

EE 313 ANALOG ELECTRONICS LABORATORY 2022-2023 FALL TERM PROJECT REPORT

AN OPTICAL WIRELESS COMMUNICATION SYSTEM: PHOTOPHONE

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Abstract—This document is about the end-term project of EE313 Analog Electronics Laboratory, namely design of an optical wireless communication system: photophone. Background theoretical knowledge, literature research and various work about design methods and mathematical analysis of them related to this project together with simulation and experimental results are defined in this document.

Index Terms—Optical wireless communication, freespace optical communication, Li-Fi, photophone, AGC, laser

I. INTRODUCTION

Communication is an integral part of our lives for us humans, who are social beings. While it was carried out by methods such as pigeons or fire in history, various types of communication have emerged with the advancement of technology. In this project, we will examine the one of the modern communication systems: optical wireless communication system. The overall diagram of the project is given in Figure 1, below for better understanding.

The aim of the project is to transmit the audio input signal that is generated by the microphone and to receive this information wirelessly. Then, the received signal is fed to the speaker at the final step while the quality of the signal is indicated by a single RGB LED. In general, the project can be grouped under two main units: Transmitter Unit and Receiver Unit, as it can be seen in Figure 1. Also, each main part consists of different sub-units, and they are explained in detail in the following sections of the report.

II. RULES

Maximum allowed DC Voltage: ± 15 Volts.

Instruments not allowed using: 6V terminal of the DC supply.

Frequency Range for Reference Signal: 10 kHz – 30 kHz.

Component not allowed to be used: audio op-amps, microphone with integrated driving circuitries, infrared and ultraviolet lasers, and visible light lasers whose power > 5 mW.

III. TRANSMITTER UNIT

A. Microphone Driver

The first part of the transmitter unit is microphone driver circuit. To transmit an audio signal using a laser, we need to detect this audio signal first. Therefore, to do this we used an electret microphone. It requires a biasing voltage to operate. Thus, we biased the microphone by connecting the positive terminal of it to the $1\text{ k}\Omega$ resistor. However, since the output voltage of the microphone is quite low, we cannot directly connect it to the rest of the circuit. In order to use this output, first, we should amplify it with a non-inverting amplifier circuit as shown in Figure 2.

There is a $10\text{ k}\Omega$ potentiometer connected between ground and the inverting input of the amplifier so that by changing its value, the gain can be adjusted, and the amplitude of the output signal is changed. The gain of the topology can be found by the following expression (1).

$$\text{Gain}, A_v = \frac{R_3 + R_4}{R_4} \quad (1)$$

Also, the simulation result of the input-output characteristics of the microphone driving circuit in LTspice is given in Figure 3 when $R_4 = 10\text{ k}\Omega$.

After non-inverting amplifier circuit, we connected a buffer circuit so that the microphone driver will not be affected from the rest of the circuit.

B. Automatic Gain Control

The second sub-unit of the transmitter part is Automatic Gain Control (AGC). We should adjust the output signal of the microphone driver circuit because the output of the microphone is distance and frequency dependent, so the output amplitude of the microphone change with time as well as distance of the speaker (person) to it. Therefore, as it is stated in project definition, we need an automatic gain

controller that controls gain and adjusts the amplitude of the microphone signal so that we will get a relatively constant amplitude audio signal at the output of the AGC regardless of the amplitude of the input signal. To achieve this, we construct the AGC circuit which can be seen in Figure 4.

In fact, the AGC circuit in Figure 4 is a negative feedback amplifier topology. The upper part of the circuit is basically a non-inverting amplifier whereas the remaining part is feedback network. The output of the microphone driver circuit is connected to the non-inverting input of the op-amp, and it is amplified by the gain of the amplifier which is determined by the resistors connected to the inverting input (2).

$$Gain, A_v = \frac{R_7 + R_8}{R_7} \quad (2)$$

At the output of the op-amp, a diode, resistor and capacitor is connected in series. Actually, it is a basic peak detector, and it is used to detect the peak voltage value of the output of the op-amp. Here, we used 1N4148 fast diode [x] since the change in the audio signal is generally fast. Then, the output of the peak detector is used to bias the BC547 NPN transistor so that the collector current is passing through the 100 k Ω resistor, and the voltage exists at the gate terminal of the 2N5460 JFET. Thus, as the amplitude of the input signal is changed the voltage at the gate terminal of the JFET is also changed. Also, since JFET can be used as a voltage controlled resistor, the voltage at the non-inverting of the op-amp is changed as well. So, this is how the negative feedback works in the AGC part. As a result, we get almost a constant amplitude signal at the output of the AGC regardless of the amplitude of the input signal. The simulation result of the AGC can be seen in Fig. 6 which shows output values for a 2 kHz sine wave with different amplitudes.

C. Low-Pass Filter

Although the frequency range of human voice is from 80 Hz to 14 kHz, in the project definition it is stated that the system only uses the portion of 300 Hz to 3.4 kHz. Therefore, we connected the output of the AGC to a low-pass filter before the addition of the high-frequency reference signal, so that the audio signal and the reference signal does not overlap in frequency spectrum. For filtering process, we utilized a 4th order Butterworth low-pass filter whose cut-off frequency can be found by (a).

$$f_c = \frac{1}{2\pi RC} \quad (3)$$

The circuit schematic of our low-pass filter is shown in Fig. x. Also, the simulation result of AC analysis of the low-pass filter in LTspice is given in Fig. x.

D. Summing Amplifier

Before transmitting the audio signal, we are going to add another signal namely reference signal so that at the receiver side, the amplitude of this signal will be treated as the measure of signal strength since it is constant. To sum up

these two signals, we used a basic summing amplifier circuit which can be seen in Figure x.

For the summing amplifier the output expression can be given as by the following equation (2). The output voltage has 180° phase shift since the audio and reference signals are given from the inverting input of the op-amp.

$$V_{out} = -\frac{R_3}{R_1}(\text{Audio_signal}) - \frac{R_3}{R_2}(\text{Reference_signal}) \quad (4)$$

The simulation result of the summing amplifier in LTspice is shown in Figure x.

In this summing amplifier we did not use a regular op-amp such as LM741 or UA741. Since we are working with high frequencies we had to use an opamp with higher slew-rate for better response. Thus, we used an LF351 opamp which has a slew rate of 16V/ μ s [x] whereas LM741 and UA741 op-amps has a slew rate of 0.5V/ μ s [x][x].

E. Laser Driver

At the transmitter unit, we need to convert the electrical signal to modulated light signal. To achieve this, we decided to use a laser rather than infrared or visible light LEDs because the visible light LEDs are very sensitive to the environmental noise whereas the infrared LEDs are not observable by naked eye so that it is hard to check whether the system works properly.

The light intensity of lasers is somewhat linearly dependent on the current, not the voltage as it can be seen from Fig. x. Therefore, the laser drive circuit is a transconductance amplifier whose input voltage coming from the summing amplifier, and the circuit schematic is shown in Fig.x.

In this circuit, we utilized the BC557 PNP transistor to construct a common-emitter amplifier with degeneration. As seen in Fig x. the light intensity of the laser is linear after some threshold current, which is also called as linear region. Thus, we want laser to operate in this region, so we need some kind of DC current offset. To achieve this, the resistors R2 and R3 is connected to the base terminal of the transistor so that the DC voltage at the base terminal is approximately equal to 7.5 V. Also, the simulation result of the DC operating point of the transistor can be seen in Fig. x.

It can be seen from Fig. x that DC current of the collector terminal, $I_c(Q1) \approx 30$ mA. This value is enough to operate the laser in linear region. In this topology we feed our signal from base terminal of the transistor, so we have a better small signal range. Also, we connected a degeneration resistor at emitter terminal of the transistor. Although the degeneration resistor decreases the gain, it improves small-signal range and provides stability to circuitry. Also, we connected 220 Ω resistor to the collector terminal because the maximum allowed operating voltage of the laser is about 6V otherwise it may be damaged.

The simulation result of the laser driver circuit is shown in Fig. x.

IV. RECEIVER UNIT

A. Photodiode

At the beginning of the receiver, we need to obtain light and convert it to a voltage signal. To do this, we decided to use the BPW34 photodiode [x] since it operates at visible light. The photodiode is a device which has a current changing with light on the device, so that we can use it to convert light into current. At this stage we decided to use a transimpedance amplifier to convert current to voltage. The schematic of the amplifier is shown in the figure below.

B. Low-Pass and High-Pass Filter

After the light signal is converted to voltage signal, we need to separate audio signal and reference signal. We need to implement a low-pass filter to obtain audio signal and a high-pass filter to obtain reference signal. We decided to use a fourth order Butterworth Filter [x] for low-pass filter and second order Butterworth Filter for high-pass filter. The schematics of the filters are shown below.

The $4.7k\Omega$ and $10k\Omega$ resistors are for amplifying the signal. The x resistors and capacitors are used to filter the signal. The cut-off frequency can be calculated by the following equation.

$$f_c = \frac{1}{2\pi RC} \quad (5)$$

The frequency responses of the low-pass filter are shown in the figure below.

C. Improved Peak Detector

At the proposal report, we decided to use a simple circuit with one diode and one capacitor to obtain the amplitude of the reference signal and it worked properly at the simulations. However, in practical case, it did not work as we expected since the frequency (20kHz) of the signal is too high. Therefore, we changed our design and used the circuit shown in [x] which is shown in the figure below.

The simulation results for the peak detector circuit is shown below.

D. Signal Level Indicator

We are expected to design a circuit to represent received signal level with a single RGB led. Chosen colors for each case are shown in the table below. The overall schematic of

TABLE I
LED COLORS FOR EACH CASE

Signal Level	Color	R Pin	G Pin	B Pin
No Signal	-	0	0	0
Weak Signal	Red	1	0	0
Moderate Signal	Yellow	1	1	0
Good Signal	Green	0	1	0
No Signal	Blue	0	0	1

the signal level indicator circuit is shown below.

At first, we used 5 resistors connected in serial between Vcc and ground to divide Vcc into four different voltages to determine regions for each signal level case. Then four comparator is used to determine at which region the amplitude of the reference signal is. When the amplitude is at the first region, all comparators will have negative output and no color will be displayed. When the amplitude is at the second and third region, the first and the second comparator will have a positive output and red and yellow color will be displayed respectively. The output of the third comparator is subtracted from the output of the first comparator by the difference amplifier so that the red pin will be turned off when the signal is at the fourth region and green color will be displayed. Similarly, the output of the last comparator is subtracted from the output of the second comparator so that the green pin will be turned off and blue color will be displayed when the amplitude is at the last region.

We used common anode RGB led so that the led has a common ground pin. We used different resistors for each color pin of the led to adjust the tone of the colors.

The simulation results are shown below.

E. Audio Amplifier

We obtained the audio signal at the output of the low-pass filter. But since it has a high output resistance, we cannot directly connect the speaker to the output of the low-pass filter. Therefore we need to implement a power amplifier circuit. The schematic of the power amplifier is shown below. We are expected to deliver at least 5W power to the speaker. The power of the speaker can be calculated by (x)

$$P = \frac{V^2}{R} \quad (6)$$

We use a speaker which has 8Ω internal resistance. Therefore, by (x), the output of the signal should be at least 8 Vpp. The simulation results of the circuit is shown below. From the simulation results, we can see that a sinusoidal signal with 4V amplitude. Therefore, we can say that the circuit can deliver 1W power to the speaker. In practice, however, our circuit is not worked as we expected since the BJTs we use has a current limit and burn at that current. Therefore, we could deliver a pretty less power and the audio can be heard from much small distance.

V. CONCLUSION

REFERENCES