- All **data** stored in a computer is stored as binary numbers
- This means that sound needs to be stored as numbers
- There needs to be a way to break the sound down into small parts and then a way to **represent** each part as a number

Sound travels through the air as a wave. These change the air pressure that affects your ear drums.



Sound waves are **analogue signals**. This means that they are continuous, without any breaks.

To store them **digitally** (in a computer) we need to break them up and represent them as

numbers.



- The process of breaking up sound waves is called **sampling**
- Sampling lets us turn analogue information into digital data
- the pixels in a digital image are samples individual points that make up the image. We need to do the same with sound waves.

- To change **analogue** sound into **digital** form we need to **sample** it
- This means we take readings at **regular** intervals along the sound wave to measure the **amplitude** (volume) and convert these into **numbers**
- The numbers can then be stored using binary

A **sample** is a measure of **amplitude** at a point in time.





- The music you play on a piano or trumpet are **analogue**
- The sound people hear when you sing is **analogue** (it may also be terrible)
- The sound on a vinyl record is **analogue**
- The sound on a CD is **digital**
- The sound from an MP3 file is **digital**