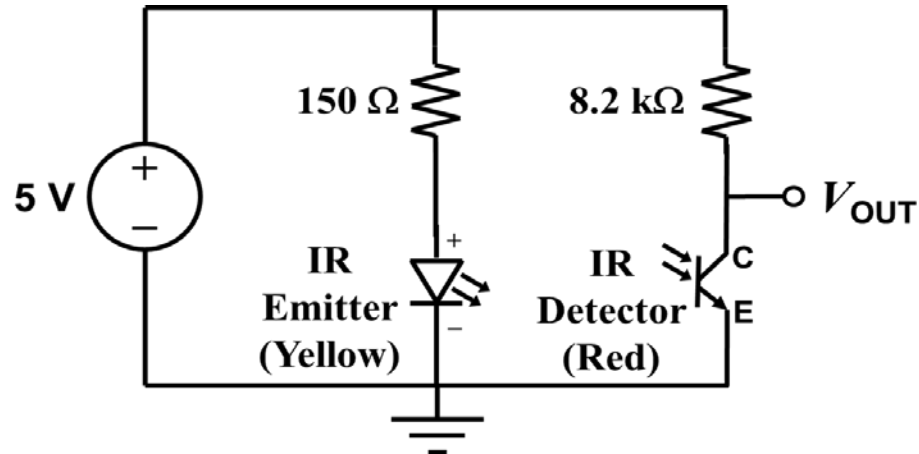


CG1111: Engineering Principles and Practice I

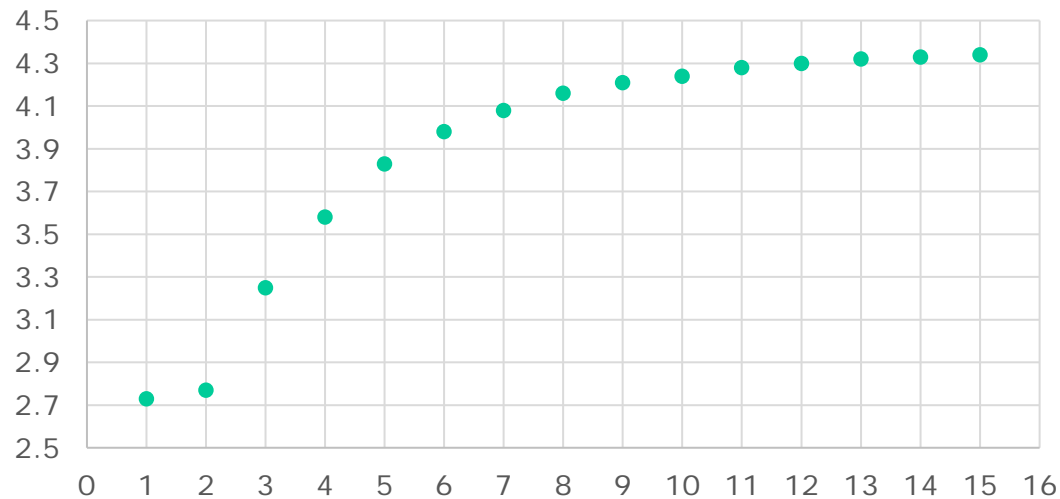
Debrief & Tutorial for Week 10
(Filters, Sensors, and Signal Processing Basics)



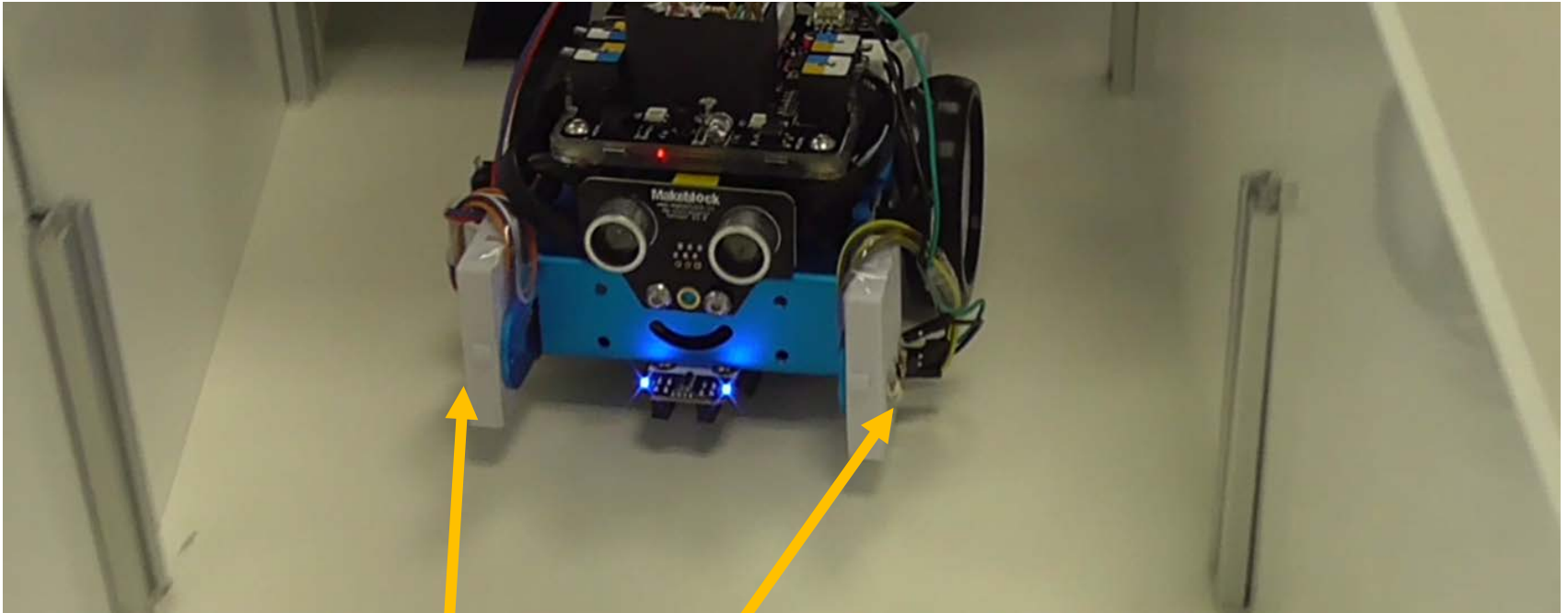
IR Proximity Sensor



V_{out} (V) vs Distance (cm)

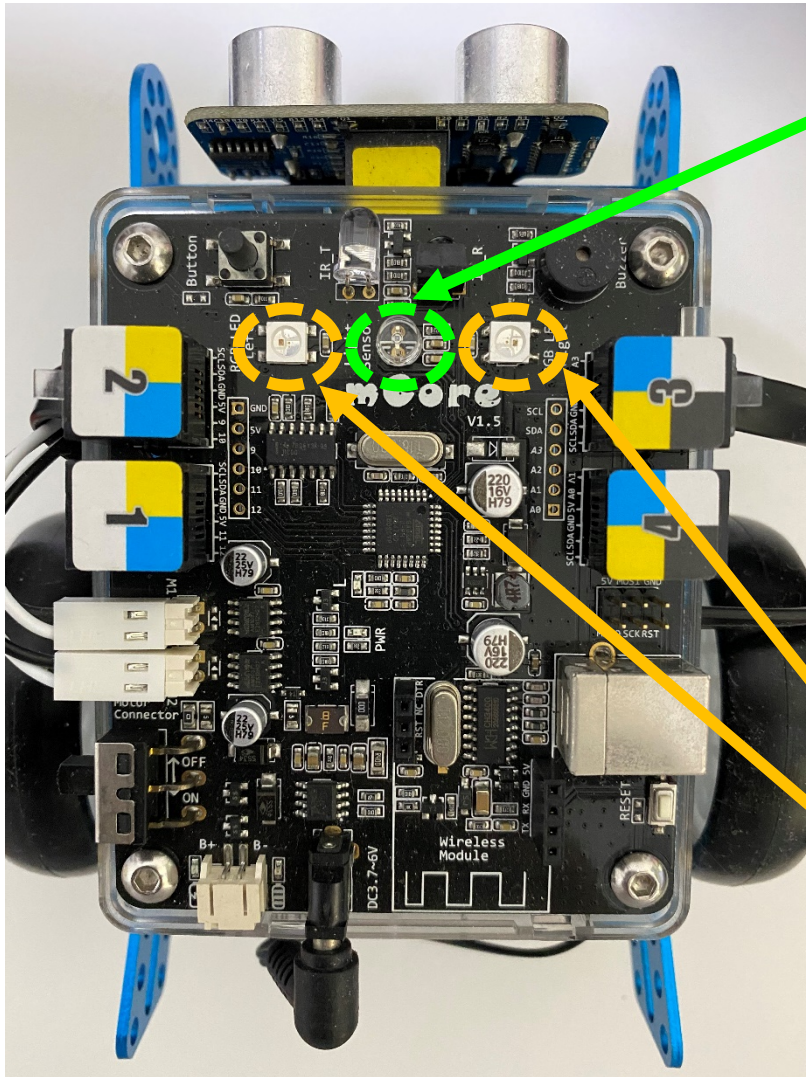


IR Proximity Sensor



IR proximity
sensors

Colour Sensing (RGB LED + LDR)



LDR

Read LDR's voltage corresponding to shining Red, Green, & Blue LEDs individually to estimate object's colour

Sensor's **response time** & **calibration** are important

Two RGB LEDs

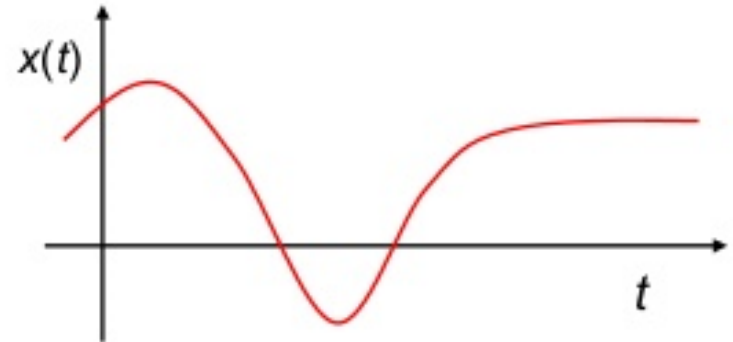
Classification of Signals

- Continuous signals

- Independent variable is a continuous variable

- Examples

- ✓ Sine wave from a function generator
- ✓ Speech signal received from a microphone

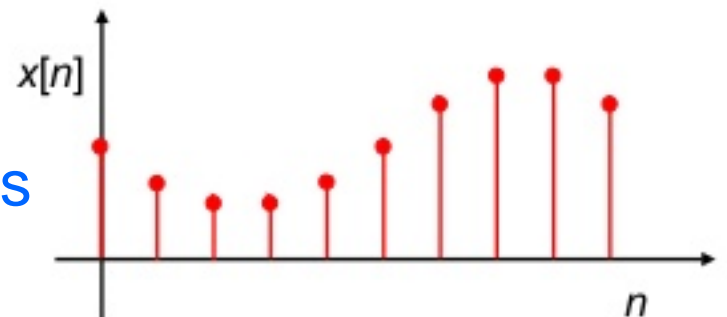


- Discrete signals

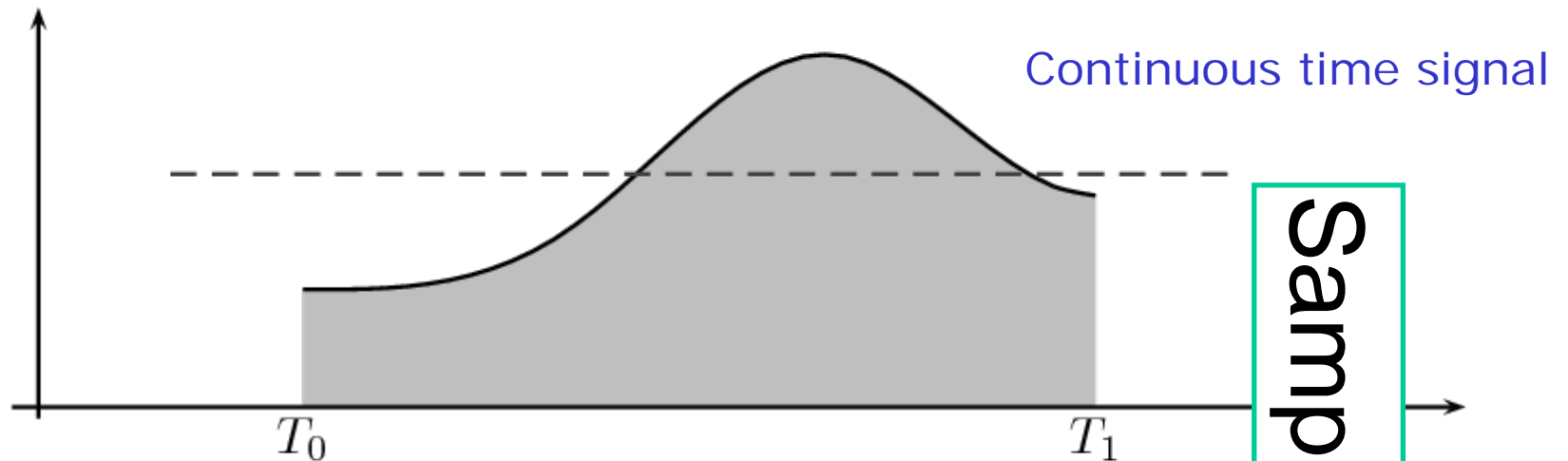
- Independent variable takes on discrete values, e.g., integers

- Examples

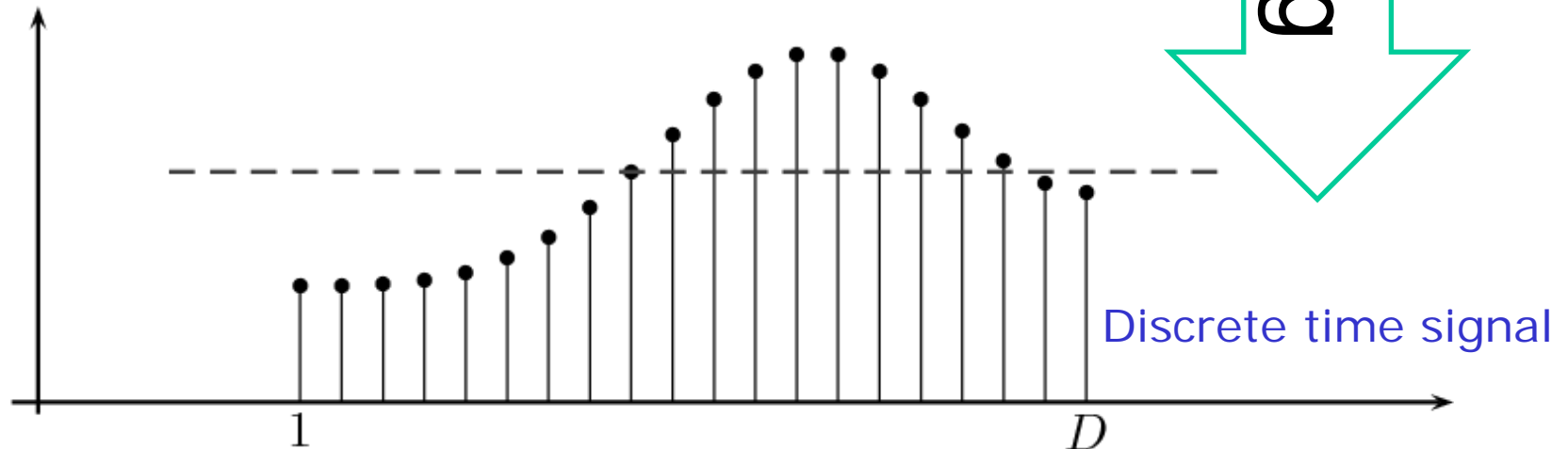
- ✓ Weekly stock market index
- ✓ Speech signal stored on a digital computer



Continuous vs. Discrete Time Signals



The **sampling rate or frequency** is the number of times a signal is read per second



Sampling

Sampling Theorem

- How frequently should the signal be sampled to ensure that information contained in the signal is preserved?

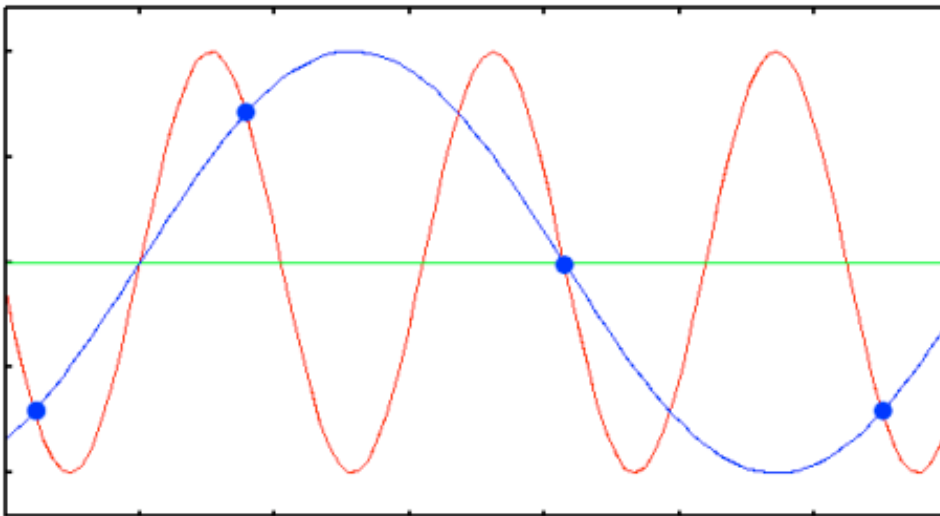
$$f_s \geq 2 \times \text{max frequency component in the signal}$$

Nyquist Rate & Nyquist Frequency

- The minimum sampling rate that is required to well represent a continuous time signal with highest frequency component f is given by $2 \times f$ and this is known as the Nyquist rate
- For a given sampling rate f_s , perfect reconstruction is possible for a continuous signal whose highest frequency component is $f_s/2$, also known as the Nyquist frequency
 - For example, audio CDs use a sampling rate of 44.1 kHz. Therefore, the Nyquist frequency is 22.05 kHz

Aliasing

- When the **sampling rate** is **lower** than the **Nyquist rate**, the signal reconstructed from samples (using DAC) is different from the original continuous signal
- This effect is known as **aliasing**



The red waveform represents the original continuous time signal

The blue dots represent samples obtained from the continuous signal with **sampling rate < Nyquist rate**

Question 1

- Design a bandpass filter for the 4th Octave of musical notes, with a passband from 261 Hz to 493 Hz
- High pass cut-off frequency = 261 Hz
- Low pass cut-off frequency = 493 Hz

Solution to Q1

- We need to design the values of R and C for the low-pass and high pass filter

- High Pass cut-off frequency $\rightarrow 261 \text{ Hz} = \frac{1}{2\pi R_H C_H}$

$\rightarrow R_H C_H = 609.79 \text{ } \mu\text{s}$

Choosing $C_H = 20 \text{ nF}$, we get $R_H = 30.5 \text{ k}\Omega$

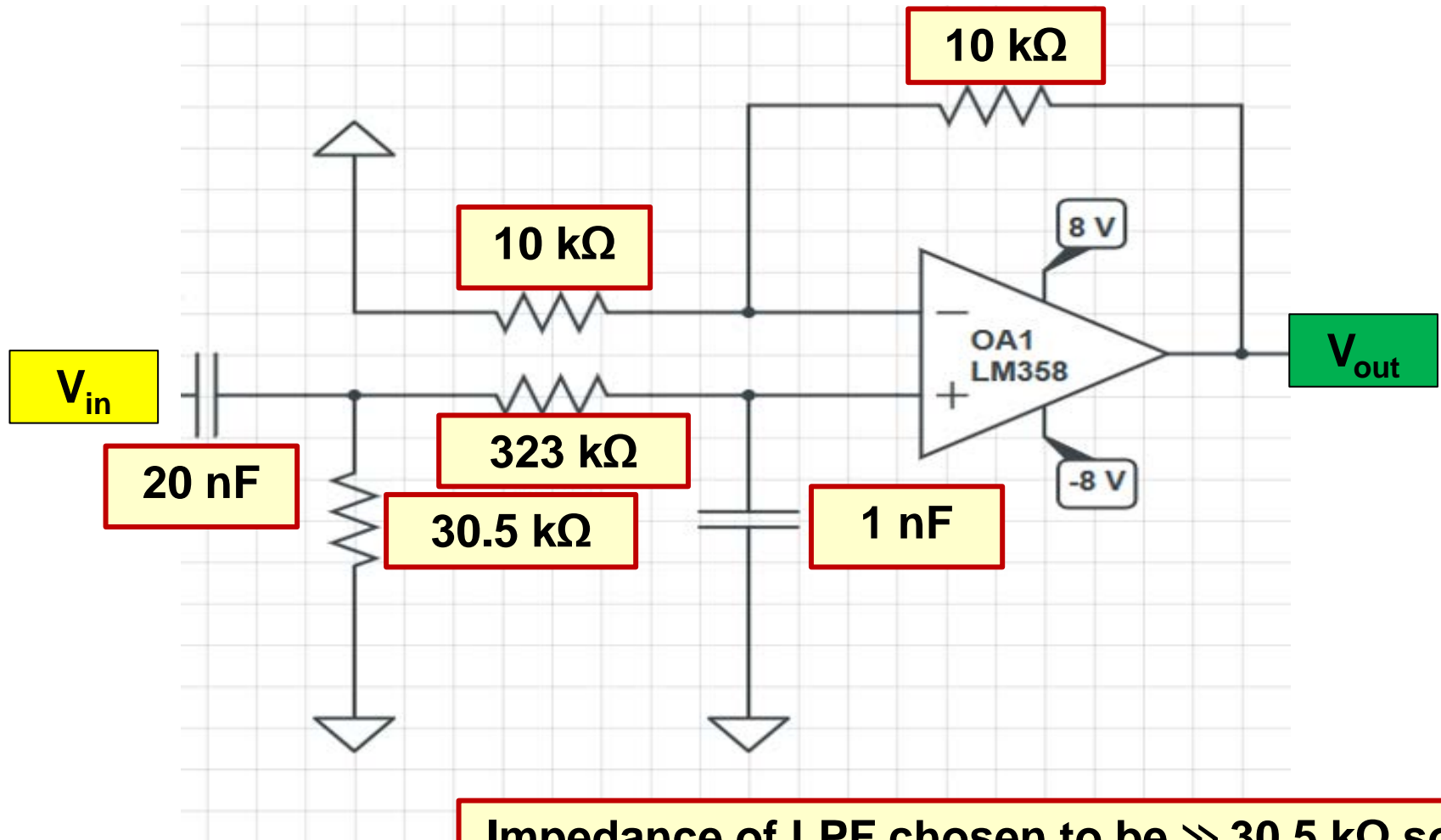
- Low Pass cut-off frequency $\rightarrow 493 \text{ Hz} = \frac{1}{2\pi R_L C_L}$

$\rightarrow R_L C_L = 322.83 \text{ } \mu\text{s}$

Choosing $C_L = 1 \text{ nF}$, we get $R_L = 323 \text{ k}\Omega$

- If we choose an active amplifier gain of 2 for the input, we can choose $R_i = 10 \text{ k}\Omega$ and $R_f = 10 \text{ k}\Omega$

Solution to Q1



Impedance of LPF chosen to be $\gg 30.5 \text{ k}\Omega$ so that loading effect on 30.5 kΩ is insignificant

Question 2

- Design a temperature sensing circuit using LM35 (temperature sensor IC), Op amps, resistors, and LEDs
- LM35 output = 250 mV at 25 °C
- LM35 output varies as 10 mV/°C
- Temperature below 35 °C, GREEN LED on
- Temperature above 35 °C, RED LED on

Solution to Q2

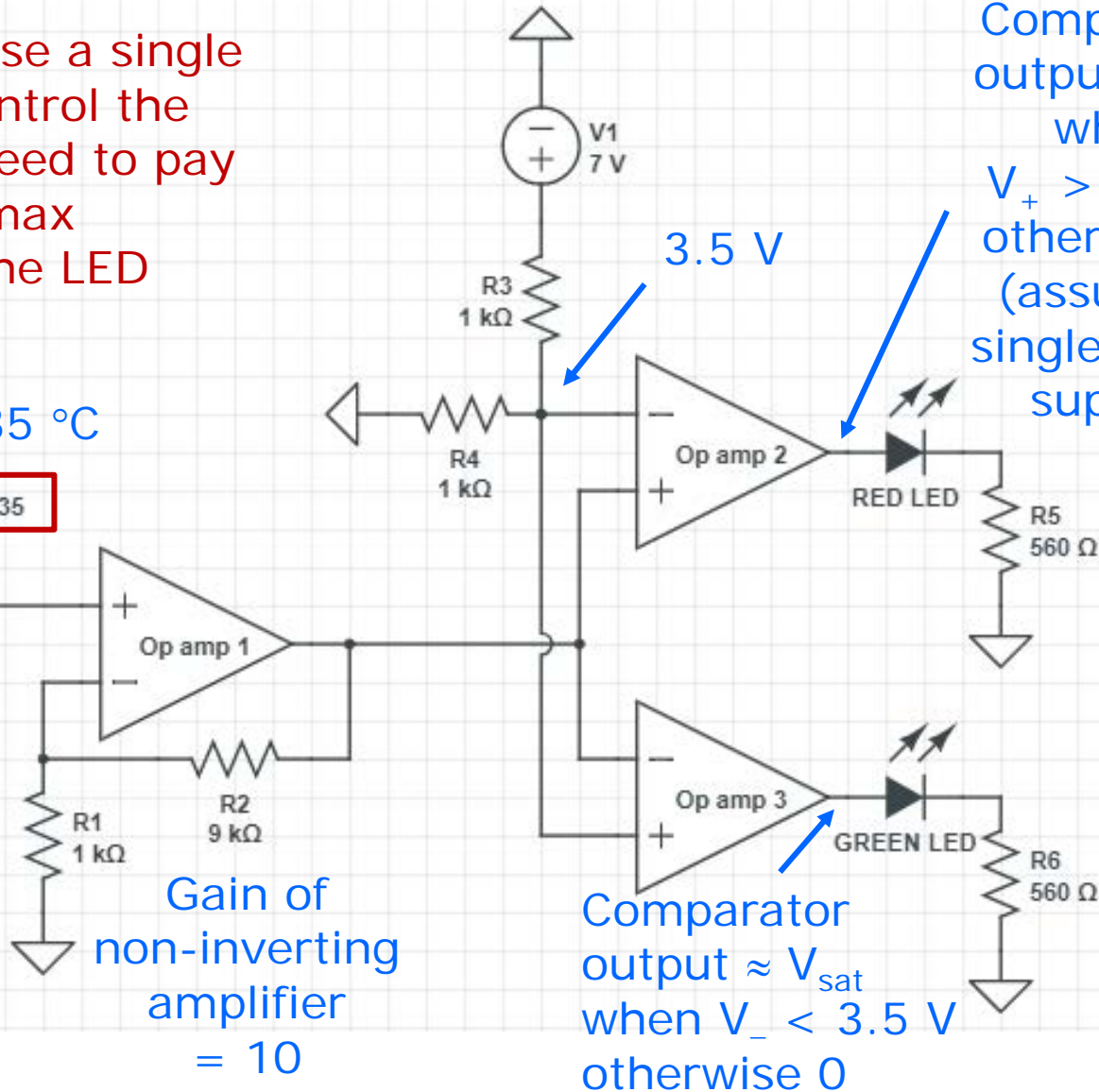
Note:

It is possible to use a single comparator to control the two LEDs. But need to pay attention to the max reverse bias on the LED

350 mV at 35 °C

Output from LM35

V_{temp}



Solution to Q2

- Op Amp 1 – closed loop gain of 10
- Op Amps 2 & 3 are comparators with bias voltage of 3.5 V
 - Op Amp 2 – non-inverting comparator
 - Op Amp 3 – inverting comparator
- When output of op-amp 1 is more than 3.5 V (i.e., when temperature > 35 °C), Red LED is turned on
- When output of op-amp 1 is less than 3.5 V (i.e., when temperature < 35 °C), Green LED is turned on

Question 3 & Solution

- System's sampling rate: 100 Hz
- Given signals:

Signal	Signal Frequency	Nyquist rate	Perfect reconstruction possible?
$5 \cos(500\pi t)$	250 Hz	500 Hz	No
$10 \sin(200\pi t)$	100 Hz	200 Hz	No
$5 \sin(100\pi t)$	50 Hz	100 Hz	Yes
$2.5 \cos(50\pi t)$	25 Hz	50 Hz	Yes

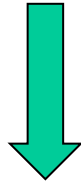
Question 4

- Find the condition for the sampling period (T_s) to correctly sample the signal $X(t)$, given by

$$X(t) = 5 \sin(10\pi t) + 2.5 \sin(4\pi t) + 3 \sin(0.1\pi t)$$

Solution to Q4

- $X(t) = 5 \sin(10\pi t) + 2.5 \sin(4\pi t) + 3 \sin(0.1\pi t)$



Highest frequency component: 5 Hz

According to sampling theorem, minimum sampling rate is 10 Hz

$$f_s \geq 10$$

$$T_s = \frac{1}{f_s} \leq \frac{1}{10} = 0.1 \text{ seconds}$$

$$T_s \leq 0.1 \text{ seconds}$$