# Intelligent implementation

Keeping the Program Lead off your back

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GDC

### Introduction

- Started in games 1990
  - Sega Genesis:
    - •6 voices of FM
    - •5 voices and one voice 8 bit 11kHz sample

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- In 20 years
  - VR Worlds, jacked in
  - still fighting with simulation and graphics for resources.

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- Trade offs need to be made
  - Biggest bang for the buck

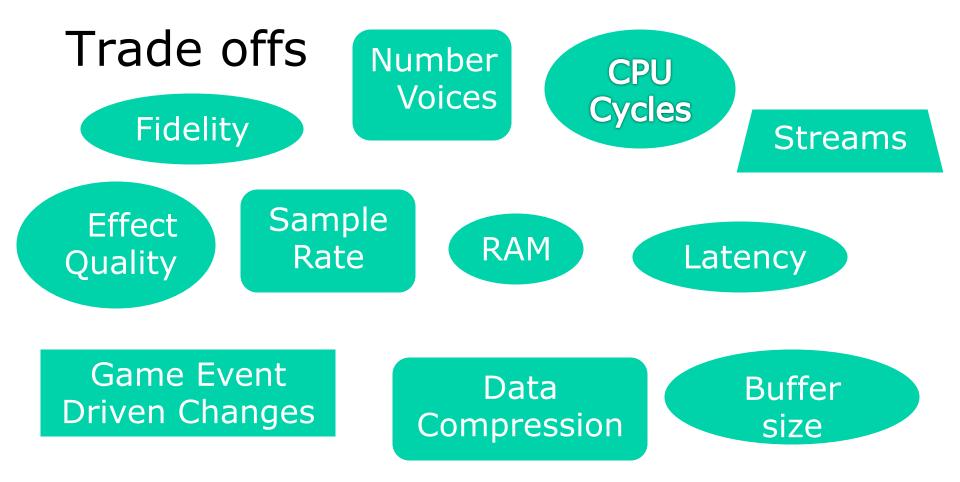
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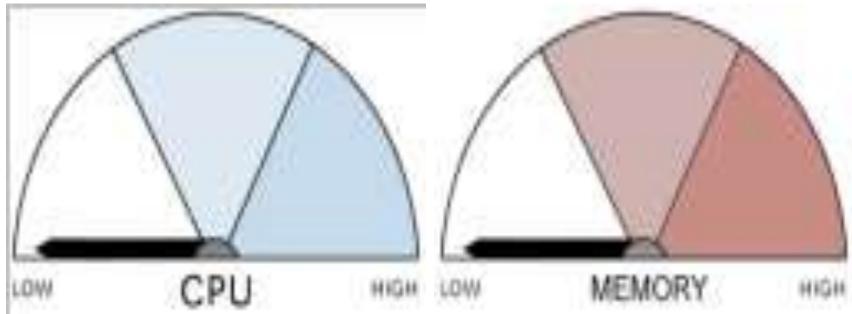
• Trade offs are different depending on the platform and game. Find the bottleneck and balance.

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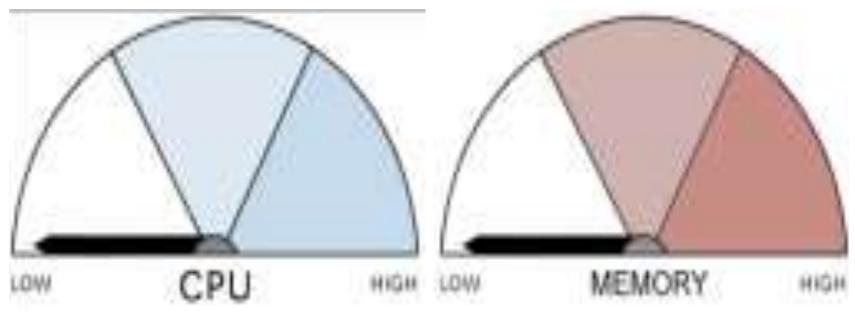
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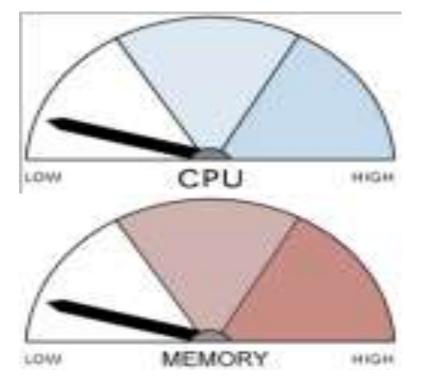
#### Volume



#### Surround Pan

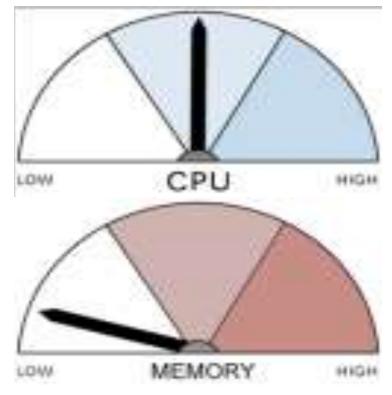


#### LoPass Simple

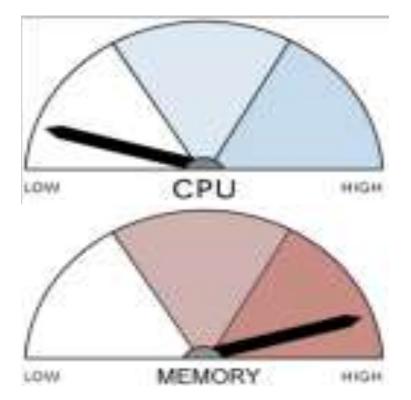


#### FMOD Lopass filter

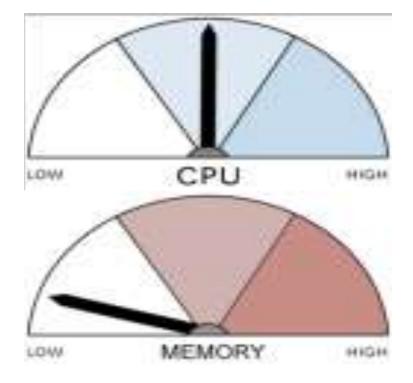
(analog resonance emulation)



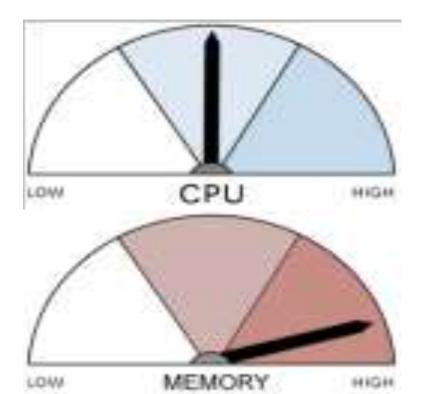
Echo



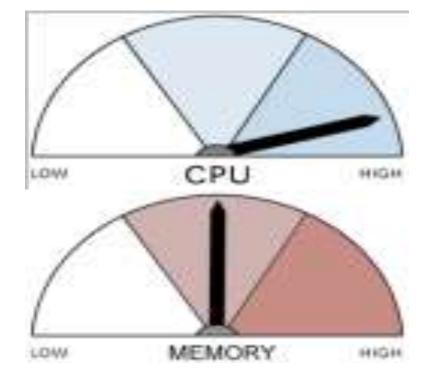
#### Parametric EQ



Chorus



#### FMOD SFX Reverb



- CPU cycles
- RAM
- Number of voices
- Hard Drive streams
- Optical Disk streams
- Latency/buffer sizes
- Sample rate/data compression/fidelity
- Game event driven changes
- Effect quality/CPU hit

- CPU Cycles vs RAM
  - Decompressing mp3 on the fly (runtime)
    - Versus
  - Decompressing mp3 into RAM

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- Number of Voices Vs CPU
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  - Each voice uses CPU cycles
  - More voices allow layering, more immersion

- Effect quality vs CPU hit
  - Higher quality EQ or reverb -> more CPU

 Bottleneck: Download under 20MB to go over 3G network – footprint potentially limited.

- Music compressed as AAC to make smaller
- iPhone can only decompress 1 AAC/mp3 stream at a time in hardware

- Result: SFX shouldn't be AAC, will affect frame rate.
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  - Or use IMA 4:1 compression

- Bottleneck: Size needs to be minimized
  - to reduce bandwidth costs and
  - Reduce download time

- Data reduction on audio assets
  - mp3,
  - Ogg Vorbis

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- Solutions:
  - decompresss into RAM before playing
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  - decompresss into RAM before playing
    (pre-cache in Flash)
  - use IMA 4:1 compression. Lighter CPU, less size reduction.

## Case study: Xbox 360

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- Bottleneck:
  - RAM
  - potentially DVD space
  - limited # of streams off of DVD

- XMA compressed files decoded on hardware
  - no CPU hit
  - Lots of voices

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  - Minimize streaming

• Game effects are optimized for CPU efficiency

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- External hardware available with DAW

• Better sound quality

- Better sound quality
- Much more variety in models for compressors, reverb, sweeter EQ's.

#### **General Rule**

 If it's not changing in realtime in-game, consider baking the effect in

#### Applying EQ or filters that never change to sounds

- Applying EQ or filters that never change to sounds
- Applying compression to a single sound source that never changes

 Applying distortion that never changes to a sound source to make it more audible

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 Using multiple layers of sounds nondynamically

- increases immersion
- provides realtime feedback to player

- Choose relevant game parameters to drive them:
  - Distance

• Choose relevant game parameters to drive them:

- Distance
- Force

• Choose relevant game parameters to drive them:

- Distance
- Force
- Speed
- Location

- dynamic EQ/filters fed by RTPC tied to game states
  - Filter or EQ on damage sounds, tied to force

- compression or limiting on the master bus
  - Can only be done realtime, keeps mix under control.

- Dynamic distortion
  - Trash distortion used in Forza 4, driven by load on engine.

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  - CPU load (of 1 thread, 1/2 core)
    - •Trash = 3 5 %
    - •FMOD = 1%

- Chorus or flanging effects best done at runtime
  - realtime chorus will help mask loop points, sounds more natural

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- Read heads only go so fast
  - Bigger the buffer, the more time available
  - More streams available
  - Higher latency

- Probably sharing with other game data
  - Textures
  - Geometry

- Speech and music typically streamed
  - Except when it isn't.

- Have to coordinate with lead programmer
  - Allocate resources as early as possible
  - Stream not available? Argue for more RAM

- Ambiences:
  - Try overlapping loops
  - Random one-shots

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  - Take up CPU tracking position and volume updates

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  - Music
  - Retrigger loops

- Use a Priority system to cull voices
  - Oldest
  - Quietest
  - Most important to least important

- intelligent instance limiting
  - Group sounds into categories, and limit the number that can play at once. (example: bird chirps, wind loops).

- Don't place individual sound points for ambient sounds (birds, insects)
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- Tracking all those points is CPU intensive
- designate an area where birds are heard and play with random positioning and volume

- Stream one voice to multiple locations
  - Example: Speakers around a race track
  - Saves on voice overhead

 Categorize your sounds, and put a voice limit on that category to avoid making your mix too dense.

# Questions?