

**Example 6.41.** Let  $x$  denote a continuous-time audio signal with Fourier transform  $X$ . Suppose that  $|X(\omega)| = 0$  for all  $|\omega| \geq 44100\pi$ . Determine the largest period  $T$  with which  $x$  can be sampled that will allow  $x$  to be exactly recovered from its samples.  $44100\pi \text{ rad/s} = 22.05 \text{ kHz}$

*Solution.* The function  $x$  is bandlimited to frequencies in the range  $(-\omega_m, \omega_m)$ , where  $\omega_m = 44100\pi$ . From the sampling theorem, we know that the minimum sampling rate required is given by

$$\begin{aligned}\omega_s &= 2\omega_m && \text{from sampling theorem} \\ &= 2(44100\pi) && \omega_m = 44100\pi \\ &= 88200\pi. && 88200\pi \text{ rad/s} = 44.1 \text{ kHz}\end{aligned}$$

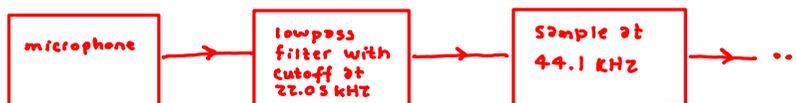
Thus, the largest permissible sampling period is given by

$$\begin{aligned}T &= \frac{2\pi}{\omega_s} && \text{take reciprocal for corresponding} \\ &= \frac{2\pi}{88200\pi} && \text{sampling period} \\ &= \frac{1}{44100}.\end{aligned}$$

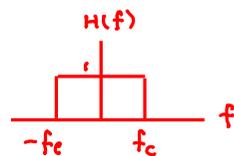
Why does CD-quality audio use a sampling rate of 44.1 kHz?

In practice, how do we ensure the audio signal to be sampled is sufficiently bandlimited?

The human auditory system (assuming pristine hearing) can sense frequencies up to about 22.05 kHz.



- filter prevents aliasing
- removed frequencies cannot be detected by humans



$$f_c = 22.05 \text{ kHz}$$