

PHY 103: Loudness and Amplification

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Reading

- Heller, Ch. 7
- Fletcher and Rossing, Ch. 9 and 12 (advanced)

Midterm Exam

- In class, Thursday October 20
- Will be largely conceptual with some basic arithmetic and logarithms
- **Topics:**
 - Properties of waves
 - Normal modes of strings and air columns
 - Scales and tuning systems
 - Loudness and amplification

Need for Amplification

- Instruments produce sound by vibrating some element (a string, a drumhead, a reed, your lips, ...)
- There is typically very little energy in this vibration
 - For plucked string, total energy is $E = 1/4 \mu \omega_n^2 A_n^2 L$
 - First E string on a guitar: $\omega_1 = 2\pi \times 329.63$ Hz
 - µ = 0.4 g/m
 - L = 0.6 m, A = 3 mm
 - Energy is about 2 mJ. Energy required to lift an apple (100 g) vertically by 1 meter against gravity:

Amplification

- In order to actually hear an instrument and project its sound we have to amplify it
- Acoustic resonators have been used for thousands of years for this purpose
- Now we can couple instruments to electroacoustic transducers, amplifiers, and loudspeakers
- Before we get into acoustic amplification, let's talk about sound power (a physical quantity) and loudness (a perceived quantity)...

Relative Power: dB

- Energy in an oscillating wave: E ~ A² [Joules]
- Power: P = energy per unit time [Joule/sec = Watts]
- Intensity: I = power per unit area [W/m²]
- Relative sound level between two sounds is given in decibels [dB]
 - Sound I has power P_1 and sound 2 has power P_2
 - Relative difference: $IO \log(P_2 / P_1) dB$
 - $P_2 = 2 P_1 \rightarrow 10 \log 2 = 3 \text{ dB}$ difference

The Decibel Scale

- Sample: noise decreased in steps of 6 dB
- Sample: noise decreased in steps of 3 dB
- Sample: speaker moving away from microphone. The sound intensity decreases as $1/r^2$, where *r* is the distance to the microphone



Time-Averaged Power

- Note: when we talk about power in this context, we don't mean the instantaneous power at one time t
- We are talking about the time-averaged power of an oscillating pressure wave

$$p(t) = A\cos(\omega t)$$

$$P(t) \propto p(t)^{2}$$

$$\left\langle P_{\text{avg}} \right\rangle = \frac{1}{\tau} \int_{0}^{\tau} P(t) dt \propto \frac{1}{\tau} \int_{0}^{\tau} A^{2} \cos^{2}(\omega t) dt = A^{2} / 2$$

$$\propto P_{\text{max}} / 2$$

Sound Intensity Reference

- Decibels measure a relative power (or intensity) ratio between two sounds
- So when we talk about a sound being 60 dB (for example), what is the reference point we are using?
- Remember that sound corresponds to pressure variations propagating through air in longitudinal waves
- Reference point: $p_0 = 0.02 \ \mu$ Pa or about 2×10^{-10} atm
- So you can think of dB as the amplitude variation of sounds (in pressure) relative to this reference point
 - sound level = $10 \log(p / p_0)^2 = 20 \log(p / p_0) dB$

Power of Common Sounds

Sound Source	Sound Level	Intensity
Hearing threshold	0 dB	1 pW/m ²
Whispering in library, 2 m distance	30 dB	1 nW/m ²
Normal conversation, 1 m distance	60 dB	1 µW/m²
City traffic noise inside a car	85 dB	
Hearing loss threshold (sustained exposure): 90 dB		1 mW/m ²
Jackhammer at 20 m distance	95 dB	
Hand drill	98 dB	
Power mower	107 dB	
Rock concert	115 dB	
Pain threshold: 120 dB		1 W/m ²
Pneumatic riveter at 1 m	125 dB	
Permanent damage threshold (sh		
Jet engine at 30 m	140 dB	100 W/m ²
12 gauge shotgun blast	165 dB	
Death of hearing tissue	180 dB	1 kW/m ²
Loudest undistorted sound at 0C, 1 atm	194 dB	

From gcaudio.com

What Happens >194 dB?

- ► 194 dB corresponds to a pressure amplitude of 1 atm
- Sound levels >194 dB require sound pressures >1 atm
- Sound waves become distorted (like a shock wave), with valleys clipped at 0 Pa (vacuum)





• OSHA guidelines for sound level exposure:

Sound Level [dB]	Power/Intensity difference	Exposure Time	From gcaudio.com
90		8 hr/day	e
92	1.58	6 hr/day	
95	3.16	4 hr/day	-
97	5.01	3 hr/day	-
100	10	2 hr/day	-
102	15.85	90 min/day	-
105	31.62	60 min/day	-
110	100	30 min/day	-
115	316.23	<15 min/day	-

• Q: how did I go from dB to intensity difference?

- Suppose you have a 70 dB speaker and you add a second 70 dB speaker. How loud is the combined set?
- Consider two possible cases:
 - I. Coherence: signals have same frequency and are in phase
 - 2. Incoherence: signals have different frequencies

- If the speakers are in phase (coherent) where you are sitting, then the amplitudes of the waves add
- Power ~ Energy ~ Amplitude²

 $p(t) = A\cos(\omega t)$ $P(t) \propto [p(t) + p(t)]^{2}$ $= 4A^{2}\cos^{2}(\omega t)$ $P(t) = \frac{1}{2}\int_{0}^{\tau} P(t) dt = \infty 2$

$$\left\langle P_{\rm avg} \right\rangle = \frac{1}{\tau} \int_0^\tau P(t) dt = \propto 2A^2$$

Average power is 4x the power of one signal ($A^2/2$)

- If the two speakers aren't playing the same frequency, they are incoherent and the calculation changes
- Remember that we're dealing with time-averaged power when we combine the signals:

$$p_{1}(t) = A\cos(\omega t), \quad p_{2}(t) = A\cos(\Omega t)$$

$$P(t) \propto [p_{1}(t) + p_{2}(t)]^{2}$$

$$= A^{2}\cos^{2}(\omega t) + A^{2}\cos^{2}(\Omega t) + 2A\cos(\omega t)\cos(\Omega t)$$
time average: A²/2 time average: A²/2 time average: 0

$$\langle P_{\rm avg} \rangle \propto A^2$$
 2x the power of 1 signal alone

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Coherent case:

- Difference in sound level = $10 \log (4) = 6 dB$
- Therefore, the total sound power of the two speakers is 70 dB + 6 dB = 76 dB
- Incoherent case:
 - Different in sound level = $10 \log (2) = 3 dB$
 - Therefore, the total sound power of the two speakers is 70 dB + 3 dB = 73 dB

Loudness Perception

dB gives us an objective measurement of sound level power, but we don't perceive loudness equally at all frequencies



frequency [Hz]

dB and Phons

- Phon: psychophysical loudness measure
- At I kHz, phon scale = dB
- Everywhere else, phons are tuned to subjective response
- Equal-loudness curve: at a given frequency, what dB sound produces same loudness w.r.t. I kHz?
- We have a strong frequencydependent response to sounds



Amplification by Reflection

- If you hold a sound source next to a wall, the waves will be reflected off the wall back at you
- You can get amplification, as if there were a second source behind the wall



Reflection from a Corner

If wavelength is large compared to the distance between the source and its N-1 images, the power in the physical region goes up by N²



Reflection from a 30° Wedge





Amplification: the Guitar

A guitar is a fretted board of strings coupled to two flat vibrating panels



From The Physics of Musical Instruments, N. Fletcher and T. Rossing

Soundboard Bracing

- Designs for bracing a guitar soundboard: Torres, Bouchet (France), Ramirez (Spain), crossed bracing
- Bracing affects the vibrational modes of the board



From The Physics of Musical Instruments, N. Fletcher and T. Rossing

Guitar Schematic

Different pieces of the guitar resonate in different frequency regimes



From The Physics of Musical Instruments, N. Fletcher and T. Rossing

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Guitar Plate Modes

Vibrational modes of a guitar plate blank without braces



Effect of Braces

Vibrational modes of guitar top plate with traditional fan bracing



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Full Guitar Modes

The whole guitar cavity has vibrational modes that are combinations of the modes of the plates and sides put together



From The Physics of Musical Instruments, N. Fletcher and T. Rossing

The Piano

The piano also uses an acoustic resonator: the soundboard



- The soundboard opposes the vertical components of the string tension (lots of force: 20 N per string)
- Acoustically, it's also the main radiating component

Upright Piano Soundboard

An upright piano has a vertically oriented rectangular soundboard



From The Physics of Musical Instruments, N. Fletcher and T. Rossing

Frame members restrict the vibrations to a trapezoidal section where the bridges run diagonally

Upright Piano Modes

Vibrational modes of the upright piano soundboard



From The Physics of Musical Instruments, N. Fletcher and T. Rossing

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Rectangular Membrane Modes







Courtesy Dan Russell, Grad. Prog. in Acoustics, PSU

Horns

From Why You Hear What You Hear by Eric Heller



Horns

- Horns are built in a cone + bell shape, usually with the width of the cone growing exponentially towards the end of the bell
- The narrow end of the horn "loads the source"
 - Lots of constructive interference of certain wavelengths
 - Sound is confined to a small space, reducing $1/r^2$ losses
- The wide end of the horn lets all vibrations with wavelength < bell diameter escape without strong reflections. Technical: no impedance mismatch

Ripple Tank Simulation



Horns in Reverse



From Why You Hear What You Hear by Eric Heller



Summary

- Decibel scale: logarithmic ratio of sound level power or intensity with respect to a reference point
 - sound level = $10 \log (P_1 / P_2) dB$
- We perceive loudness differently depending on the frequency of the sound
- Amplification can be achieved by reflection from a surface, if the distance between the surface and sound source is much bigger than the wavelength of the sound
- In musical instruments, amplification is achieved by coupling vibrating elements to a resonator with low impedance sound effectively transmitted into the air