

Design and Implementation of Audiometric Instrument Based on Microcontroller

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Abstract

The aim of this paper is to design and implement an audiometry testing instrument (Audiometer) by using PIC 16F877A microcontroller that has the capability to apply an audio sound to the patients' ear with a frequency range of (200 Hz to 4 KHz) with an intensity ranging from (-10 to 110 dB), and recording the patients' response through an input device. The instrument hardware was implemented by using a push button switch, headphone circuit, power supplies, two output units using 2 X 16 LCD display and RS 232 serial port interface connected with Personal Computer (PC) from through the ports COM 1 or COM 2 and microcontrollers programmer with its special program (Topwin 6).

The proposed design of audiometer and simulation results of many cases of hearing states are carried out using ISIS 4 professional simulator and Matlab 2010 environment. Finally, Proton IDE Basic compiler (high level language) is used to write a program which is employed to program microcontroller.

Keywords: Audiometry; PIC 16F877A Microcontroller.

تصميم وتنفيذ جهاز مقياس السمع بالاعتماد على المسيطر الدقيق

الخلاصة

الهدف من هذا البحث هو تصميم وتنفيذ جهاز الاختبار السمعي (مقياس السمع) باستخدام المسيطر الدقيق PIC 16F877A لارسال مدى من الترددات من 200Hz- 4KHz الى المريض ومعرفة استجابته لهذه الترددات من خلال وحدات الاخراج. ينجز التصميم المادي للجهاز باستخدام مسيطر دقيق PIC 16F877A واحد, مفاتيح ضغط, دائرة السماعه, مجهزات القدرة وحدتين اخراج ؛ عارضة LCD 2 X 16 ومنفذ التوالي RS 232 المربوط مع الحاسب الشخصي (PC) من خلال المنافذ COM 1 او COM 2 ومبرمجة المسيطرات الدقيقة مع برنامجها الخاص (Topwin 6).

ينفذ التصميم المقترح لجهاز مقياس السمع ونتائج المحاكاة لعدة حالات من حالات السمع باستخدام المحاكى *ISIS Professional* وبيئة برنامج *Matlab ٢٠١٠*. يستعمل برنامج *Proton IDE Basic* (لغة المستوى العالي) لكتابة البرنامج الذي يستخدم لبرمجة المسيطر الدقيق.

١. Introduction

An audiometer is used to measure the ability of a patient to hear at specific frequencies. Fundamental to this measurement is pure tone measurement. The audiometer is used to generate pure tone signals at specific frequencies within the ٢٥٠ Hz to ٨ kHz range. For each frequency the level of loudness is incremented from soft to loud. The patient is asked at which point he/she starts to hear the sound, which will then represent the patient's hearing threshold at that frequency. The final result is plotted as an audiogram that will be interpreted by medical professionals to determine proper treatments [١].

The weakest sound heard at a selected frequency is the hearing level in decibels (dB HL) for that particular frequency. This is a relative value; the intensity reference is ٠ dB HL, or audiometric ٠, which corresponds to the average threshold response (for a normal intensity range of -١٠ to +٢٥ dB HL) of a group of ١٨- to ٢٥-year-olds with no otologic pathology. The sensitivity of the normal ear varies with frequency; therefore, ٠ dB HL represents different levels of sound pressure at different frequencies. (Minus dB HL readings indicate that hearing sensitivity is greater for that particular frequency than for the average value) [٢].

At present, there are various types of audiometer readily available on the market. They can be different depending on the specifications and features, but generally a dedicated hardware is needed for high-quality and reliable measurements, resulting in high price. In terms of research, historically the earlier work focused on hardware implementation, usually at the integrated circuit (IC) or embedded system levels [٣, ٤]. Techniques such as direct digital synthesis (DDS) have been considered [٥]. As the availability of good-quality personal computer expanded, the focus has shifted towards PC-based systems [٦]. Some has also attempted on integrating other hearing loss measurement features into the same device [٧, ٨, ٩]. Finally, remote hearing scanning and active noise control based on PC have introduced in [١٠, ١١] respectively.

The PIC١٦F٨٧٧A microcontroller is manufactured by Microchip [١٢]. Currently they are one of the most popular microcontrollers, used in many audio applications [١٣-١٤]. This PIC has chosen to manufacture audiometric test for reasons of generating high precision and flexible pure-tone signal, digitally programmable, (frequency and amplitude of a waveform can be easily adjusted without the need to change the hardware components), in addition to; speed, digital ports ability, memory storage, and number of I/O (input or output) ports. This proposed system contains on push bottom switches, headphone circuit, display output units ٢×16 LCD design and serial communication design based on PIC١٦F٨٧٧A microcontroller.

This paper is organized as follows: The hardware design for audiometric test is first introduced in Section ٢, the software and simulink design is then described in Section ٣,

Testing and results are discussed in Section 4, and finally, concluding remarks and observations are given in Section 5.

2. Hardware Design

The block diagram of the proposed hardware which consists of five basic parts is shown in Fig. (1).

1. PIC16F877A Microcontroller.
2. Switches selection circuit.
3. Headphone circuit.
4. Output units.
5. Power supplies.

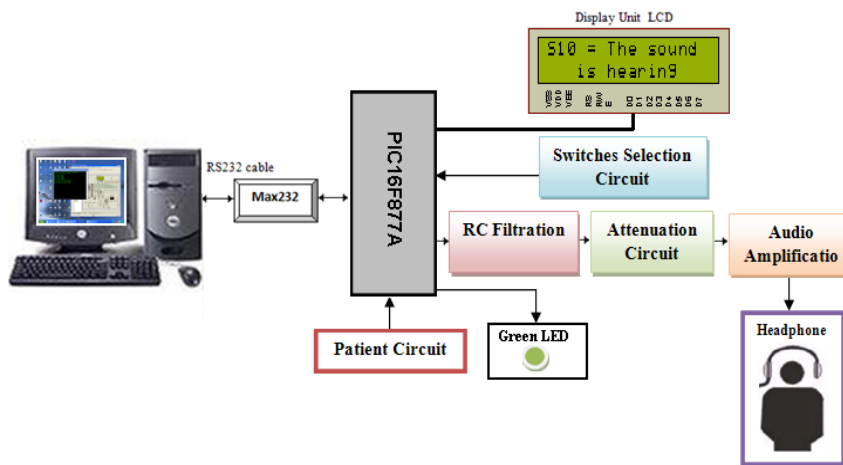


Fig. (1) Block diagram audiometric test based on PIC16F877A microcontroller.

2.1. PIC16F877A Microcontroller

Suitable for various digital applications, the PIC16F877A RISC microcontroller has five I/O port (33 lines), $8K \times 14$ flash program memory, 256×8 EEPROM data memory, three timer and 8-bit timer/counter and 10-bit ADC, at a clock rate of 20MHz, which is obtained by connecting a crystal oscillator between OSC1 and OSC2 pins with two 22pF capacitors. The built in power on reset circuitry provides a safe start-up, therefore the Master Clear pin (MCLR) is connected to reset circuit. The pin diagram of PIC16F877 is shown in Fig. (2) [10].

2.2. Switches Selection and Patient Circuit

The push bottom switches S_1 to S_4 are connected to microcontroller port B (RB_0 - RB_7) to select the range of frequencies from 20Hz-20KHz. S_5 is connected to MCLR pin to reset the microcontroller and S_6 is connected to the pin RD_6 to know patient's response, which is displayed in LCD or PC as message "The sound is Heard" in addition to light green LED connected to the pin RD_7 . The connections of push bottom switches are illustrated in Table (1).

Table (1) connections of push bottom switches with the PIC16F877A microcontroller pins.

Switches	PIC pin	Pin No.	Function
S1	RB0 (input pin)	33	Select 200 Hz
S2	RB1 (input pin)	34	Select 300 Hz
S3	RB2 (input pin)	35	Select 400 Hz
S4	RB3 (input pin)	36	Select 500 Hz
S5	RB4 (input pin)	37	Select 600 Hz
S6	RB5 (input pin)	38	Select 700 Hz
S7	RB6 (input pin)	39	Select 800 Hz
S8	RB7 (input pin)	40	Select 900 Hz
S9	\overline{MSLR} pin	1	Reset circuit
S10	RD7 (input pin)	30	Patient's response
Green LED	RD5 (output pin)	28	Patient's response

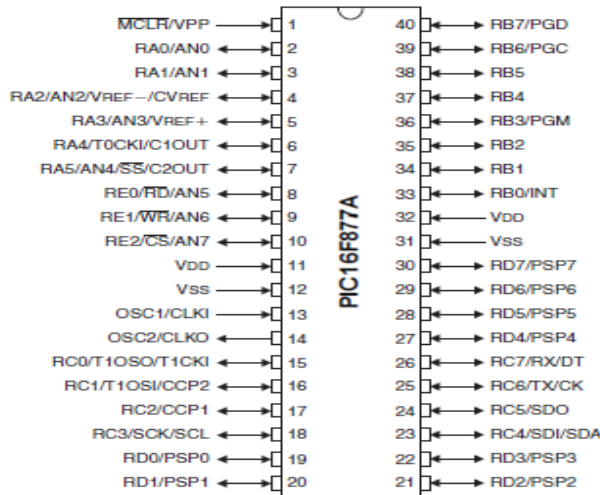


Fig. (2) Pin connection of PIC16F877A.

2.3. Headphone Circuit

The headphone circuit consists of three main parts; RC filtration, attenuation circuit and audio amplification. The PIC16F877A microcontroller generates a signal with one or two different frequencies on the specified PIN by using “Freqout” instruction. The generated signal is a square wave and filtering may be required to obtain a smooth signal and to remove the quantization noise are the attenuation module and amplification module. The attenuation circuit is implemented using a 6-stage (- 80, - 40, - 20, - 10, - 0, - 20, 0 dB) logarithm resistive

ladder network to achieve the 120 dB (-10 dB to 110 dB) dynamic range with a step resolution of 0 dB [8].

Since, the amplitude of the output signal may not be adequate, the audio amplification module is required to provide sufficient current drive and to match the load resistance of the audiometric headphone at 37 Ω [8].

2.4. Output Unit

Two output units were used in this paper, 2 × 16 LCD and RS232 serial port. The LCD is alphanumeric display, which is frequently used in microcontroller-based applications. Some of the advantages of LCDs are their low cost and low power consumption. (SC160C) is one of the most popular LCD modules used in the industry and also by hobbyists. The circuit diagram of the LCD and the PIC16F877A microcontroller is shown in Fig. (3) [16]

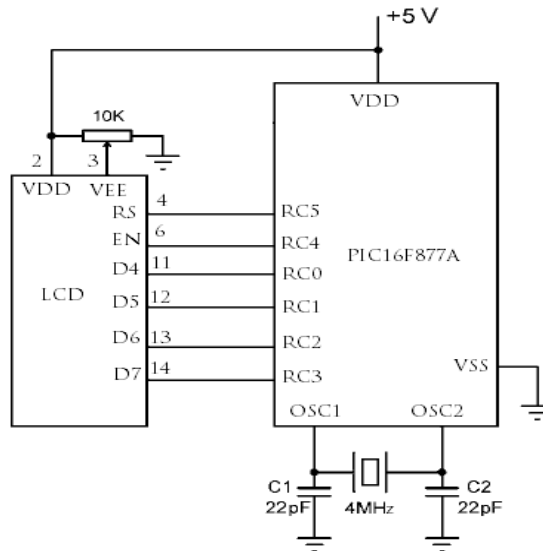


Fig. (3) circuit diagram between the LCD and the PIC16F877A microcontroller.

In addition LCD output unit, the RS232 serial port can be connected in this instrument. The pins RC4 and RC5 of the PIC16F877A are configured as RS232 serial output and input, respectively. RS232 voltage levels are ±12V. Normally RS232 voltage levels are converted to CMOS levels using RS232-level converter chips, such as the MAX232. An RS232-level converter chip converts the 0 to +5 V output from the microcontroller into ±12V RS232 levels. Similarly, the RS232-level output from a device is converted into 0 to +5 V suitable for the microcontroller inputs. MAX232 is a 16-pin IC having dual RS232 transmitters and receivers. This IC requires external capacitors for its operation to adjust the voltage level differences between the PC-based logic and the PIC-based logic as shown in Fig. (4) [17].

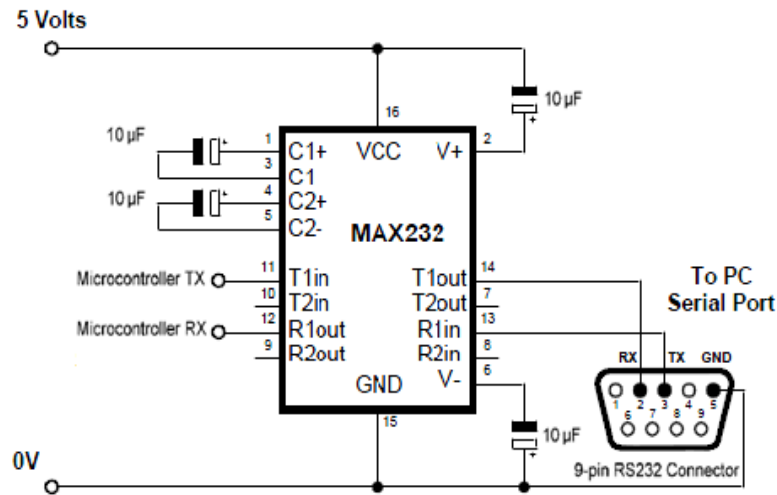


Fig. (4) MAX232 RS232-level converter.

3. Software and Simulation

The software is written using Proton Basic IDE; the Proton language is a high level language consisting of 160 instructions. The proposed coding by using Proton IDE compiler and Software flowchart of the audiometric test are respectively shown in Fig. (5) and (6).

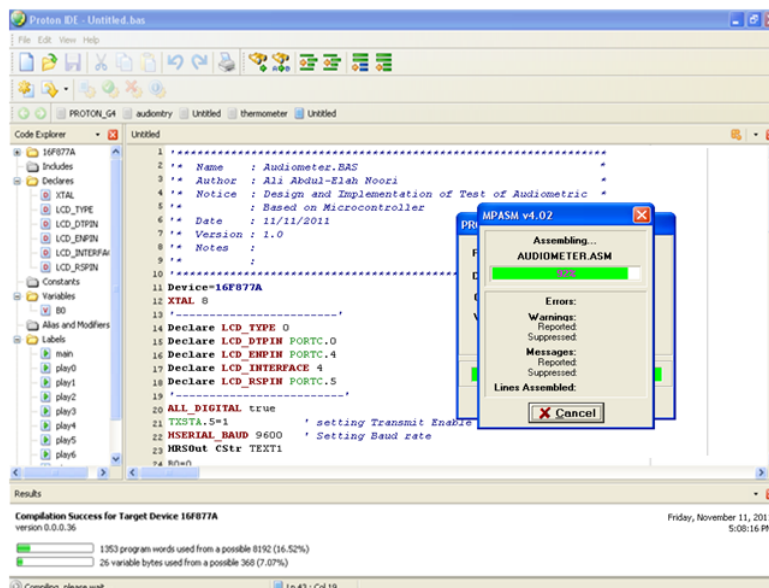


Fig. (5) Proton IDE coding of audiometric test.

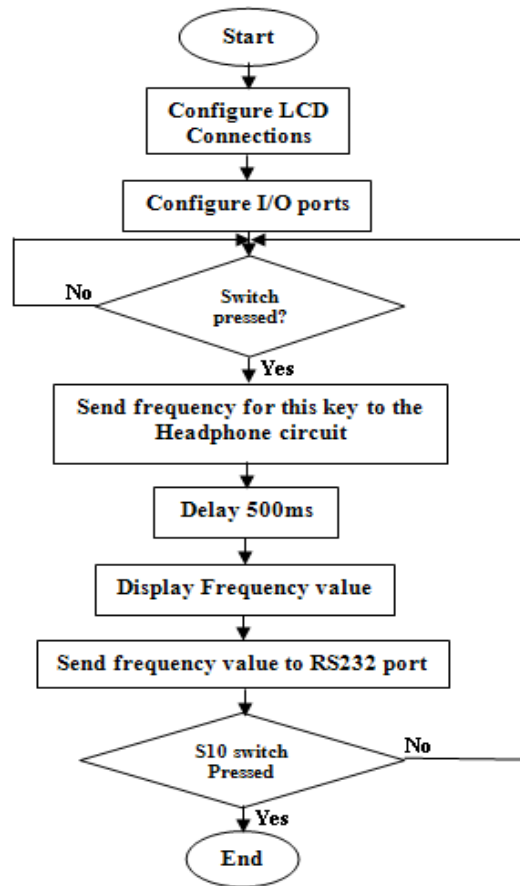


Fig. (1) The proposed flowchart of PIC microcontroller program.

The hex file of Proton IDE is downloaded into microcontroller programmer (Top programmer) through Topwin¹ Program which is accompanied with the microcontroller programmer. The Top PIC programmer device designed to operate with the USB port is shown in Fig. (2). After the PIC microcontroller programming is finished, the PIC^{16F877A} becomes ready to connect in hardware circuit.



Fig. (2) Top programmer and Topwin¹ program disk.

The hardware design is implemented by “ISIS Professional simulator” to simulate the electronic circuit of the audiometric test instrument as shown in Fig. (8).

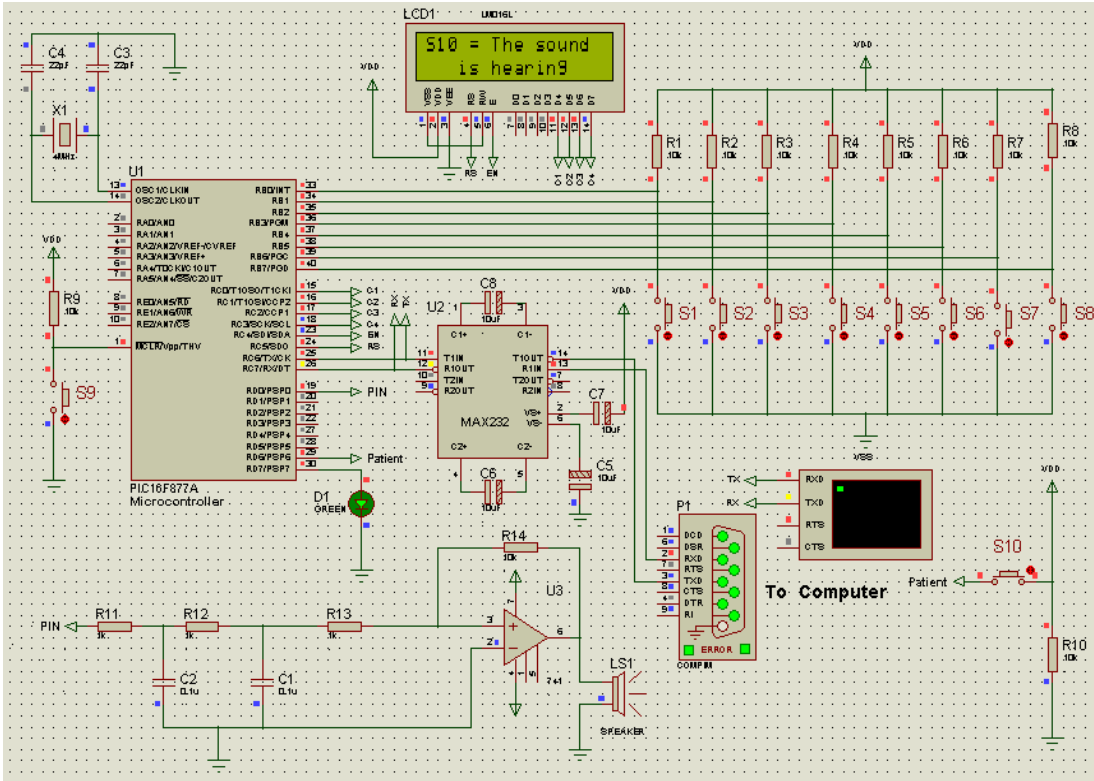


Fig. (8) Electronic circuit of test of audiometric instrument.

4. Testing and Results

In the experimental work, source code is written in the Basic language. The source file is then compiled by invoking the Proton IDE program. The code was tested using ISIS Professional simulator. When the examiner presses one of the S1 – S8 switches, the instrument applies a sound wave of the corresponding frequency to the patients’ ear through the headphone. Simultaneously, the frequency value is displayed on the LCD. The patient presses S10 when he hear the sound, causing a green LED to be lite and displaying a message of “the sound is heard” on the LCD screen. The values of frequencies are displayed on an LCD screen as shown in Fig. (9).

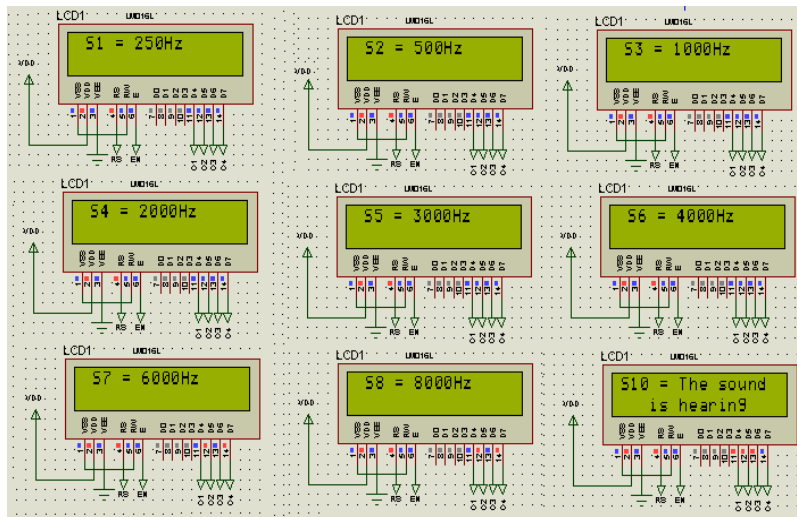


Fig. (9) Frequencies values display in LCD after pressing on different switches.

In addition LCD output unit, the RS232 serial port can be connected with COM1 or COM2 in PC to display values of frequencies through “Smarterm terminal emulation program”. This program can be activated on the PC to communicate with the audiometric instrument. The frequency range obtained is shown in Fig. (10) when the SmartTerm terminal emulation program is used.

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Audiometric Test Based on Microcontroller PIC16F877A
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Press Any Switch To Start...
S1 is pressed = 250Hz
S2 is pressed = 500Hz
S3 is pressed = 1000Hz
S4 is pressed = 2000Hz
S5 is pressed = 3000Hz
S6 is pressed = 4000Hz
S7 is pressed = 6000Hz
S8 is pressed = 8000Hz
S10 is pressed = The sound is hearing
    
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Fig. (10) Frequencies values display in PC after pressing on different switches

The simulation results of audiometer cases by using the proposed hardware are implemented in environment of Matlab 2010 program to examine the left and right ear in different case of hearing as shown in Fig. (11)

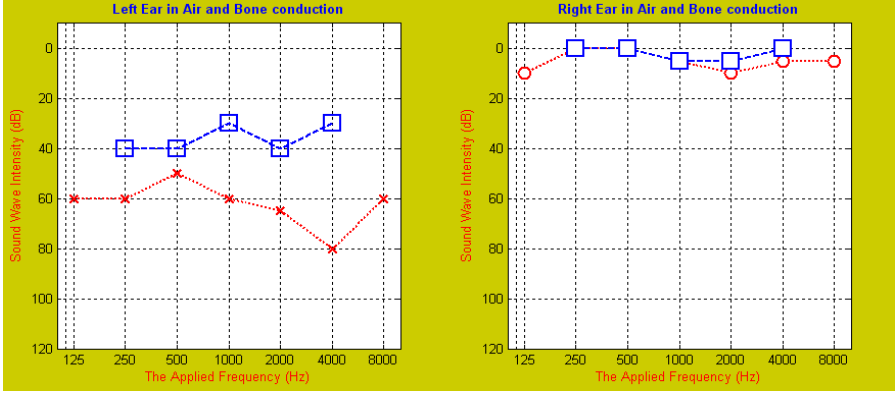
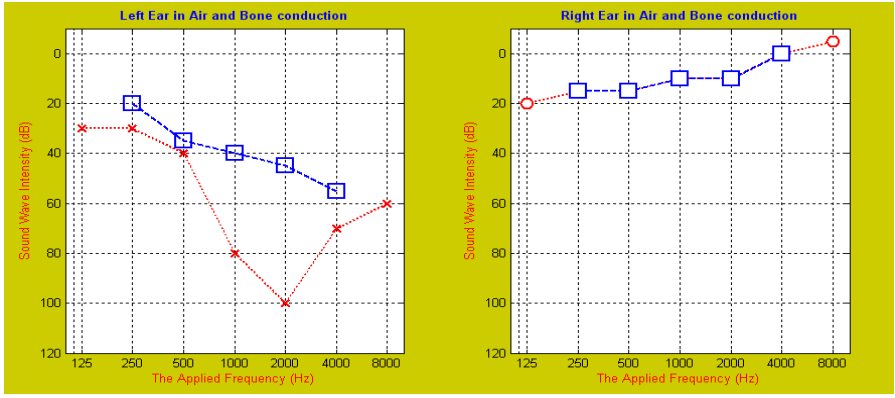
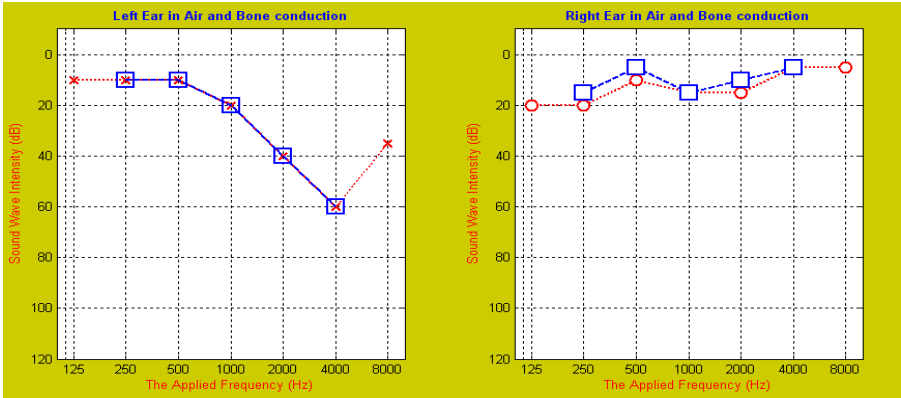
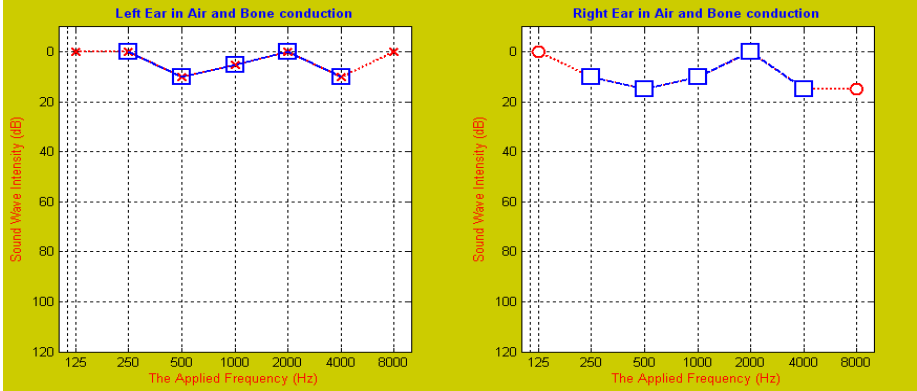


Fig. (11) Haering cases: (a) Normal case, (b) Sensorial hearing loss case, (c) Conductive hearing loss case, (d) Mixed hearing loss case.

9. Conclusions

Many conclusions can be derived in this paper; the most important results can be summarized as follows:

- a. The audiometric instrument based on PIC microcontroller has been designed and implemented successfully. The simulation and hardware implementation results have been presented to verify the feasibility of the system.
- b. Using PIC microcontroller unit, the frequency range and amplitude can be easily changed through programming without further hardware changes or by increasing push bottom switches connected to PIC microcontroller.
- c. The audiometric testing instrument based on PIC microcontroller offers high performance at low cost, and hence is suitable for commercial and industrial applications.
- d. The audiometric testing instrument based on PIC microcontroller is portable and easy to use.

References

- [1] A. Hemakom, A. Noymai, P. Israsena, S. Boonyanukul & C. Chinnarat, "Remote Hearing Screening As Part of Auditory Telerehabilitation; A Preliminary Report" *International Conference on Virtual Rehabilitation*, 27 – 29 June, 2011.
- [2] Bulletin, *Audiometric Testing: Information for Employers and Workers*, Alberta, January 2008. <http://industry.alberta.ca/whs-ohs>
- [3] M. Maloberti, S. Brigati, F. Francesconi, G. Grassi, D. Lissoni, P. Malcovati, A. Nobile, & Poletti, "An 0.8 μm CMOS Mixed Analog Digital Integrated Audiometric System" *Proc. IEEE Solid-State Circuits Conference*, , 1998, pp. 116 – 117.
- [4] P. Malcovati, "Design of Integrated Circuits for Audiometric Applications", *Proc. IEEE International Symposium on Circuits and Systems*, , 2001, pp. 171 – 174.
- [5] H.C. Wu, L.P. Yang, C.H. Chen, S.T. Young, & T.S. Kuo, "A Low-cost Architecture of Audiometer Based on Direct Digital Synthesizer", *Proceedings of The first Joint BMES/EMBS Conferences Serving Humanity, Advancing Technology*. Vol. 2, October, 1999, pp. 894, 13-16.
- [6] C.A. Martinez-Baez, L.F. Borjon, & A. Noyola, "Manual And Automatic Programmable Audiometer", *Proc. IEEE Inter Conf on Engineering in Medicine and Biology Society*, Vol. 13, 1991, pp. 1910 – 1916.
- [7] K. Kochanek, L. Sliwa, J. Zajac, & H. Skarzynski, "A Universal Computer Audiometer for Objective Hearing Testing and Screening," *Proc. Medical Measurement and Applications*, pp. 1 – 3, 2007.
- [8] S. L. Tan, S. K. Loh, & W. C. Chee, "Speech-Enabled Pure Tone Audiometer", *Proc. IEEE Intelligent Signal Processing and Communication Systems*, pp. 361 – 364, 2007.
- [9] Y. Faycal, B. Wahiba, B. Lotfi, B. Ratiba, & A. Benia, "Computer Audiometer for Hearing Testing", *Proc. Advances in Electronics and Micro-electronics*, 2008, pp. 111 – 114.

- [10] D. Azeez, M. A. Ali, H. Husain, G. K. Beng, C. Umat & R. Mustafa, "Implementation of Active Noise Control for Hearing Test Application Using PC" *International Confererence on Instrumentation, Communication, Information Technology and Biomedical Enginerring*, pp. 1-4, 2009.
- [11] Website of Microchip Technology, Inc. <http://www.microchip.com/1010/index.htm>,
- [12] A. Lay-Ekuakille, G. Vendramin, & A. Trotta, "Design of an Energy Harvesting Conditioning Unit for Hearing Aids" *3rd Annual International IEEE EMBS Conference*, pp. 2310-2313, 20-24 August, 2008
- [13] K. R. Kumer, & R. Seetha, "Development of Reciver Stimulator For auditory Prosthesis" *Internation Journal of Computer Science Issues*, Vol. 7, No. 3, 2010, pp. 23-29.
- [14] Cheng-Yuan Chang & Sheng-Ting Li, "Active Noise Control in Headsets by Using a Low-Cost Microcontroller" *IEEE Transations on Industrial Electronics*, Vol. 58, No. 5, May 2011, pp. 1936-1942.
- [15] A. V. Deshmukh, *Microcontrollers Theory And applications*, Tata McGraw-Hill, 2000.
- [16] I. Dogan, *PIC BASIC Projects Using PIC BASIC and PIC BASIC PRO*, ELSEVIER, 2006.
- [17] S. Lee, Y. Li, & V. Kapila, "Development of a Matlab-Based Graphical User Interface Environment for PIC Microcontroller Projects" *American Society for Engineering Education Annual Conference & Exposition*, S. 2220, 2004.