

SOUND 1

ACOUSTICS AND PSYCHOACOUSTICS



SOUND 1 -- ACOUSTICS



The ear is the organ of hearing.





The outer ear or "pinna" collects sound waves and directs them into the "ear canal".





The ear canal conducts vibrations to the "inner ear".





The "eustachian tube" serves to equalize air pressure differences between the inner and outer ear.





The ear canal ends in the ear drum or "tympanic membrane". Sound waves entering the ear canal cause the ear drum to vibrate.





A series of small bones, the "ossicles", attached to the ear drum, serve as a mechanical transformer, to maximize the transfer of energy from the ear drum to the liquid-filled "cochlea".





The cochlea is a spiral tube, filled with liquid and lined with small, hair-like fibres. Vibrations induced in the liquid, stimulate the auditory nerve which sends information to the brain.





This is a view of the cochlea, uncoiled for clarity.





The lower or "basal" end is sensitive to high frequencies;





The lower or "basal" end is sensitive to high frequencies; while the upper or "apical" end is sensitive to low frequencies.





Reflections from the pinna (outer ear) make the ear directional in the vertical plane.



Sound originating in front of a listener arrives at the ear canal via two paths:







Sound originating in front of a listener arrives at the ear canal via two paths:

directly





Sound originating in front of a listener arrives at the ear canal via two paths:

- directly
- reflected from the folds of the pinna





The reflected path is slightly longer than the direct path.







At some frequency, the two paths will differ by a half wavelength.





At that frequency, the two waves will cancel each other out.





For instance, if the difference is 0.8 inches, the reflected sound would arrive 0.062 msecs. after the direct sound.





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This would result in a cancellation at 8.1 kHz.





If a sound originated from a higher elevation, the difference between the two paths would change,





If a sound originated a higher elevation, the difference between the two paths would change, causing the cancellation to happen at a different frequency.





In the real world, the earbrain system uses two methods to identify horizontal direction:



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relative amplitude



In the real world, the earbrain system uses two methods to identify horizontal direction:

- relative amplitude
- arrival time



Sound from a source directly in front of the listener will arrive at both ears:





Sound from a source directly in front of the listener will arrive at both ears:

equal in loudness





Sound from a source directly in front of the listener will arrive at both ears:

- equal in loudness
- coincident in time





Sound from a source slightly to the left of the listener will arrive at the left ear :





Sound from a source slightly to the left of the listener will arrive at the left ear :

slightly louder





Sound from a source slightly to the left of the listener will arrive at the left ear :

- slightly louder
- slightly earlier





Sound from a source extremely to the left of the listener will arrive at the left ear :





Sound from a source extremely to the left of the listener will arrive at the left ear :

much louder




Sound from a source extremely to the left of the listener will arrive at the left ear :

- much louder
- much earlier





In the theatre, one could use an individual speaker to reproduce the location of every possible sound source.



For some productions, or for a moving source, this could result in an unmanageable number of speakers.



It is more practical to use two loudspeakers to simulate left-right directionality in the horizontal plane.



In the Casgrain, for instance, we have speakers placed to the left and right of the proscenium.



If a sound is played at equal level on two speakers, the sound seems to emanate from the centre line of the speakers.





If a sound is played at equal level on two speakers, the sound seems to emanate from the centre line of the speakers.

If a sound is played slightly louder on the left speaker, it seems to emanate from slightly to the left of the centre line of the speakers.

Amplitude Difference





If a sound is played at equal level on two speakers, the sound seems to emanate from the centre line of the speakers.

If a sound is played slightly louder on the left speaker, it seems to emanate from slightly to the left of the centre line of the speakers.

If a sound is played much louder on the left speaker, it seems to emanate from the left speaker only.

Amplitude Difference





If a sound is played at equal level on two speakers, the sound seems to emanate from the centre line of the speakers.

If a sound is played slightly louder on the left speaker, it seems to emanate from slightly to the left of the centre line of the speakers.

If a sound is played much louder on the left speaker, seems to emanate from the left speaker only.

No matter how much louder the sound is played on one speaker, we cannot make the sound appear to emanate from beyond the location of that speaker.

Amplitude Difference















On the console, we use the "pan pot" to adjust the apparent left-right directionality of a sound.





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An acoustic or electronic delay can be used to mask the apparent direction of a sound source.



This is known as the precedence or Haas effect, after the Dutch acoustician who first studied it.



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A sound, arriving at the listener's ears within a window of 5 to 30 msec. after another, will be interpreted as a reflection of the earlier sound.





We often use a delay of 20 msec., as this value falls in the centre of the range over which the effect is operative.





The ear-brain system will take the direction of the earlier sound as the point of origin.





The ear-brain system will take the direction of the earlier sound as the point of origin, even if the delayed sound is as much as 10 dB louder.





If a sound is played at equal level on two speakers, the sound seems to emanate from the centre line of the speakers.





If the sound to the left speaker is delayed by 20 msec. , the sound will appear to come from the direction of the right speaker.





If the sound to the left speaker is delayed by 20 msec. , the sound will appear to come from the direction of the right speaker, even if the left speaker is 10dB louder.





If we feed a sound to an upstage speaker, while delaying the same sound to a proscenium speaker, the sound will appear to come from the direction of the upstage speaker.





We can use this effect to simulate an onstage radio or television. The audience will perceive the sound as coming from the upstage source.





The time delay to the proscenium speaker would have to be 20 msec. longer than the travel time from the upstage speaker.





At 35 ft., the travel time for the upstage speaker would be about 35 msec. At 15 ft., the travel time for the proscenium speaker

would be about 15 msec.





If we want the sound from the proscenium speaker to arrive 20 msec. after the sound of the upstage speaker, we would have to set the electronic delay to 40 msec.





This would make the sound appear to arrive from the upstage speaker, even though it is much louder coming from the proscenium.





To invoke the Haas effect, we could use an acoustic, rather than electronic delay.



The centre proscenium speaker in the Casgrain, is flown at a height that results in an acoustic travel time of 30 msec. at the centre of the house.





This is 5-10 msec. longer than the direct travel time throughout most of the house.





If we use the centre speaker to reinforce vocal mics., the audience will perceive the vocals as coming from the stage, rather than the speaker.




Haas Effect:

In a large theatre, to fill in the acoustic shadow under the balcony, additional speakers can be added.



Haas Effect:

The signal to these speakers can be delayed electronically – 20 msec. longer than the acoustic delay from the main speakers.



Haas Effect:

The sound will appear to come from the main speakers, even if the level from the under-balcony speakers is 10 dB louder.



Reflection:



Reflection:



Diffraction:



Diffraction:



Refraction:



Refraction:



In an enclosed space, the sound heard by the audience is comprised of:

In an enclosed space, the sound heard by the audience is comprised of:

direct sound



In an enclosed space, the sound heard by the audience is comprised of:

- direct sound
- early reflections



In an enclosed space, the sound heard by the audience is comprised of:

- direct sound
- early reflections
- reverberation



Early reflections, which arrive at the listener within 35 msec. of the direct sound, are perceived by the ear-brain as adding to the intensity of the direct sound.



Reflections arriving at the listener later than 50 msec. after the direct sound, tend to decrease intelligibility of the direct sound.



In general:



In general:

– early reflections = good

In general:

- early reflections = good
- late reflections = bad

The time it takes for the reverberant field to die away to 60 dB less than the original sound, is called the reverberation time or RT_{60} of a space.

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- intended use
- volume



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The RT₆₀ of the Casgrain is 1.75 sec.:

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- excellent for spoken word

- The RT₆₀ of the Casgrain is 1.75 sec.:
- excellent for spoken word
- somewhat short for music

Because of this, for a musical, we often mic. the singers' voices in order to add artificial reverberation.

Human voice:



- Human voice:
- Horizontal Plane



Human voice:

- Horizontal Plane
- approx. 90° @ 4 kHz



Human voice:

- Horizontal Plane
- splayed side walls



Human voice:

- Horizontal Plane
- splayed side walls
- redirect early reflections to audience



- Human voice:
- Vertical Plane



Human voice:

- Vertical Plane
- approx. 60° @ 4 kHz



Human voice:

- Vertical Plane
- approx. 60° @ 4 kHz
- 15° above horizontal



- Human voice:
- Vertical Plane



Human voice:

- Vertical Plane
- raked seating


Directivity:

Directivity Factors of Bounded Sources

Source/Boundary Configuration	(• • • •	•		
Solid angle seen by source	4π	2π	Π	π/2
Relative sound pressure	0dB	+6dB	+12dB	+18dB
Directivity Factor (Q)	1	2	4	8

Directivity:

Directivity Factors of various sound sources.

Source	Directivity Factor (Q)	
Point Source (Non-directional)	1	
Cone Loudspeaker (in baffle)	2	
Human Voice	2.5	
Loudspeaker (in box on floor against wall)	4	
Horn (Short Throw)	10	
Horn (Long Throw)	20	

Sound intensity decreases with the square of the distance from the source:

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- -6 dB for doubling

Sound intensity decreases with the square of the distance from the source:

- -6 dB for doubling
- -20dB for factor of 10x

Sound intensity decreases with the square of the distance from the source:

-6 dB for each doubling



Indoors or out, direct sound decreases with the square of the distance from the source -- 6 dB less for every doubling, or 20 dB for every factor of 10.





The reverberant sound field is constant. It doesn't vary with distance from the source.





At some distance from the source, both direct and reverberant sound fields are equal in level.





We call this the critical distance. Beyond this distance, the listener hears more reverberant sound than direct sound.





Critical distance (Dc) can be increased by increasing the directivity (Q) of the source, or by decreasing the reverberation time (RT_{60}) of the room.





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