## Pitch vs. Frequency:

Pitch = human ear's perception of frequency of a sound vibration
Low pitch $\Leftrightarrow$ low frequency of vibration/oscillation
High pitch $\Leftrightarrow$ high frequency of vibration/oscillation
Q: Is the relation between \{perceived\} pitch vs. frequency linear?
$e . g$. a straight line $y=m x+b$ relation? A: No... See figure below:
Define frequency "units" associated with $\underline{\text { subjective }}$ pitch $=\underline{\text { mels }} \Leftrightarrow$ analogous to $\mathbf{H z}$.


## The Audible Frequency Range of Human Hearing (when young):

$$
20 \mathrm{~Hz}<f<20 \mathrm{KHz} \quad(\simeq 3 \text { orders of magnitude })
$$

As we grow older, the ange of frequencies that we can hear decreases (both high and low frequencies - mostly on the high frequency end...)

Frequency ranges of musical instruments typically $\sim 100 \mathrm{~Hz}$ to $\sim$ few KHz
e.g. guitar Low $\mathrm{E}=82 \mathrm{~Hz}$

High E $=330 \mathrm{~Hz}$
Piano highest note is ~ 4200 Hz
Very little above $\sim 10 \mathrm{KHz} \quad$ (squeals \& scrapes)
The human ear needs to be able to perceive a sound for minimum length of time $\Delta t$. In order to determine a pitch - i.e. pure/single-frequency tone - the minimum duration time $\Delta t$ of the pure tone depends on its frequency:


For $f \sim 100 \mathrm{~Hz}(\tau \sim 10 \mathrm{msec}): t_{\text {min }} \sim 40 \mathrm{msec} \quad$ ( $\sim 4$ cycles)
For $f \sim 1000 \mathrm{~Hz}(\tau \leq 1 \mathrm{msec}): t_{\text {min }} \sim 13 \mathrm{msec} \quad$ ( $\sim 13$ cycles)
The minimum duration time $\Delta t$ for human perception of a pitch is certainly in part due our ear \& brain processing, but for low frequencies especially, minimum time duration is also due to the uncertainty principle $\Delta f \Delta t=1$, which tells us that a pure tone/single-frequency sine wave signal of finite duration $\Delta t$ in fact has a finite frequency spread $\Delta f$ ! Only as the time duration $\Delta t \rightarrow \infty$ does $\Delta f \rightarrow 0$.

This can be seen by taking the Fourier transform of a finite-length \{time duration $\left.\Delta t_{0}\right\}$ pure-tone/single frequency $\left\{f=f_{0}\right\}$ time domain sinusoidal signal $p(t)=p_{0} \sin \omega_{0} t$ to the frequency domain $p(f)$ :

$$
\text { If } p(t)=p_{0} \sin \omega_{0} t=p_{0} \sin 2 \pi f_{0} t \text { for }|t| \leq \frac{1}{2} \Delta t_{0} \text {, and: } p(t)=0 \text { for }|t|>\frac{1}{2} \Delta t_{0} \text {, and }
$$ defining the \{rectangular\} window function $\mathrm{w}(t)=1(=0)$ for $|t| \leq \frac{1}{2} \Delta t_{\mathrm{o}} \quad\left(|t|>\frac{1}{2} \Delta t_{\mathrm{o}}\right)$, respectively, then:

$$
p(f)=\int_{t=-\infty}^{t=+\infty} p(t) \cdot \sin \omega t d t=p_{0} \int_{t=-\infty}^{t=+\infty} \mathrm{W}(t) \cdot \sin \omega_{0} t \cdot \sin \omega t d t=p_{0} \int_{t=-\frac{1}{2} \Delta t_{0}}^{t=+\frac{1}{2} \Delta t_{0}} \sin \omega_{0} t \cdot \sin \omega t d t
$$

Using the trigonometric identity $\sin A \sin B=\frac{1}{2}[\cos (A-B)-\cos (A+B)]$ :

$$
p(f)=\frac{1}{2} p_{0} \int_{t=-\frac{1}{2} \Delta t_{0}}^{t=+\frac{1}{2} \Delta t_{0}}\left[\cos \left(\omega-\omega_{\mathrm{o}}\right) t-\cos \left(\omega-\omega_{\mathrm{o}}\right) t\right] d t=\frac{1}{2} p_{\mathrm{o}} \Delta t_{\mathrm{o}}\left\{\frac{\sin \left[\frac{1}{2}\left(\omega-\omega_{\mathrm{o}}\right) \Delta t_{\mathrm{o}}\right]}{\left[\frac{1}{2}\left(\omega-\omega_{\mathrm{o}}\right) \Delta t_{\mathrm{o}}\right]}-\frac{\sin \left[\frac{1}{2}\left(\omega-\omega_{\mathrm{o}}\right) \Delta t_{\mathrm{o}}\right]}{\left[\frac{1}{2}\left(\omega-\omega_{0}\right) \Delta t_{\mathrm{o}}\right]}\right\}
$$

The sinc function $\operatorname{sinc}(x) \equiv \frac{\sin x}{x}\{n . b . \operatorname{sinc}(0)=1\}$, hence we can write $p(f)$ as:

$$
p(f)=\frac{1}{2} p_{0} \Delta t_{\mathrm{o}}\left\{\operatorname{sinc}\left[\frac{1}{2}\left(\omega-\omega_{0}\right) \Delta t_{\mathrm{o}}\right]-\operatorname{sinc}\left[\frac{1}{2}\left(\omega+\omega_{\mathrm{o}}\right) \Delta t_{\mathrm{o}}\right]\right\}
$$

The power spectral density functions $S_{p p}(f) \propto|p(f)|^{2}$ (a frequency domain quantity) for infinite length and finite length sine-wave signals are shown below:


- 3 -
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The sinc function $\operatorname{sinc}\left[\frac{1}{2}\left(\omega-\omega_{0}\right) \Delta t_{0}\right] \equiv \sin \left[\frac{1}{2}\left(\omega-\omega_{0}\right) \Delta t_{0}\right] /\left[\frac{1}{2}\left(\omega-\omega_{0}\right) \Delta t_{0}\right]$ for sinewave signals of $\{$ short $\}$ time duration $\Delta t_{o}=1 \tau_{0}, 2 \tau_{0}, 3 \tau_{0}, 4 \tau_{0}$ where $\tau_{0}=1 / f_{0}$ and the corresponding \# of cycles of oscillation $N_{c} \equiv \Delta t_{o} / \tau_{o}=1,2,3,4$ are shown in the figure below. Note that the width $\Delta f_{\mathrm{o}}$ of the main peak (at $f=f_{\mathrm{o}}$ ) depends $\underline{\text { inversely }}$ on the time duration $\Delta t_{0}$ of the signal, due to the uncertainty principle $\Delta f \Delta t=1$.


Human perception of pitch also depends $\{\sim$ weakly $\}$ on the loudness of the sound.

* Effect arises due to non-linearities in the $f$ \& $I$ response of the human ear.
* Pitch (perceived $f$ ) changes as loudness increases - see graph below...
* Effect exists only for pure/simple tones (!!!)
* Complex tones show no perceived pitch changes with loudness! (why??)


Two ears of same person may NOT perceive sound of a given frequency as having the same pitch!!! = DIPLACUSIS - happens only for diseased, and/or injured ears.

For normal musical purposes, frequency and pitch are synonymous (usually) n.b. applies only to periodic sounds.

Sound pulses are made up of a continuum of frequencies, sound pulses are thus anharmonic and hence have no characteristic frequency and/or pitch.

The human ear can discriminate changes in sound intensity levels/sound pressure levels/loudnesses of $\boldsymbol{J N D}=\Delta L=\left|L_{1}-L_{2}\right| \sim 1 / 2 \boldsymbol{d B}$; Our ability to do so also depends on frequency and sound pressure level/loudness:


Fic. 2. Just noticeable difference in sound pressure level for three frequencies.

A $\boldsymbol{J} \boldsymbol{N D} \sim 1 / 2 \boldsymbol{d B}$ change in sound intensity level corresponds to a fractional change in sound intensity of $\Delta I / I \sim 12 \%$. Thus, due to the $\sim$ logarithmic response of the human ear, it is not terribly sensitive to changes in the loudness of sounds.

The typical human ear can discern changes in pitch/frequency at the $\Delta f \sim 3 \mathrm{~Hz}$ level in the frequency range $\sim 30 \mathrm{~Hz} \leq f \leq 1000 \mathrm{~Hz}$. Again, has frequency dependence:


## Note that:

At very low frequencies: $\quad \Delta f / f \simeq 3 / 30=10 \%$ ( $\simeq 2$ semitones),
Whereas at higher frequencies: $\Delta f / f \simeq 3 / 1000=0.3 \%$ ( $\simeq 0.1$ semitones)
A good musician can discern frequency changes significantly smaller than this e.g. above $\boldsymbol{f} \geq 500 \mathrm{~Hz}: \approx 0.03$ semitone (i.e. $\Delta f / f \simeq 1 / 1000=0.1 \%$ )!!!
$\therefore$ The human ear/brain is capable of detecting small changes in frequency!!!

The human ear/brain is capable of perceiving a fundamental even when no fundamental is actually present!!! This is the so-called missing fundamental effect.

This effect is \{again\} a consequence of the non-linear response in/inside the human ear itself, and/or a non-linear response(s) in the human brain's processing of frequency information - whenever e.g. a quadratic non-linear response exists (in any system), if two signals $A$ and $B$ with frequencies $\boldsymbol{f}_{A}$ and $\boldsymbol{f}_{B}$ are input to that system, then sum and difference frequencies ( $f_{A}+f_{B}$ ) and $\left|f_{A}-f_{B}\right|$ are produced! Thus, e.g. a $2^{\text {nd }}$ harmonic $2 f_{1}$ and a $3^{\text {rd }}$ harmonic $3 f_{1}$ can produce a "missing" fundamental from the difference frequency, $\left|3 \boldsymbol{f}_{1}-2 \boldsymbol{f}_{1}\right|=\boldsymbol{f}_{1}$ !!! For further details on distortion, read Physics 406 Lecture Notes on "Theory of Distortion I \& II".

For some musical instruments - e.g. the trumpet, the oboe and/or the bassoon - the $2^{\text {nd }}$ (or even $3^{\text {rd }}$ and higher) harmonics can actually have a larger amplitude than that of the fundamental, however we perceive/hear the "note" that is played on the trumpet (and/or oboe, bassoon) as that of the fundamental!!!

The harmonic spectra - aka power spectral density functions $S_{p p}(f) v s . f$ and associated \{time-averaged\} relative phase harmonic phasor plots are shown below e.g. for the steadily-played notes $A 4(440.0 \mathrm{~Hz})$ played on the oboe, and F2 (87.3 Hz ) played on the bassoon:




Note that the vertical axes of $S_{p p}(f)$ vs. $f$ are displayed on a logarithmic scale.

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