

# ECS614U/ECS749P: Sound Recording and Production

Michael Terrell

`michael.terrell@eecs.qmul.ac.uk`

`http://qmplus.qmul.ac.uk/course/view.php?id=3243`

Centre for Digital Music  
School of Electronic Engineering and Computer Science  
Queen Mary University of London

Semester 1, 2013–14

# Mixing: Gain/Pan/EQ

# Gain Effects

- Gain based audio effects change the level of the signal.
- Gain based effects include:
  - Fader gain - change amplitude of a track.
  - Panning - applying different gains to different output channels.
  - Equalization - change the gain of parts of the frequency spectrum.

# Fader Gain

# Fader Gain

- Gain is used to set the relative amplitude of each instrument in the mix.
- It is the simplest control but is the most important.
- It is important to learn how to judge the relative levels in a mix (examples within lecture resources).
- The units of gain is decibels.
- *Crazy Scientist Video.*

# Decibels

# Decibels

- Decibels express the ratio between two quantities.
- The unit of decibels is dB.
- Decibels use the logarithmic function, and are defined with respect to power, P:

$$\mathbf{dB} = 10 \log_{10} \left( \frac{P_1}{P_2} \right) \quad (1)$$

# Why Decibels?

## Weber-Fechner law:

- Our perception of the just noticeable difference (JND) in the magnitude of a stimulus is proportional to the stimulus magnitude.

$$\text{JND} = k \log_e \left( \frac{S}{S_0} \right) \quad (2)$$



# Why Decibels?

**Easy to manage a huge range of values:**

<b>dB</b>	100	80	60	40	20	0
<b>P Ratio</b>	$10^{10}$	$10^8$	$10^6$	$10^4$	$10^2$	1

<b>dB</b>	0	-20	-40	-60	-80	-100
<b>P Ratio</b>	1	$10^{-2}$	$10^{-4}$	$10^{-6}$	$10^{-8}$	$10^{-10}$

# Why Decibels?

## It makes the maths easier:

- It is easier to add decibel values than to multiply by their equivalent ratios.
- This is particularly true for large changes, e.g. 100 dB!

# Decibel Values

What is the dB change if we double or halve the power?

- Double:

$$\mathbf{dB} = 10 \log_{10} \left( \frac{2}{1} \right) = +3.0 \text{ dB} \quad (3)$$

- Halve:

$$\mathbf{dB} = 10 \log_{10} \left( \frac{1}{2} \right) = -3.0 \text{ dB} \quad (4)$$

# Decibel Values

- What if we multiply the power by 8?

$$8 = 2 \times 2 \times 2 \rightarrow \mathbf{dB} = 3 + 3 + 3 = 9 \text{ dB} \quad (5)$$

- What if we divide the power by 8?

$$\frac{1}{8} = \frac{1}{2} \times \frac{1}{2} \times \frac{1}{2} \rightarrow \mathbf{dB} = -3 - 3 - 3 = -9 \text{ dB} \quad (6)$$

# Decibels in Pressure

- The power is proportional to the square of the pressure.

$$\mathbf{dB} = 10 \log_{10} \left( \frac{p_2}{p_1} \right)^2. \quad (7)$$

- Using a property of the log function, this gives,

$$\mathbf{dB} = 20 \log_{10} \left( \frac{p_2}{p_1} \right). \quad (8)$$

# Decibels in Voltage

- The power is also proportional to the square of the voltage.

$$\mathbf{dB} = 10 \log_{10} \left( \frac{V_2}{V_1} \right)^2. \quad (9)$$

- Using a property of the log function, this gives,

$$\mathbf{dB} = 20 \log_{10} \left( \frac{p_2}{V_1} \right). \quad (10)$$

# Decibels in Pressure (or Voltage)

Double pressure:	→	+6 dB.
Halve pressure:	→	-6 dB.
Mpy pressure by 8:	→	+18 dB.
Divide pressure by 8:	→	-18 dB.

# Sound Pressure Level

- The sound pressure level (SPL) is proportional to the square of the power.

$$\text{dB}_p = 10 \log_{10} \left( \frac{p}{p_{\text{ref}}} \right)^2. \quad (11)$$

- Using a property of the log function, this gives,

$$\text{dB}_p = 20 \log_{10} \left( \frac{p}{p_{\text{ref}}} \right). \quad (12)$$

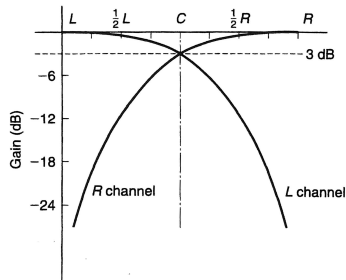
- The reference SPL for human hearing is  $20 \mu\text{Pa}$  (rms).



# Panning

# Panning

- Panning controls apply an un-even gain in the left and right channels.
- Panning is used to:
  - Reduce masking between two signals.
  - Add interest to a mix.
  - Give an impression of the source location within the stereo field.
- The attribution of gain depends on the panning law that is used. This is defined by the gain on L and R channels when the signal is centered.



# Panning

- You can only pan a channel within the actual width of the loudspeakers.
- Panning is used to trick the brain as to the direction of the source.
- There are additional considerations to take into account when using panning on headphones.

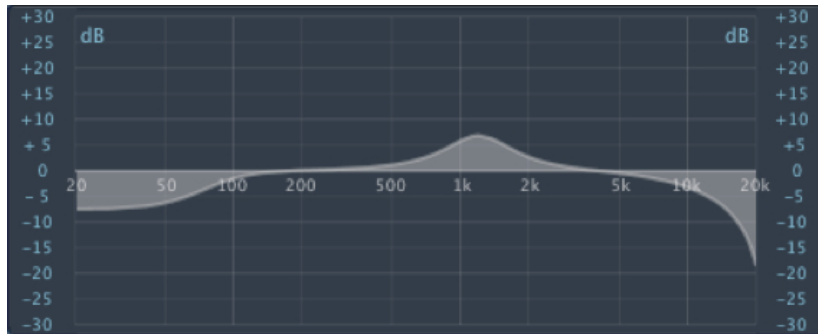
# Equalization

# Equalization

- A equalizer is used to change the gain in a certain part of the frequency spectrum.
- There are three main reasons for using an equalizer:
  - To improve the sound of the signal (very subjective!).
  - To separate the spectrum of two sources which are masking one another.
  - To remove unwanted artefacts from the signal.

# Equalization

- The affect of the equalizer must be examined as a function of gain and frequency.
- The frequency is plotted on a logarithmic scale.



# Equalizer controls

- A typical equalizer contains a number of separate filter bands. Each band will be of a certain type.
  - **Peak** - increase the gain around a certain frequency.
  - **Notch** - decrease the gain around a certain frequency.
  - **High shelf** - increase or decrease the gain at high frequencies.
  - **Low shelf** - increase or decrease the gain at low frequencies.
  - **High pass** - cut the gain below a certain frequency.
  - **Low pass** - cut the gain above a certain frequency.

# Equalizer controls

- Each filter type has three parameters.
  - **Frequency** - the frequency about which the filter is centred.
  - **Gain** - the amount of gain applied (not HP or LP).
    - Boost for positive gain.
    - Cut for negative gain.
  - **Q-factor** - the rate at which the change in gain is applied.
    - High Q-factor for sharp change.
    - Low Q-factor for smooth change.



# The affect of EQ on phase

- Whenever we make changes to the frequency spectrum of a signal we not only change the magnitude (what you are seeing on the EQ interface) but we also change the phase.
- Care must be taken to avoid unexpected phase summation or cancellation when using EQ.
- Phase changes are more severe for large, sharp magnitude changes (i.e. high gain and Q-factor).