

# 1B Paper 6: Communications

## Handout 1: Introduction, Signals, and Channels

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## Course Information

- Seven lectures, recordings will be accessible after each lecture via the Panopto block on Moodle
- Lecture handouts (both filled and unfilled) will be posted on Moodle: <https://www.vle.cam.ac.uk>
- Feedback via email (rv285@cam.ac.uk) or using anonymous feedback facility

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# Topics

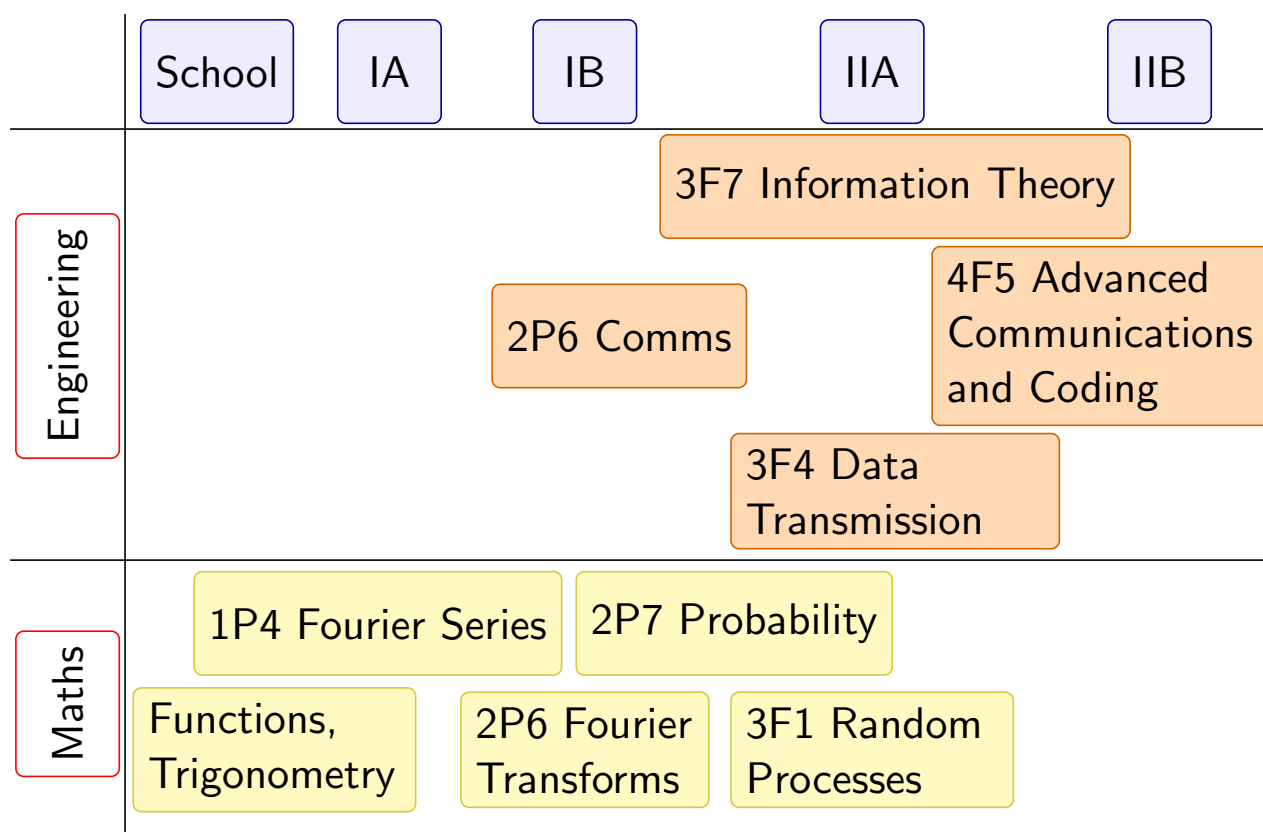
- Signals and Channels
- Analogue Modulation (AM, FM)
- Digitisation of Analogue Signals (sampling recap and quantisation)
- Digital Signals and Modulation
- A brief introduction to Channel Coding
- Multiple Access

## References:

-  S. Haykin and M. Moher, *Introduction to Analog & Digital Communications 2nd Ed.*, John Wiley & Sons, 2007
-  R. G. Gallager, *Principles of Digital Communications*, Cambridge University Press, 2008

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## “Communications” teaching



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# A Brief History

## Analogue Communications

- Telephone: patented in 1876
- Radio: AM since early 1900s, FM patented in 1930s
- BBC broadcast analogue TV from 1936-2012

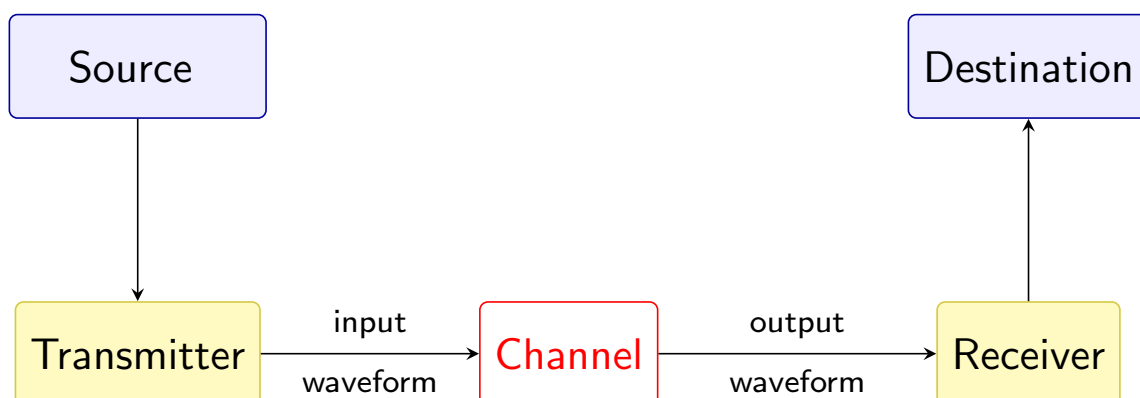
## Digital Communications

- Telegraph: first optical/semaphore 1767, electrical 1816
- Mobile Communications: GSM (1991) → 3G → 4G LTE
- Wi-Fi, first deployed in 1997, Bluetooth in '98
- Asymmetric Digital Subscriber Line (ADSL), up to 4Mbit/s, appeared early 2000
- Digital Video Broadcasting (DVB), first broadcast ever in the UK, in 1998. Since 2012, all broadcast TV in the UK is digital

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## The Basic Idea

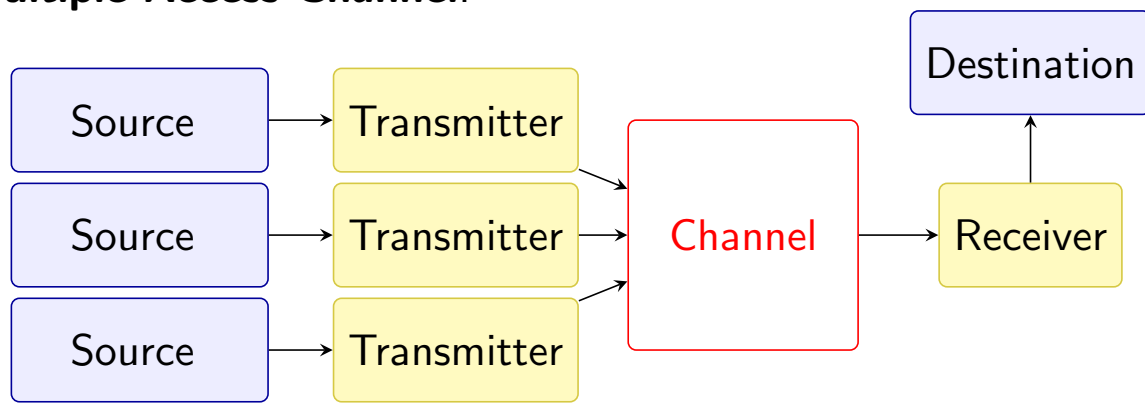
**Communication:** The process of delivering **information** from an information **source** to a **destination** through a communication **channel**.



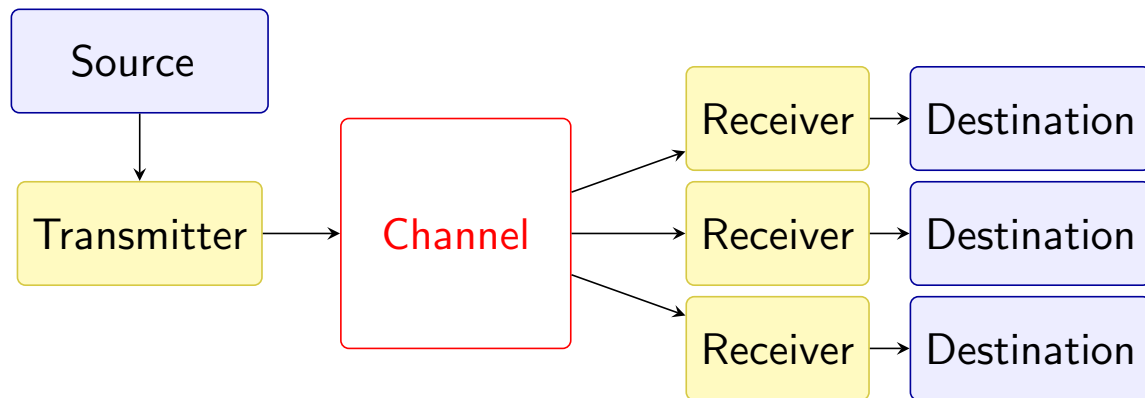
More generally, we could have multiple sources delivering information to multiple destinations through a common channel

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## Multiple Access Channel:

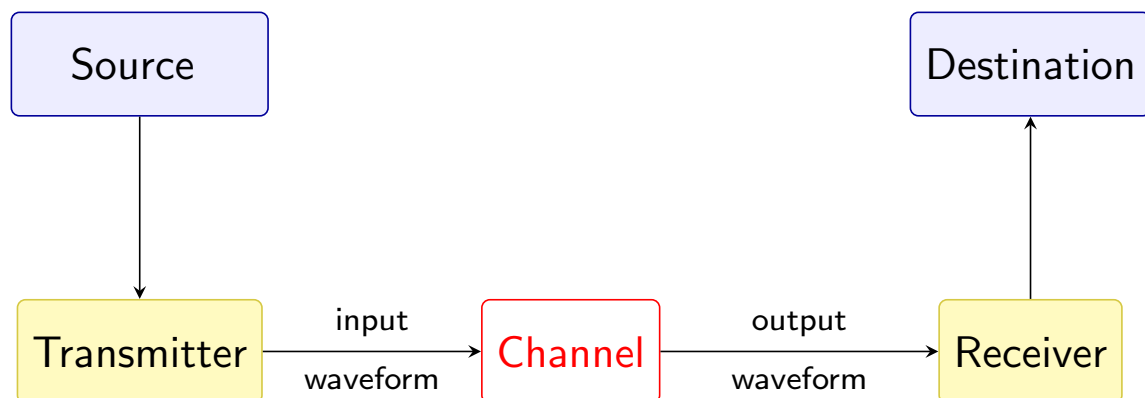


## Broadcast Channel:



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For most of this course, we will focus on the point-to-point communication model:



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# Block Diagram Components

- Source of information: May be analogue (voice, music, video), or digital (e.g., e-mail, any file on your computer)
- **Transmitter**: translates the information into a signal suitable for transmission over the channel
- **Channel**: medium used to transmit the signal to the receiver
  - E.g., optical fibre, wireless channel, magnetic recording...
  - May distort transmitted signal, e.g., add noise or attenuate it
- **Receiver**: reconstructs the source of information from the received signal
- Destination: for whom the information is intended

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## Key Signal Properties

Two properties of signals that are important for communication:

1. **Power**
2. **Bandwidth**

Let us define these terms and understand why they are relevant.

## Signal Energy

$$X(f) = \int_{-\infty}^{\infty} x(t) e^{-j2\pi ft} dt$$
$$x(t) = \int_{-\infty}^{\infty} X(f) e^{j2\pi ft} df$$

The **energy** of a signal  $x(t)$  is defined as

$$E_x = \int_{-\infty}^{\infty} |x(t)|^2 dt$$

If  $X(\omega)$  is the Fourier transform of  $x(t)$ , recall Parseval's theorem:

$$E_x = \int_{-\infty}^{\infty} |x(t)|^2 dt = \frac{1}{2\pi} \int_{-\infty}^{\infty} |X(\omega)|^2 d\omega = \int_{-\infty}^{\infty} |X(f)|^2 df$$

- $\omega = 2\pi f$  is the frequency in radians,  $f$  is frequency in Hz
- $|X(f)|^2$  is the **energy spectral density**

Can think of  $|X(f)|^2 df$  as the energy of the signal in the frequency band  $[f, f + df]$

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## Signal Power

For a signal  $x(t)$  whose energy is infinite, the **power** is defined as

$$P_x = \lim_{T \rightarrow \infty} \frac{1}{T} \int_{-T/2}^{T/2} |x(t)|^2 dt$$

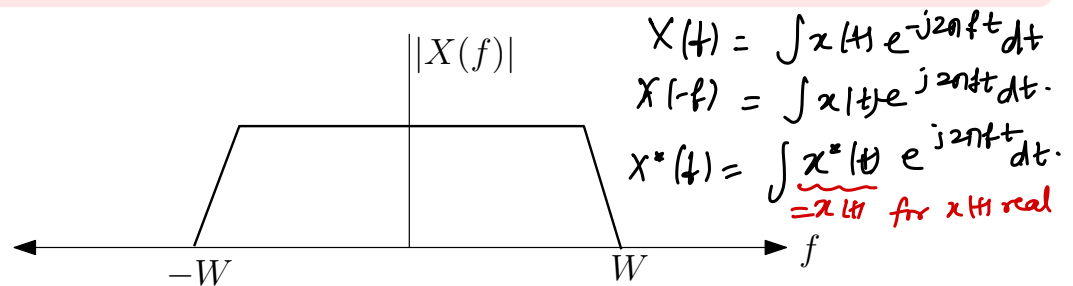
Why is signal power important?

- We are usually concerned about energy of the transmitted signal per unit time, i.e., *transmit power*
- Lower transmit power implies longer battery life for your phone
- But lower transmit power also makes signal harder to detect at the receiver in the presence of noise!
- Need clever Tx + Rx designs that make judicious use of available transmit power

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# Bandwidth

The bandwidth of a signal is roughly the range of frequencies over which its spectrum (Fourier transform) is non-zero.

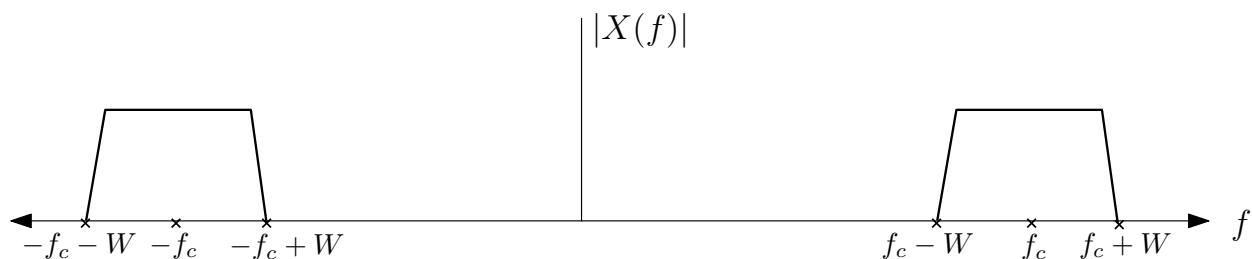


- For real signals, bandwidth measured as the range of **positive** frequencies as  $|X(f)|$  is symmetric around 0  
(as  $X(-f) = X^*(f)$  for real  $x(t)$ )  $\Rightarrow |X(-f)| = |X^*(f)| = |X(f)|$
  - In communications, signal bandwidth typically specified in Hz
- A signal is called *low-pass* or *baseband* if its spectral content is centred around  $f = 0$ .
- The bandwidth of the baseband signal above is  $W$
  - E.g., audio signals are baseband with bandwidth  $\approx 20$  kHz  
Voice signals in telephone systems have bandwidth  $\approx 4$  kHz

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## Passband signals

A signal is said to be *passband* if its spectral content is centred around  $\pm f_c$ , where  $f_c \gg 0$

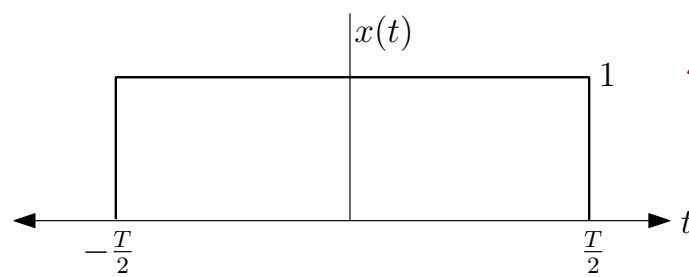


The bandwidth of this passband signal is  $2W$

Examples of passband signals:

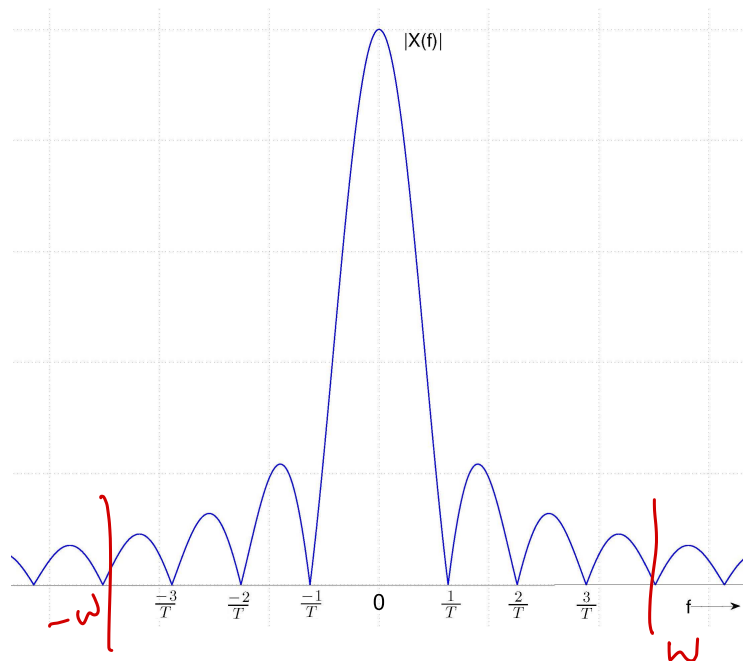
- AM (Amplitude-modulated) radio signals have bandwidth  $\approx 10$  kHz around  $f_c \approx 1$  MHz
- Transmitted signals in a WiFi network have bandwidth  $\approx 20$  MHz around  $f_c \approx 2.4$  GHz

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$$X(f) = T \operatorname{sinc}(\pi f T)$$

$\operatorname{rect}(t/T)$  is the rectangular pulse, which is 1 for  $-\frac{T}{2} \leq t \leq \frac{T}{2}$ , and 0 elsewhere. What is its bandwidth?



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## Bandwidth – A sensible definition?

Many real-world signals are time-limited  
 $\Rightarrow$  These *will not* be **strictly limited** in frequency

The absolute bandwidth of  $\operatorname{rect}(t/T)$  is  $\infty$ .

Other, more practical, definitions of bandwidth:

1. 90% bandwidth: The range of frequencies which contain 90% of the energy of the spectrum
2. 3-dB bandwidth: The range of frequencies which contain 50% of the energy of the spectrum
3. *Null-to-null* bandwidth: The width of the “main lobe” of the spectrum for the rect signal

- The “main-lobe” bandwidth of  $\operatorname{rect}(t/T)$  is  $\frac{1}{T}$
- If we also include one side-lobe, bandwidth of  $\operatorname{rect}(t/T)$  is  $\frac{2}{T}$

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Thus, bandwidth is a measure of the *extent of significant spectral content* of the signal

Bandwidth is a scarce resource, especially in mobile (cellular) communication:

- Wireless bandwidth licensed and regulated by OFCOM
- A company has to buy a slice of spectrum, say few tens of MHz around  $f_c \approx 2$  GHz, and restrict its transmitted signals to *within* the spectrum
- Passband 4G spectrum of few tens of MHz auctioned for hundreds of millions of £ to telecom companies!

Wired channels such as telephone lines and USB cables act like linear systems or *filters*:

- Their transfer function is roughly flat over a band of frequencies  $[-W, W]$  around 0, and then attenuates to 0 for higher frequencies.
- Therefore, transmitted signals need to be bandlimited to  $W$

In both wired and wireless communication, need good Tx + Rx designs that make optimal use of available bandwidth

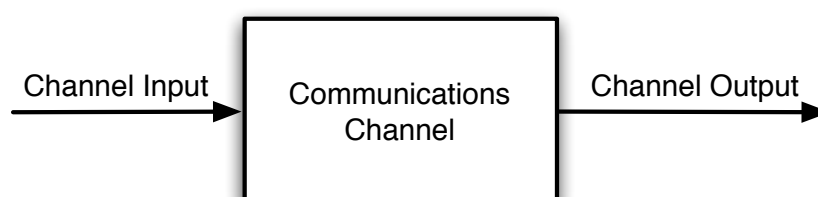
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## Communication Channels

### What is a channel?

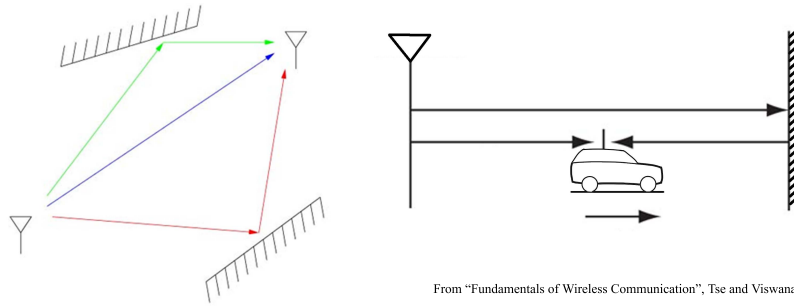
The medium used to transmit the signal from transmitter to receiver.

- Introduces **attenuation** and **noise**
- So the received signal is a faded and noisy version of what the transmitter sent
- Noise and attenuation can cause **errors** at the receiver



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# Some Real-world Channels



## 1. Mobile Wireless Channel:

- There is distortion of the signal caused by multipath propagation and mobility
- Exact type of distortion depends on the signal bandwidth

## 2. Optical Fibre Channel:

- Very large BW, cheap production, low attenuation
- Cons: dispersion of optical pulses, expensive regenerators reqd.
- Used in the core of the internet, for long-distance communication networks

## 3. Electrical Wire Channel:

- Twisted pair cables (e.g., Ethernet) have limited bandwidth, high attenuation; Cheap, used for short distances

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# Modelling a channel

KEY Q: How to model a channel ?

Channels are often modelled as *linear systems* with additive noise:

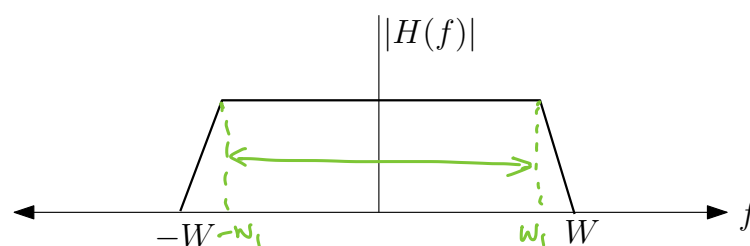
Channel output  $y(t)$  generated from input  $x(t)$  as

$$y(t) = h(t) \star x(t) + n(t)$$

In frequency domain:

$$Y(f) = H(f)X(f) + N(f)$$

For example, the frequency response of a telephone cable may look like:



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## Additive Noise Channel

If the input is restricted to the band where the channel  $H(f)$  is flat, then the channel is

$$Y(f) = X(f) + N(f)$$

or

$$y(t) = x(t) + n(t)$$

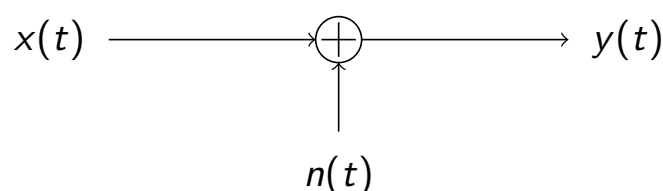
This is a very popular and useful model. What about  $n(t)$ ?

$n(t)$  is thermal noise at the Rx:

- Thermal noise is the noise generated by the thermal agitation of electrons inside an electrical conductor
- Happens regardless of the applied voltage
- All receivers (WiFi, mobile phone, AM, FM,...) generate thermal noise

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## Additive Gaussian Noise



Thermal noise  $n(t)$  is modelled as a *Gaussian* random process:

- At each time  $t$ ,  $n(t)$  is a Gaussian random variable
- A rigorous description requires knowledge of random processes (in 3F1)
- The additive Gaussian noise channel is the workhorse of communication theory: good model for many real-world communication systems
- Channels whose frequency response  $H(f)$  is not flat are important in practice, but outside the scope of this course

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In the remainder of the course:

- we will learn how to design both analogue & digital communication schemes ( $T_x + R_x$ )
- keeping in mind power and bandwidth constraints
- we'll then study how noise affects the performance of these schemes