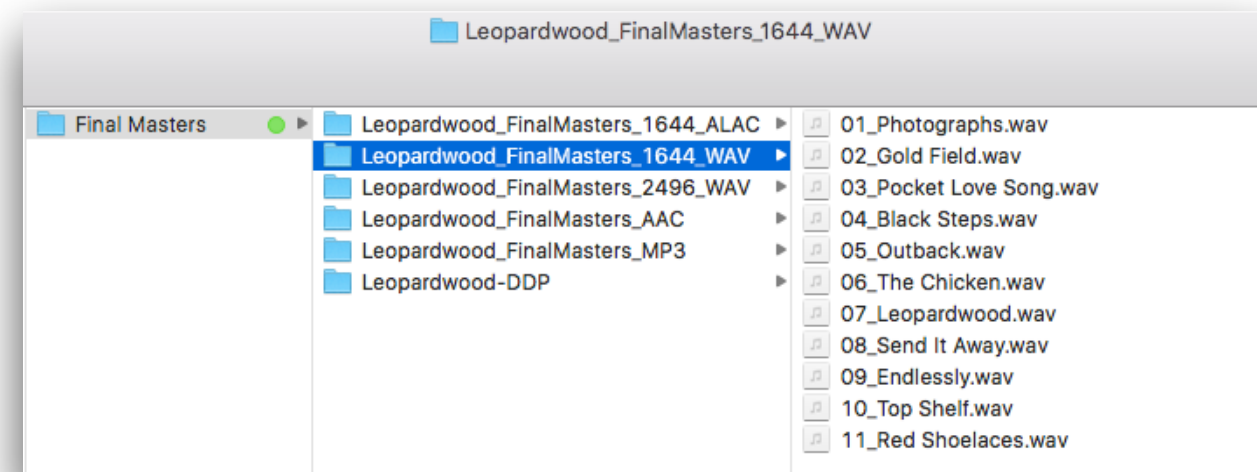
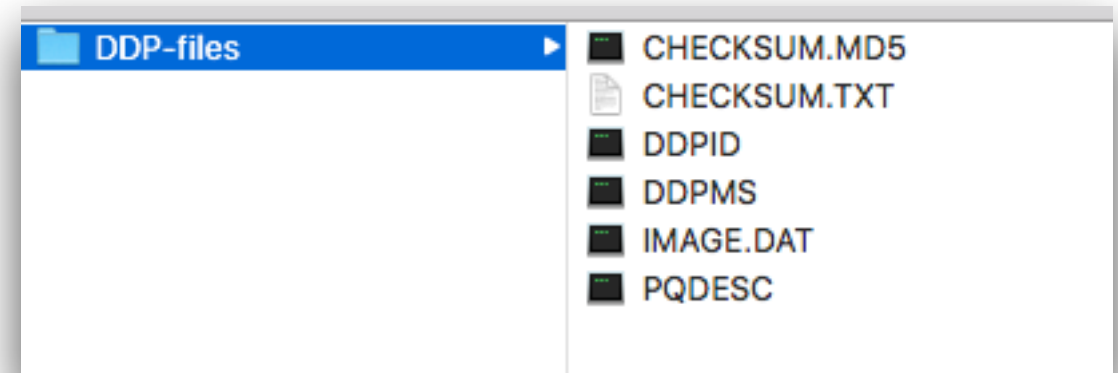
# Mastering

## Metadata

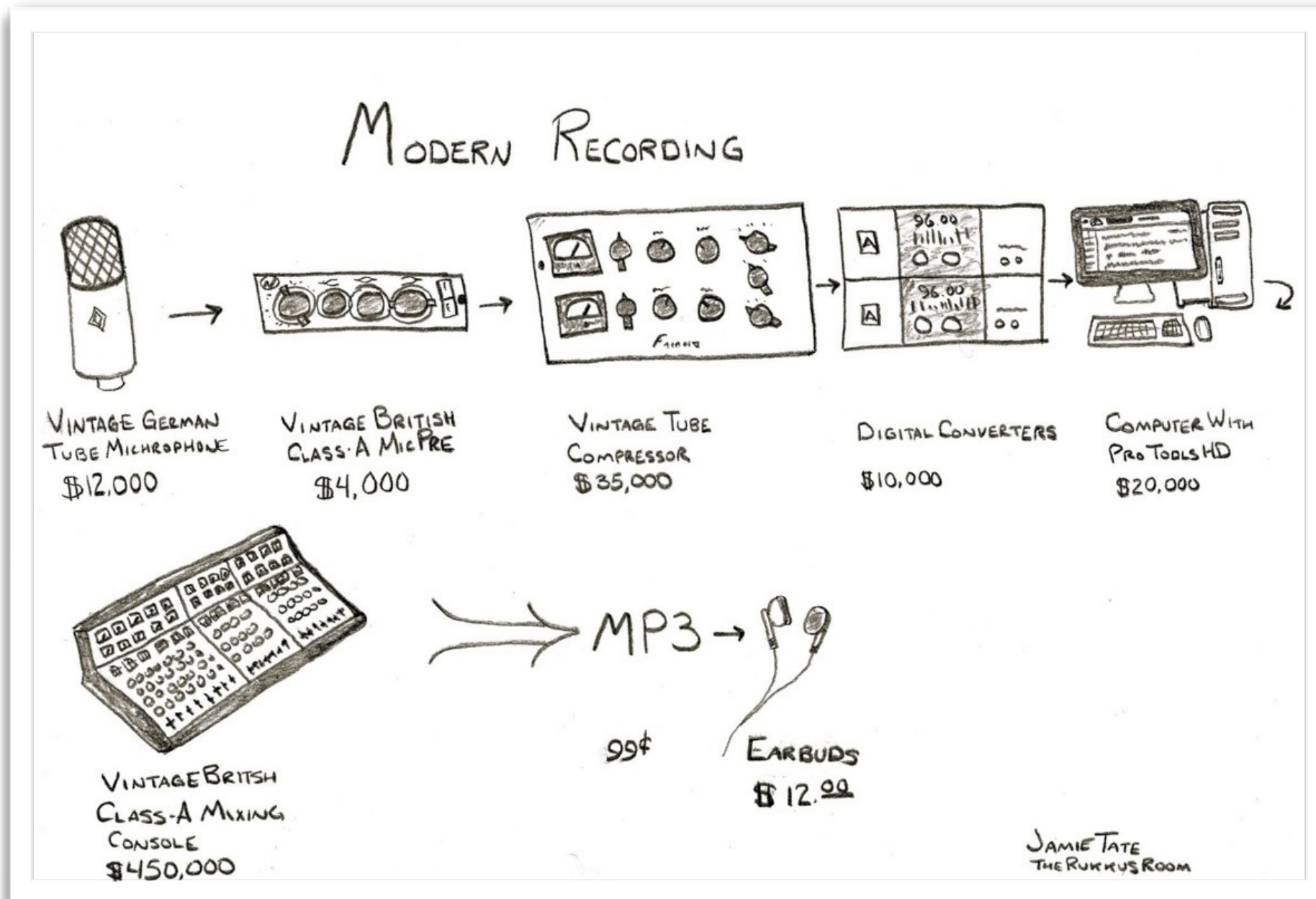
- Information about the music that is stored *in* the files for each track
- ISRC codes, Track Number, Disc Number, Artist Name(s), Album Name, Song Names, Composers, Album Artwork - anything that might be useful
- Different audio codecs can store different metadata
  - More on audio codecs in a moment...

# Mastering

- What you get *back* from mastering:
  - **DDP** - Disc Duplication Protocol
    - Folder containing a specific set of files (see image)
    - Album can be previewed via DDP player (software) for approval
    - Can be easily verified for data-integrity
    - This what you send off to a professional CD manufacturing facility, not a CD-R
  - Full Quality **WAV** files
  - Other file types on request (ALAC, MP3, etc.)
    - Of course, all you really need is the full quality copy and you can make your own versions from that as needed



# Audio Codecs



# Audio Codecs

- What's a codec?
  - Stands for **C**oder + **D**ecoder
    - We encode the file to reduce (compress) its size
    - We decode it to play it back
- Consumer audio codecs:
  - MP3, AAC, OGG Vorbis, WMA, FLAC, ALAC

# Consumer Audio

- Today most of the audio we hear was encoded with an audio codec
- These formats are not for audio production, only for consumer delivery.



# Audio Codecs

- Why bother to encode?
- Full quality files are big. Some common concerns:
  - Internet bandwidth (downloads or streaming) - though this is becoming less of an issue
  - Storage limitations (iPods, phones, portable media players, gaming consoles, etc.)
- How big of a difference?
  - Hi resolution (WAV @ 96 kHz, 24bit) 1 hour of stereo = 2 GB
  - CD quality (WAV @ 44.1 kHz, 16 bit) 1 hour of stereo = 620 MB
  - iTunes store (AAC @ 256 kbps) 1 hour of stereo = 112 MB

# Audio Codecs

- There are two types of Codecs:

## **Lossy**

- Codecs that eliminate data to reduce file size
- MP3, WMA, AAC, OGG Vorbis

## **Lossless**

- Codecs that reduce file size while preserving all data
- ALAC, FLAC



# Lossy Codecs

- Lossy Codecs use **perceptual coding** to decide what data to throw away
  - Based on studies of *psychoacoustics* (our perception of sound)
  - Takes advantage of the limitations of human hearing, in particular the ability for one sound to ‘mask’ another from our perception (in either the time domain or the frequency domain)
- What is thrown away?
  - Generally, low level details. Less bits are assigned to elements that are more likely to be masked, and more bits are assigned to elements deemed more perceptually relevant
- Once you encode a file to a lossy format, this information is gone forever.

# Lossless Codecs

- Reduce file size without throwing anything away
- This is accomplished by looking for redundancies in the code and replacing them with placeholders which take up less space
- On playback, the swap happens in reverse and the original material is presented without any loss of data
  - Grossly Simplified Example:
    - 11111111111111111111
    - 1(20)

# Photography

- Images can be stored as **lossy** or **lossless** file types just like audio
  - Some typical photo file types:
    - jpg, gif
    - These are **lossy**!
- When we try to save space by making images smaller, we get subtle changes in the quality of colour and light.
- In audio this would be similar to subtle changes in high frequencies or spatial content.



High resolution (18MB)





High resolution .gif (2.9MB)





Medium resolution .gif (1.8MB)





Low resolution .gif (1.1MB)





# Good Practices

- The morals of the codec story...
- When working on a project, it is important that you keep your assets at full quality as long as possible.
  - Once you go to a lossy format, you can't go back.
- Lossy copies should be made from the master copy and the master copy should be retained. Do not transcode lossy codecs. If you need a different lossy format, go back to the original to make the copy.
- Don't use lossy-encoded audio in production work
- Be conscious of audio quality
  - Consider ripping CDs to ALAC instead of MP3
  - Look for high quality music online (Pro Studio Meters, Bandcamp, HD Tracks)

# Listening time!

- Let's see if we can hear the difference...