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Focus on circuit simulation

19:Rev./Chorus

September 2010

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15 configurable sound effects

The Elektor DSP Radio (2)

Antennas and PC Software

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BACKSIDE VIEW

A bit more on simulation!

In this edition, in accordance with our 2010 Publishing Plan, we pay special attention to the issue of electronics simulation. Simulation programs for the PC have been around for decades and in all that time they've become increasingly sophisticated and 'realistic'. But they're not the easiest programs to handle - you have to invest quite some time before you feel confident using them at any level. That is the main reason why many designers still seem to think "All that simulation takes too much time, I'll give it a quick try on the breadboard". But the time and effort in learning about simulation is really well spent. Once you have mastered a simulation program, you'll notice the ease of running such a simulation on a number of different options or settings in a thorough way, without any soldering to do or components to order by mail.

Two articles in this edition approach 'electronics simulation' from a different perspective. There is a hands-on article on the possibilities of the free program LTSpice, which is actually intended to develop switch-mode power supplies, but really has a much wider, largely undiscovered, application range. In 'Virtual Measurements and Predictions' a real-life case is presented of a simulation set up to compare the results of a simulation run against measurements carried out on the actual hardware thing. If you do not feel like doing any circuit simulation at all, no problem, there's a bunch of classic style projects in this edition to explore, assemble and use in practice. Like an advanced audio effects unit, a high voltage differential probe, a smart video detection system based on an inexpensive PIC microcontroller, and the second part of our beautiful DSP radio.

The Scrtach-and-Win sweepstake published in the July & August edition is getting a good response from you, our US & Canadian readership. If you haven't yet tested your good fortune, go find that edition #19/20. At www.elektor.com/ scratch we're waiting for your lucky number until August 31, 2010! Enjoy the read, the build... and the scratch!

Jan Buiting, Editor



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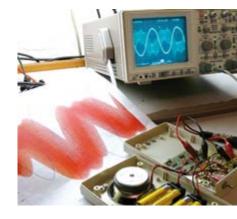
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36 The Elektor DSP Radio (2)

The operation of our famous DSP Radio can be optimized by modifying the antenna and front-end circuit configuration and by modifying the software. This creates various configurations ranging from a plain PC-controlled radio to a portable world receiver.



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Elektor International Media provides a multimedia and interactive platform for everyone interested in electronics. From professionals passionate about their work to enthusiasts with professional ambitions. From beginner to diehard, from student to lecturer. Information, education, inspiration and entertainment. Analog and digital; practical and theoretical; software and hardware.



No. 21, SEPTEMBER 2010

ISSN 1947-3753

Elektor aims at inspiring people to master electronics at any personal level by presenting construction projects and spotting developments in electronics and information technology.

Elektor (ISSN 1947-3753) is published monthly (except for one issue in July/August) at \$39.95 per year, Canada add \$11.00 per year; by Elektor International Media LLC, 4 Park Street, Vernon, CT 06066, USA. Phone: 860-875-2199, Fax: 860-871-0411. www.elektor.com Elektor is also published in French, Spanish, German and Dutch. Together with franchised editions the magazine is on circulation in more than 50 countries.

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Elektor US, 4 Park Street, Vernon, CT 06066, USA. Phone: 860-875-2199, Fax: 860-871-0411 Internet: www.elektor.com E-mail: subscriptionsus@elektor.com

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Low-cost weatherproof antennas suit 2.4 GHz applications

Pulse has introduced a new line of low cost, weatherproof antennas for 2.4 GHz devices used in industrial, smart grid metering, security, broadband access, and other types of machine-to-machine (M2M) applications. These robust, high-efficiency, compact antennas comply with relevant standards for Bluetooth, WLAN, WiFi, IEEE 802.11b/g, ZigBee IEEE 802.15.4, and other systems operating in the 2.4 GHz ISM band.



Pulse's W5001, W5010, and W5011 antennas are ruggedized for operation under severe conditions. They feature IP65 water ingress protection in accordance with international standard IEC 60529, UV protection, and the ability to withstand 100 mph wind loading. They have a maximum gain of 1.5 dBi in the 2,400 to 2,500 MHz range, an efficiency of 70%, and an operating temperature range of -40 °C to 85 °C. The antennas have a small footprint and low profile, measuring only 128 x 18 mm.

Pulse's W5001, W5010, and W5011 antennas are available in three SMA connector configurations: fixed right angle, straight SMA, and RP-SMA. They are packaged in quantities of 20 pieces per bag.

> www.pulseeng.com (100555-VII)

Elnec unveils new series of extremely fast multiprogramming systems

Elnec, a leading manufacturer of device programmers, has released a new series of extremely fast multiprogramming systems optimized for programming NAND and NOR flash memories. These new programmers are able to decrease programming times of NAND and NOR flash memories up to 70% in comparison to their predecessors. In addition to high programming speed, these new systems feature broad spectrum device support; they can program more than 53,000 device types.

device	Size (bits)	Program+Verity	
0825F640533	800200bx8 bit	1999	
(serial Flash)	(64 Mega)	30,7	
Am29DL640G	400060bx16 bit		
(saralle1NOR Flash)	(64 Mega)	24,0	
K8P6415UQ8	400100 tox 16 bit		
(paralle1NOR Flash)	(64 Mega)	13,0	
KIFIGOBUOM	8400000hs8 bit		
(NAND Flash)	(1 Giga)	122,7	
A MORE ALL DURING	10000Hx8		
ATB9C51 RD2	1 MANUAL MO		

The new series of multiprogramming systems, which are primarily intended to be used to program medium sized batches of device, comprises the BeeHive 204 and BeeHive 208S models.

The BeeHive 204 is a concurrent multiprogramming system designed for high volume production programming with minimum operator effort. The programmer has four independent programming modules. Each programming module starts programming when a chip is detected in its socket, regardless of the state of the other programming modules. For example, three of the modules can be carrying out programming, verification or blank checking while the operator is removing a programmed device from the fourth module and inserting a blank device. The BeeHive 204 reliably programs a wide range of programmable devices in its ZIF socket, with more than 800 models of socket converters available. Programming is also available through the ISP connector. Another special feature is that Elnec's software allows two BeeHive 204 programmers to be connected and operated simultaneously from one IIII PC for even more flexible production. Operating as a stand-alone concurrent multiprogramming system, the BeeHive 208S has eight independent programming modules. Aside from having more programming modules, this programmer differs from the BeeHive 204 in its stand-alone operating mode, which means that the programmer does not need to be connected to a PC while in use. The BeeHive 208S can be controlled easily by a touch screen display interacting with the internal Windows XP Embedded operating system. A 320 GB hard disc drive and software upgrade capability using a flash drive make this unit a reliable production solution with no additional costs.

www.elnec.com (100555-IX)

Multi-channel radio modems ease link setup

The TXL2 transmitter and RXL2 receiver modules from wireless solutions developer Radiometrix are half-duplex multi-channel OEM radio modems offered in the company's popular SIL standard footprint. Operating in the European license-free 433 MHz industrial scientific and medical (ISM) frequency band, with a choice of five user programmed channels, the modems can be used to form transpar-



ent one-way radio data links supporting a data rate of 9600 bps with a line-of-sight range of 500 m (typical). Unlike raw FM radio module alternatives, the modules look after preamble generation, synchronization, bit bal-

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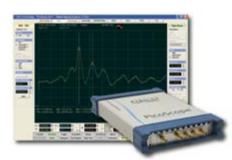
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ancing, error checking and noise squelching, as well as all transmit timing, valid data identification, and data stream buffering activities. This allows a data link to set be set up with a minimum of engineering effort. Each module can be assigned one of eight quasiunique addresses, enabling networks with multiple radio links as long as two modules do not attempt to transmit on the same channel at the same time. The transmitter has an output power of 10 dBm (10 mW), and the receiver has a sensitivity of –107 dBm. The modules require a 5-V power supply, drawing just 28 mA for transmitting and 22 mA for receiving. They have an operating temperature range of –20 °C to +70 °C. The transmitter and receiver both comply with the EN 300 220-2 and EN 301 489-3 standards. Key applications for these modules include handheld and portable terminals, industrial and commercial telemetry systems, data loggers, and in-building environmental monitoring systems.

www.lemosint.com/radiometrix/ (100555-V)

Saelig debuts 12 GHz TDR/TDT sampling oscilloscope

Saelig Company, Inc. has introduced the PicoScope 9211A, a new, economical instrument specifically designed for time-domain reflectometry, providing a low-cost method for analyzing the real-world characteristics of cables, connectors, circuit boards and IC packages.



The PicoScope g211A uses two independently programmable step generators with 100 ps rise time to stimulate the device under test and samples a sequence of transmitted or reflected pulses

via its 12 GHz sampling inputs to build up an accurate performance picture. The results can be usefully displayed as volts, ohms or reflection coefficient versus time or distance, all at a cost much lower than previously available.

Along with time-domain reflectometry (TDR) and time-domain transmissometry (TDT) analysis, the PicoScope 9211A can be used for mask limit testing against communications standards such as Ethernet, InfiniBand 2.5G and 5.0G, XAUI, ITU G.703, ANSI T1/102, RapidIO 1.25G and 3.125G, G.984.2, PCI Express 2.5G and 3.0G, and Serial ATA 1.5G and 3.0G. Over 150 industry-standard masks are included with the unit.

The PicoScope 9211A features three trigger inputs: a DC to 1 GHz direct trigger, a 1 to 10 GHz prescaled trigger, and a 12.3 Mbps to 2.7 Gbps clock recovery trigger, as well as an internal 10 Gbps software pattern sync trigger for averaging eye diagrams.

Unlike traditional bulky bench-top instruments, the PicoScope 9211A takes up very little bench space. The analyzer connects to a Windows PC via its USB 2.0 port, and an Ethernet port is also provided for remote operation over a network.

http://www.saelig.com/pr/9211a.html (100555-VIII)

Robotics & Electronics



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NEWS & NEW PRODUCTS

iNEMO design contest announced

STMicroelectronics and chip distributor Digi-Key have announced a design contest for North American engineering students, based on ST's iNEMO multisensor inertial measurement unit. The winning contestant will receive a \$5,000 cash prize along with an all-expense-paid trip, including airfare and accommodation for three nights, to Europe to visit the ST development laboratory in Catania, Sicily. Cash prizes of \$2,500 and \$1,000 will also be awarded to entrants selected as the second and third



place winners.

Entrants must be currently enrolled students, at least 18 years of age, who are residents in North American locations that do not prohibit, tax, or restrict such contests. They will also have to buy an iNEMO development kit, which costs US \$369 for the board, CD and cable, debugger unit and software.

Each entry must include an iNEMO inertial measurement unit. Entries will be judged on the number of degrees of freedom used, the quality of the hardware, firmware and software, the design's novelty, use of daughter boards, the project's ease of use, and the appeal of its user interface.

The iNEMO inertial measurement unit (IMU) is an evaluation and development board that offers three-axis sensing of linear, angular and magnetic motion as well as temperature and barometer/altimeter readings, along with a 32-bit microcontroller and dedicated software. It targets virtual reality, platform stabilization, human-machine interfaces and robotics applications. The module has a memory card socket for data logging and dedicated connectors for wired/wireless connectivity, including USB, ZigBee® and GPS.

A second generation of the iNEMO family, the iNEMO v.2 module, runs a sophisticated sensor fusion algorithm (attitude heading reference system) to provide static and dynamic orientation and inertial measurements. This 10-DOF inertial system integrates five different sensors, measures 4x4 cm, and comes with all necessary firmware and software.

http://www.digikey.com/iNEMO (100555-I)

Rohm launches multifunctional sound processors for car audio systems

Semiconductor manufacturer Rohm has announced the development of eighteen different sound processor ICs optimized for increasingly sophisticated car audio and home audio systems. The large number of inputs and outputs required to accommodate multiple audio sources and speakers for high fidelity playback, coupled with the need for fine tuning of audio systems to match the acoustic characteristics of specific environments, create a demand for high-performance, multifunctional sound processors.

To meet these needs, Rohm offers a complete lineup of eighteen different types of sound processor ICs with a variety of input source channels, speaker outputs, and acoustic correction functions. They feature low distortion and low noise, which are essential for high fidelity audio, along with advanced switching functions that reduce unpleasant switching transients by smoothing discontinuous waveforms resulting from volume level switching.



In addition, versions are available with integrated acoustic correction functions, including parametric equalization, loudness control, and subwoofer low-pass and highpass filtering that enable optimized audio playback based on vehicle cabin characteristics. All versions also share a common register map, allowing simple control with most existing microcontroller software and facilitating performance upgrades by simply swapping in a different sound processor. www.rohm.com (100555-II)

Liquavista releases new transflective product platform

Liquavista has launched its next-generation transflective display platform, called LiquavistaVivid, which is claimed to provide a level of performance unparalleled by other transflective displays. Combining monochrome reading with vibrant color imagery, LiquavistaVivid is a highly versatile, intelligent adaptive display technology specifically developed to support a broad range of feature rich, multi-functional consumer electronic devices.



LiquavistaVivid offers superior, high brightness monochrome reading both indoors and outdoors as well as hi-fidelity video and super-rich colors on demand. Depending on the environment and application, LiquavistaVivid displays can intelligently switch from a sharp, paperlike monochrome mode to a vivid color-rich multimedia mode with TV-like color. Each mode comes with seamless interactivity and hi-fidelity video capability. As with all Liquavista technology platforms, LiquavistaVivid displays have the added benefit of the widest dynamic addressing range of any display technology. This allows Liquavista displays to support hi-fidelity video without compromising on power, while at the same time supporting ultra-low power static reading modes.

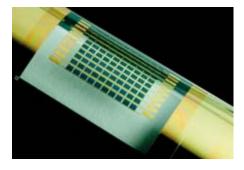
According to Liguavista CTO Johan Feenstra, Liquavista's very low power consumption enables users to be truly mobile by benefiting from longer battery life or lighter devices due to reduced battery requirements. LiquavistaVivid combines a high brightness reflective mode with a brilliant, high contrast, high brightness transmissive mode based on field-sequential color principles. In reflective mode, LiquavistaVivid provides the same high front of screen performance as LiquavistaBright, while in backlight mode it is able to display vivid colors without compromising battery life thanks to the high efficiency of the electrowetting optical switches and sophisticated power management algorithms.

http://www.liquavista.com (100555-III)

Low-cost GaAs solar cell fabrication method developed

Thanks to a new semiconductor manufacturing method pioneered at the University of Illinois, the future of solar energy just got brighter. Although silicon is the industry standard for most electronic devices, including the solar cells that convert sunlight into electricity, it is hardly the most efficient material available. For example, gallium arsenide and related semiconductors offer nearly twice the efficiency of silicon in solar devices, but they are rarely used in utility-scale applications because of their high manufacturing cost.

At the University of Illinois, professors John Rogers and Xiuling Li explored lower-cost ways to manufacture thin films of gallium arsenide that also allowed versatility in potential applications of the films. Gallium arsenide is typically deposited in a single thin layer on a small wafer, and the desired devices are made directly on the wafer or the coated wafer is cut into chips of the desired size. The Illinois group decided to





deposit multiple layers of the material on a single wafer, creating a pancake stack of thin gallium arsenide films.

Growing several layers on a single wafer reduces production costs because loading and unloading the wafers, with temperature ramp-up and ramp-down, takes a lot of time. The overhead savings of the multilayer approach, in terms of the machine, preparation, processing time and people, provide a significant cost reduction.

The individual layers must be peeled off and transferred. To allow this, the stack has alternating layers of aluminum arsenide and gallium arsenide. Bathing the stack in a solution of acid and an oxidizing agent dissolves the layers of aluminum arsenide, freeing the individual thin films of gallium

arsenide. A soft gripper picks up the layers one at a time and transfers them to another substrate, such as glass, plastic or silicon, depending on the application. Then the wafer can be reused for another growth. Freeing the material from the wafer also opens the possibility of flexible, thin-film electronics made with gallium arsenide or other high-speed semiconductors. Another advantage of the multilayer technique is freedom from area constraints, which is especially important for solar cells. As the layers are removed from the stack, they can be laid out side by side on another substrate to produce a much larger surface area, eliminating the restriction to the wafer size with the usual single-layer process.

http://news.illinois.edu (100555-IV)

Give your Projects Simple Grace

With the Elektor ProjectCase

A brief tour of the World Wide Web clearly shows one of the problems facing many electronic enthusiasts: the enclosure. All too often it's a plastic box in a dismal shade of gray, cluttered with an awkward arrangement of connectors, controls and indicators. You spent hours designing a stunning PCB layout, assembled the board neatly, and it ends up in a humdrum plastic box. Isn't there a better way?

The answer is yes, and it's easy. Peter Groen, one of the founders of FabLab in Groningen, The Netherlands, came up with the idea of using simple, transparent polycarbonate (Lexan) panels and a few standoffs. You simply fit the PCB in the middle, with room on all sides for wiring and connectors.

The display can be viewed directly and everything looks a good deal better, with the board nicely visible and well protected. To give you an idea of what it looks like, we used it to package the DSP radio board described elsewhere in this issue of Elektor. The polycarbonate panels are easy to machine, so making holes for controls, connectors, slots and whatnot shouldn't present any problems. Four sturdy rubber feet on the bottom keep everything rocksolid. Welcome to the ProjectCase!

To make things easier for you, we had a large number of ProjectCase kits made up for us. The panels measure 175 x 115 x 3 mm, which matches the Eurocard format.

The kit also includes all the necessary standoffs for securing the PCB and the polycarbonate panels. With a bit of creative imagination, you can do just about anything with it. The ProjectCase components are supplied with a protective film. Be sure to leave it on until all the holes, slots and so on have been made.

The price of the Project Case may be found on the Elektor website. We'll be happy to slip a ProjectCase kit into an envelope and send it to you so you can get down to work. Order your ProjectCase kits from the Elektor Shop at www.elektor.com.

(100500-l)





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Build a Scrolling LED Message Board in One Day using the mbed rapid prototyping platform

By Clemens Valens (Elektor France Editor)

When my friend Grégory showed me a 64×32 pixel two-color LED panel picked up from an Internet retailer for about 50 dollars, I immediately ordered one too. That would be a great opportunity to put the ARM mbed module I had lying around through its paces. I wondered, would this project be as easy as the mbed crew wants us to believe?

The first time I learned about mbed was when reading an article about it in Circuit Cellar #227. I was immediately seduced by the concept and wanted to try it. Having the right kind of relations I managed to force my way into the mbed beta testing program and got a couple of modules to play with. These beta modules were based on NXP's LPC2368 ARM7 processor. Shortly after mbed went live the ARM Cortex-M3 module using NXP's LPC1768 was introduced and I managed to get one of those too. All I needed to do at that time was find an application for these boards.

For those who did not read Tom Cantrell's article (you have no excuse as it can be downloaded for free [1]) I will first briefly explain the mbed concept.

So what exactly is mbed?

mbed (pronounced "embed") is a rapid prototyping platform for ARM processors, developed and maintained by ARM. Although two different mbed modules exist, both based on NXP chips, I don't think you can still buy the ARM7 one. Other modules using chips from other manufacturers may be expected, but don't hold your breath.

At first blush an mbed module just looks like any other microcontroller evaluation board with some LEDs, a reset button and a USB connector. It's small (25 ×53 mm), it's got 40 pins spaced 0.1" and there isn't much on the board besides the processor, five LEDs, a pushbutton and a USB connector. The top side of the board that is, because when you flip it over you a greeted by a second processor, a DB83848 Ethernet transceiver and an AT45DB161 16 Mbit serial flash memory chip. The second processor, an LPC2148 ARM7 (rebranded as 'mbed interface' on the current LPC1768 module), is almost as powerful as the main processor and this is where all the innovation is lurking.

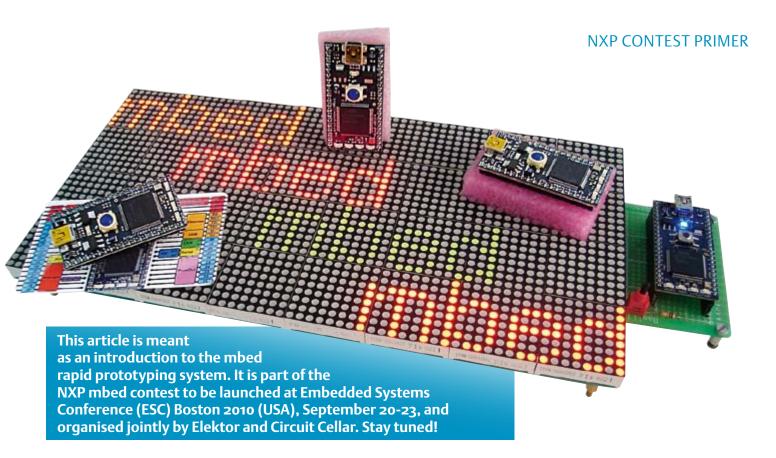
The 'mbed interface' makes the module appear on your host computer as an MBED (note the capitals) mass storage device and it contains one HTML file. When you open the file you are taken directly



Figure 1. This is the web page displaying when you open the HTML file on the mbed module. Create an account and kick off!

to the mbed login page (if you are connected to the Internet, of course), see **Figure 1**. After creating an account you can start programming right away, because the compiler (free and for unlimited use) is waiting for you online. That's right, there is nothing to install! Plug and play, you are up and running in like five minutes.

But that's not all. The mbed crew did most of the hard work for you by developing a library that supports all the peripherals of the processor and more. It was probably inspired by Arduino as it implements similar concepts like DigitalIn/Out and AnalogIn/Out. This library lets you concentrate on your application instead of getting bogged down in the swamp that goes by the name of Low-Level Register Programming. A real time saver, as ARM processors have about a zillion registers.



When your online source code compiles without errors the system suggests to download the executable. You save it on the MBED drive and then press the Reset button on the board. The mbed interface chip will now program the mbed processor and your application starts running. Sounds easy, eh? When your project is finished you can download it as a .zip file for archiving purposes or whatever.

If your program is not behaving as it is supposed to be, you're unfortunately more or less on your own. A debugger is not available, although you can add a serial port to the mass storage device so that you can print debug messages to a terminal on a host computer.

The mbed web site ^[2] plays an important part in the concept. Apart from the compiler you will find on the web site detailed explanations of the mbed library functions in the Handbook, sample projects in the Cookbook, a blog, a forum and more. You can publish your projects for other users to enjoy (hopefully) as well as import projects from other users into your own project. It's all very community-structured and open. The only thing that's definitely not open is the mbed library, which is of course very disappointing.

Why? Because it is now difficult to figure out if the library is efficient or buggy or suited or not for what you are trying to do. The mbed library implements the most common functions deemed useful in many cases, but it doesn't allow you to exploit the processor in all its glory. Understandable, sure, because if it would, it would be a hell of a lot more difficult to use, but annoying for those who need to know how exactly the library tweaks the bits and registers. Of course you can reverse-engineer the executable and inspect the registers, but that sure takes the 'rapid' out of the concept. Also, when the library is updated you may have to do it all over again.

And now that I am bashing, let me also mention the blue power-on LED on the board that shines so bright that it hurts my eyes even in clear daylight. When will designers finally understand that you do not *have* to force ten or so milliamps through a modern LED? A nice glow would have been more than sufficient.

Although the processor has 100 pins, the module only has 40. They

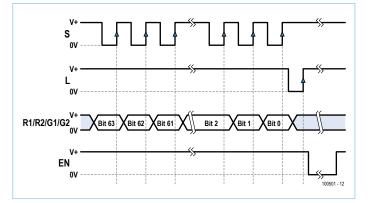


Figure 2. The timing diagram of the 32 line LED panel looks simple enough. However a subtlety is hidden in the four bit data word R1/ R2/G1/G2: R1/G1 control the upper 16 lines of the panel, R2/G2 the lower 16 lines. Therefore this pattern has to be repeated only

16 times to draw all 32 lines. The EN signal controls the power to the LEDs. Activating it (logic

Low) for a while after a line has been written gives nicer results than simply activating it all the time.

have been chosen in such a way that all the peripherals integrated in the controller are accessible, but not all at the same time. If, for instance, you need three serial ports, you can't have an I²C bus as well. The power supply (0 V, 4.5 V to 14 V in, USB 5 V out and 3.3 V out), the Ethernet interface and the USB port have dedicated pins, there is a reset input and two IF pins (these are for the USB port of the mbed interface, not the target processor and you are not supposed to use them). The remaining 25 pins can be GPIO or allocated to some other function.

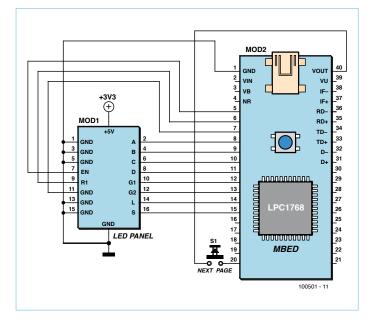


Figure 3. The circuit diagram is really simple since no special parts are needed to control the LED panel. Note that in this basic setup the mbed module is powered from the USB bus, whereas the LED panel needs a separate supply because it needs more current than mbed can provide. Also note that even though the LED panel is specified for 5 V, it will run from 3V3 without any problem. The panel at 5 V driven by a 3V3 mbed will work too.

On to the application

The goal of the project was to build a message board with an easy way to put, well, messages on it! Today's electronics often equates to USB and that's why I ended up using the mbed module. Since the module doubles as a 2 MB USB mass storage device it would be very easy to load message files on it produced on a computer with a proper text editor. That would be an intuitive and comfortable user interface that even non-electronics engineers can understand. The user manual of the LCD panel (DE-DP-DP029~033_Ver1.0_ EN.pdf^[3]) told me I had to do some bit-banging as its serial interface is not standard SPI or something similar. Figure 2 shows the timing diagram. The display has 32 rows of 64 two-color LEDs, i.e. a grand total of 4096 LEDs! The addressing of the 64 rows (32 twocolor rows) is a bit strange because the address bus is only four bits wide and can thus only address 16 rows. This problem was solved by the manufacturer by breaking the display in two 16 row displays and using a 2-bit data bus. The two 16-row displays are addressed in parallel and the two data bits select the red and green LED for the row. The display has only enough memory for two rows, one upper row and one lower row. This means that the display has to be refreshed continuously in order to display a message or image. An advantage of this technique is that the display will not consume 40 amps or so when all the LEDs are on.

After a line is written it should be activated using the EN signal. This signal determines the brightness of the display but as it does so, it also affects the refresh rate. This will in turn govern flickering as well as the scroll speed, since scrolling does not use a separate timer and depends directly on the refresh rate.

A little bit of hardware

How I hooked up the display to the mbed module is shown in Fig-

ure 3. Since most mbed pins can be GPIO, I simply started at the first available one (pin 5) and worked my way down. I later added a special feature pushbutton to pin 20 simply because it was nearest to the pushbutton.

For compatibility reasons I ran the display from 3V3 (3.3 V) instead of the specified 5 V, because mbed is 3V3 and the display let me do it. Once I mastered the display I powered it from 5 V and that worked fine too, the advantage being a much higher brightness. The display was powered from a separate bench supply as it requires several amps when all the LEDs are on and the mbed module cannot provide this much. I powered the mbed module itself through the USB port.

I had a bit of a hard time implementing the not terribly complicated serial protocol for the simple reason that the user manual of the display has an error. In the table on page 8 and 9 that I was using the pin numbers of the 'L' and 'S' signals are swapped with respect to the connector pinout shown just above it. The sample code I had extracted from the document worked fine, but my own code didn't. Very frustrating. With this problem out of the way everything went smoothly and I was able to very quickly plot a pixel anywhere on the display in one of the four available colors (red, green, orange and black).

It's interesting to note at this point that the mbed library has pin and bus functions. It seems logical to use the BusOut function for both the address and the data buses, but it turned out that the bus function is much slower than the pin function. I therefore ended up using four DigitalOut calls for the data bus and one BusOut call for the much slower address word. The reason for this performance difference would probably be clear if only the source code of the mbed library had been published.

Dots & fonts

It was now time to start writing characters on the display. Characters need a font and bitmapping a font is an intensely boring and tedious job. In the spirit of letting other people do the hard work for you, I turned to the Internet. Many people have been busy over the years bitmapping fonts and writing utilities to do it automatically, so after some ten minutes I found myself playing with The Dot Factory ^[4] (**Figure 4**). This Windows utility makes it a snap to convert any font installed on your PC into C source and header files. Select the font, set some options and click the Generate button. The only weird thing is that you can't save the files; you have to copy and paste. That's a minor inconvenience though for a nice free tool, so I created the files without complaining and imported them into my mbed project. Adding the code to display the characters anywhere I wanted them was pretty straightforward too.

Adding an INI file system

The next step was adding a way of putting messages on the system. This was the main reason why I had opted for an mbed module: it has a USB interface and the mbed library provides a convenient file system. And convenient it was. All I had to do was to add 1 (say, one) line of code to my program to get it working:

NXP CONTEST PRIMER

LocalFileSystem local("local");

Now my program was able read the files that I copied to the MBED mass storage device from my PC.

I had decided to use an INI-type file for my message files. I wanted several pages, browsing through the pages would be possible thanks to the pushbutton. Every page would be a section in the INI file and it would be possible to specify, on a section by section basis, the text for each line. With my 8-point Arial font I had room for four lines per page without characters spilling into other lines. I turned again to the Internet to look for an easy INI file C-library. There are many around and I picked one more or less randomly: inih ^[5] it would be! This turned out to be a good choice as it's a remarkably easy to use library and no more than five minutes were needed to integrate it into my project.

Time to stop

I had now come to a point where 'features' started creeping in. First of course message scrolling: left, right and no scrolling. Then scrolling at any speed. Now that was a tough one. Not so much to implement as it was to look at, because it resulted in pixel bleed. Characters became blurred and the display became uncomfortable to read. This feature was therefore scrapped, but another one quickly found: every line its own colour and its own (x,y) starting position. I tried many more things but finally settled for the features mentioned above plus page skipping. Then I called it a day.

Tool trouble

Building this application took me about one day of work. I did not do it all in one go, but I could have if my agenda had let me. Using the mbed module instead of some other microcontroller board definitely saved enormous amounts of time as did the mbed library. Not a single ARM register was inspected by me and I did not have to refer to the processor's datasheet or programming manual at all. The little mbed experience I had before starting this project probably helped a bit too, because I knew what I could expect from the mbed library.

The only severe hindrance was the mbed server going down or offline. This happened once during my project, wasting a couple of hours and it makes you feel silly. There you are happily programming away and suddenly: Bam! No more tools... And no one you can yell at (what was that mbed helpdesk phone number again?). Another inconvenience is that the compiler or the server seems to get stuck every once in a while. This is rather annoying as it probably means loss of those clever modifications you just added to your code. Exactly what happens I don't know, but you have to close and then reopen the compiler window without being able to save your work. The mbed crew should add some temporary saving for recovery purposes to the environment as this happened to me several times.

Note that you do not *have* to use the online compiler. The executable that you download and program into the target is an ordinary binary, i.e. no proprietary file format. If you can produce such a file with a toolchain you already own you're good to use the mbed board too.

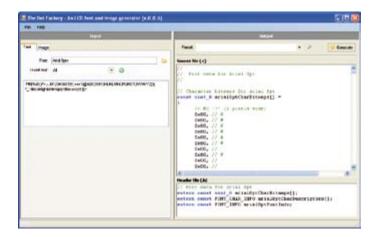


Figure 4. The Dot Factory: Now why did the author(s) forget to implement file saving?

The mbed library can also be used since the header files and a binary archive (.ar) can be downloaded. This is not the easy way however, and, understandably, not really supported by the mbed crew. However, some people have found ways to mbed off line.

Similarly you don't need an mbed board to run the executable. Any LPC1768 board can be used, which means that you don't have to design an mbed module into your product.

Conclusion

While working on the project I was using the beta LPC2368 mbed, but for this article I had to use the LPC1768 mbed. Switching between the two modules turned out to be extremely easy; all I had to do was selecting the right mbed from a dropdown list on the compiler menu. The recompiled code then worked immediately without any modifications.

All in all I think mbed is pretty cool. This project made very low demands on the power of mbed, the Ethernet interface is worth a try too. Although I would not recommend mbed for professional or consumer quality software development since you have little or no control of the tools and the library (what if ARM decides to pull the plug on mbed?) and there is no debugger, it is definitely a great tool for quickly putting a proof of concept together. It is also great for those one-off projects where you need just a little more processing power than you can squeeze from systems like an Arduino. Hats off to the mbed crew who have come up with a great too!

(100501)

Internet Links & Literature

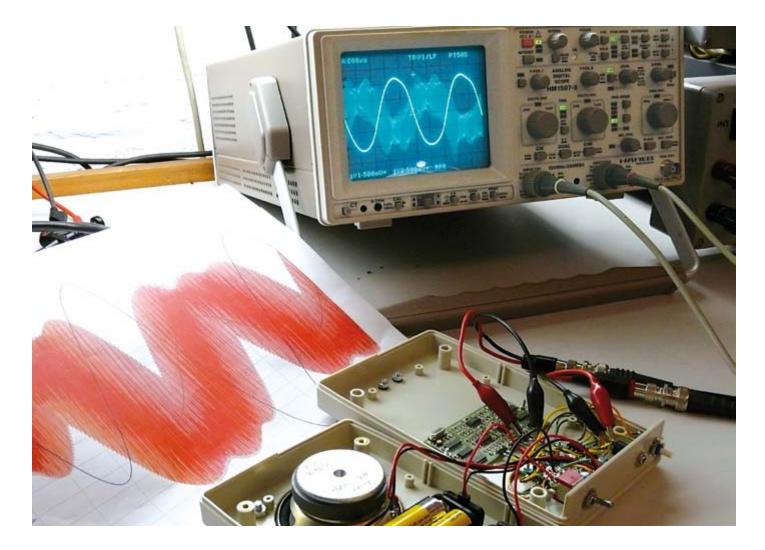
- [1] Easy (E)mbed, An Alternative Approach to Embedded Programming, Tom Cantrell, Circuit Cellar #227, June 2009, pp 68-72.
 (Free download at www.circuitcellar.com/archives/viewable/ Cantrell-227.pdf).
- [2] mbed.org
- [3] LED panel: www.sureelectronics.net/goods.php?id=718
- [4] The Dot Factory: www.pavius.net
- [5] INI Not Invented Here: code.google.com/p/inih/
- [6] Software: www.elektor.com/100501

TEST AND MEASUREMENT

Virtual Measurements and Predictions

By Thijs Beckers (Elektor Netherlands Editorial)

As an electronics enthusiast or engineer, you'll know that the design of electronic circuits can be complex. These days, however, you can get help from computer programs, so you'll know in advance if your design will function as intended. Or perhaps you won't? We compared the measurements made on a real hardware prototype with the simulated data produced by a leading program and drew our conclusions.



These days the design of electronic circuits happens on the PC. In a completely virtual environment the circuit diagram ('schematic') is drawn, its operation is simulated and the circuit board layout is produced. The designer puts a lot of trust in the simulation results. To be completely sure, however, a (hardware) prototype is usually built that can then be used to confirm that the circuit works as it should. But how good are those simulations really? We decided to try out Proteus, from Labcenter Electronics [1], a leading CAD package for circuit drawings, PCB layouts and simulations.

Circuit simulations: how reliable are they?

For the simulation we picked the circuit for the frequency shifter from the portable sound system with feedback suppression, which was published earlier this year in the February 2010 edition of Elektor. This circuit contains analog as well as digital building blocks, and it isn't the easiest design to simulate, partly due to the high switching frequencies and phase shifts.

We picked several interesting points in the circuit where we would like to see the waveform of the signal; these are shown as A to F in the circuit diagram on the following page. The most interesting one is of course the output signal (F), but designers obviously also like to know what happens at intermediate stages of the circuit, so that they can be sure that the circuit behaves as expected.

CAD

To simulate this circuit we'll use the Proteus 7 Professional package from Labcenter Electronics. This package in fact consists of two programs: ISIS and ARES. ISIS is used to draw the circuit diagram and simulate it, whereas ARES is used to design the PCB layout. In this article we're not interested in the PCB layout so we'll ignore the latter program.

The Proteus Design Suite is specifically made to simulate embedded systems and has the facility to simulate microcontroller code in a mixed-mode SPICE environment. Labcenter calls this VSM, Virtual System Modelling. Of course it can also cope with 'normal' SPICE models. We aren't really asking a lot of the simulation software with our frequency shifter, but the extensive functions of the software will be welcomed by many.

Before you can start the simulation you have to input the circuit into ISIS. Unfortunately, this means that the schematic has to be completely redrawn. The various CAD packages on the market all have their own format for storing the drawings and generally aren't able to export or import other formats.

Once the circuit has been drawn, you can add probes and measuring instruments

to your hearts content. Theoretically, you could add a complete arsenal of measuring instruments to each point in the circuit, but this is of course not necessary. Having selected several interesting points in the circuit and put a virtual probe there, you are in a position to visualize the signal at those points.

Simulation

When the probes have been put in place, you can choose several virtual measuring instruments, in this case three oscilloscopes and an FFT analyzer. For each of the virtual instruments you have to specify which probe they have to use. Next, you can start the SPICE simulation by hovering the mouse pointer above a measurement window and pressing the space bar. Some time later (depending on the processing power of the PC) the simulated voltage appears on the screen.

Tips and tricks

The opamp models chosen for the simulation were the 'ideal' opamp models, which reduces the number of required SPICE nodes from a hundred down to only a dozen per opamp. This speeds up the simulation significantly, but still offers sufficient accuracy for this application.

The simulation for the clock generator around the crystals has also been turned off at the component level. The SPICE model for the 4060 has a parameter ('clock') allowing the operating frequency of this component to be set, irrespective of the value of the clock signal at its clock inputs. You could even have chosen to replace the 4060 and 74HC74 flip-flops with a pulse generator that output the required clock pulses, but this would have deviated too much from the original circuit.

These 'tricks' are very useful to know and give speedy and good results. It's therefore important that you know what simulation options there are and what settings give the best results. For other tips and tricks you can refer to the manual, the technical support department and last but not least, several forums where you can ask any questions you may have.

Results

The details of the simulation results are shown in the spread on the following page. We were pleasantly surprised by the fantastic results. The measured and simulated values corresponded almost perfectly. The real-world measurements only deviated slightly from the ideal (simulated) values. The slight mismatches were probably caused by component tolerances. Not only could differences be measured between a number of frequency shifters that had been built, but there was also an audible difference. Some clearly sounded better than others did.

Hats off for Proteus, which in our opinion passed the 'test' with flying colours. Not only is the end-result (signal F) amazingly correct, but all the intermediate points that have some rather 'strange' waveforms also correspond to the measured signals. No complaints here then.

Do It Yourself

On the web page created for this article [2] you can find the ISIS circuit including the virtual measuring instruments for you to download. A demo version of Proteus 7 Professional is available for download as well, so you can verify the results above and carry out further simulations if you want. The demo version of Proteus does not allow you to save your designs, nor create print layouts, nor create circuits containing micro-controllers. However, you are able to modify the program code in existing (example) designs with microcontrollers.

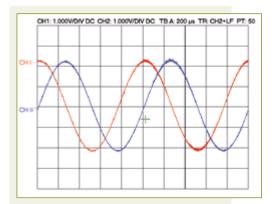
As an example, try to run a simulation of about 20 ms or more, where the input signal and the inverted output signal are shown in one graph. You should see that the signals have a slight difference in frequency and that one signal appears to 'overtake' the other (double-clicking on the title of the scope image opens up the graph in a new window).

(100359)

TEST AND MEASUREMENT

Hardware versus software

On the left-hand page you can see the measurement results for the measurements carried out on the real-life circuit, while the righthand page shows the results of the simulations. On the circuit diagram a number of points (A to F) are indicated that we thought would be interesting to measure and compare with the simulation. The signals at points A and B should have a phase difference of 90 degrees



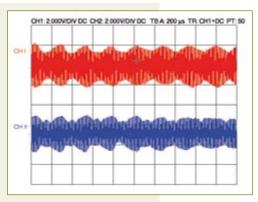
[1] Signal at A and B

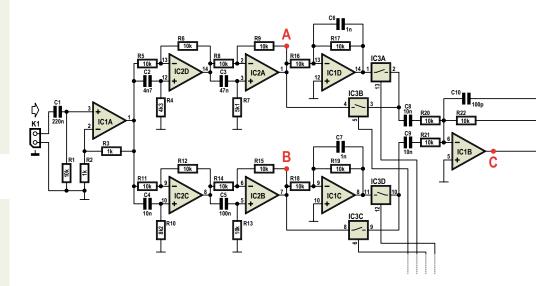
The signals at these points should be the same as the input signal, but with a phase difference of 90 degrees between them.

That means that when the voltage of one of the signals is at zero, the voltage of the other signal should at its maximum or minimum (depending on the phase). This is more or less the case in the measurement; it isn't exactly 90 degrees, but it comes very close.

[2] Signal at C and D

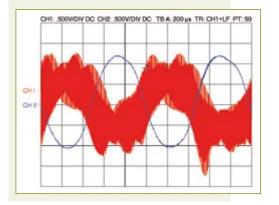
At point C is the high frequency square-wave modulated signal. At point D is the filtered version of the signal at C. Due to the small difference in the fundamental frequency and the first and subsequent harmonics there doesn't appear to be much difference between C and D.





[3] Signal at E and F

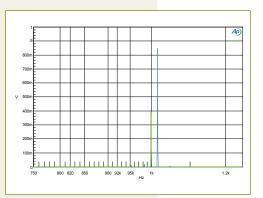
At point E you can almost recognize the output signal (modulated by various mixer products). At point F all redundant RF signals have been removed from the signal and the result of the frequency shifter becomes apparent. The output sinewave is slightly distorted, which is partly caused by the tolerances of the components used.



[4] FFT output signal

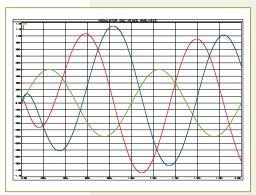
When we perform an FFT analysis on the input and output

signals it's easy to see the frequency shifting that has taken place. The original signal had a frequency of 1 kHz, whereas the output signal was about 1.01 kHz, i.e. it has become 10 Hz higher, which was the intention of this 'feedback-killer'.

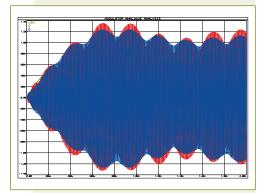


TEST AND MEASUREMENT

between them. C and D should be the modulated and the filtered, modulated signal respectively. At E is the output signal with a nasty looking RF signal superimposed on it and at F is the clean input signal, but with an increased frequency of about 10 Hz.

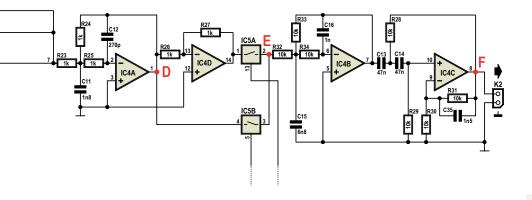


[5] Signal at A and B In the simulation you can also see that the phase shift isn't exactly 90 degrees. It's clear that the design for the phase shifter is a compromise; the bandwidth in which the circuit works 'perfectly' is limited.



[6] Signal at C and D

Despite the complex RF waveform of the signal, the SPICE simulator from Proteus keeps its cool and shows a very close resemblance to the measured signal from the actual hardware.



Internet Links

[1] www.labcenter.com

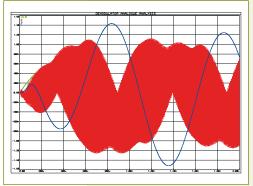
[2] www.elektor.com/100359

[8] FFT output signal

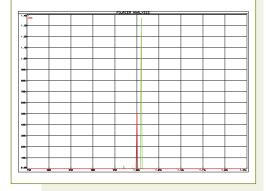
Even the simulation of the FFT of the output signal shows an exact 10 Hz frequency shift... What more could you ask for?



There is not much to say about the signal at E and the



filtered output signals. The only difference is that the hardware version has some distortion due to the component tolerances, which is something that the software hasn't taken into account in this case.



Simulation Beats Trial and Error

By Dr Günter Spanner (Germany)

Simulation tools allow circuits to be tested and characterised even before the first components have been ordered. LTSpice is a powerful program and available free of charge. Here we use two examples to show how this program can be used for circuit analysis, among its other features.

Simulation programs have become indispensable for the commercial development of electronic products. The traditional development method for new products, consisting of building a lab prototype, designing an initial PCB layout and producing a pilot series before final production release testing, became outdated a good while ago.

Nowadays new circuit designs are thoroughly tested before the first real components are ordered. In addition to basic circuit functionality, this covers the full allowable operating voltage range and beyond – which does not incur physical risks with a simulation. The worst case situation in which the circuit must continue to operate reliably can be determined with a computer simulation by configuring a wide range of temperatures and highly improbable component tolerances.

Until a few years ago, usable simulator programs were as dear as hen's teeth, with prices of full-function versions easily reaching the three-figure range. Today tools with outstanding features are available on the Web for free download. This makes circuit simulation extremely attractive even for non-professional users.

In this article we describe the LTSpice simulation program (previously known as SwitcherCAD), which is available free of charge from Linear Technology [1]. The core of the program, which computes the actual simulation, is based on the well known Spice simulation program, but Linear Technology has refined this program, in particular to obtain better results for switching converters (which are after all their main products). A convenient user interface has also been wrapped around the simulation software to provide features such as visual schematic design and editing. For more

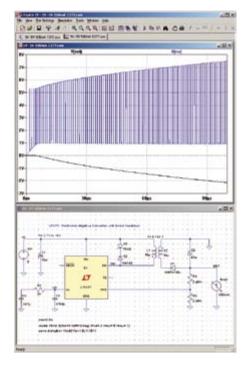


Figure 1. Convenient presentation of the schematic and the simulation results (example from the LTSpice installation package).

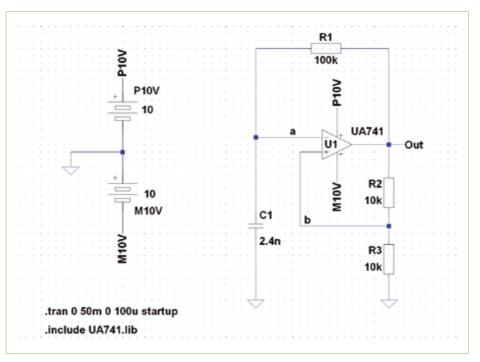


Figure 2. A square wave generator. The opamp is simulated using a circuit not shown here; the circuit model is located in the UA741.lib file.

Practical experience with the free simulation tool LTSpice

information on Spice and a good introduction to how simulation tools work, see the October 2006 issue of Elektor [2].

Schematic editor

Installation of LTSpice is easy and does not require any explanation here. The implementation of the schematic editor is quite nice and more than justifies the use of this program. In addition to basic component symbols, the program comes with an extensive component library. Most circuit symbols are US style; a European resistor symbol is available in the library under the 'Misc.' category.

The individual components can be positioned freely and interconnected easily (a few notes for users are provided in an extra file that can be downloaded from the Elektor website [3]). All components can be rotated and flipped as necessary to produce a readily understandable schematic diagram. There is also a component editor that allows users to generate their own component symbols.

Semiconductor devices in particular are composed of a large number of internal components, so that their behaviour can be simulated as realistically as possible. These subcircuits are represented by Spice models, and the necessary data can often be obtained free of charge from the manufacturer or elsewhere on the Web. As a Spice derivative, LTSpice naturally supports integration of these Spice models.

Figure 1 shows a typical working environment for circuit analysis. The schematic is shown in the bottom window, while the simulation results are displayed in the top window. The inverting voltage converter shown here is an example circuit included in the library of the installation package. To find out what this simulation program can do, we tested it with two relatively simple example circuits.

Example 1: Square wave generator

Our first example simulation is a simple square wave generator. The circuit shown in Figure 2 is built around a standard opamp

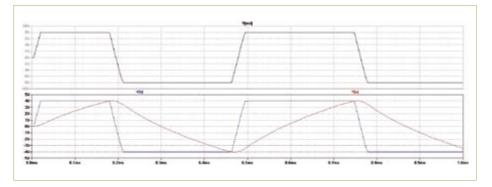


Figure 3. The waveforms at test points 'a', 'b' and 'out' (see Figure 2).

and should be familiar to every electronics enthusiast.

After the schematic is entered, the model for the operational amplifier (a 741), which can be found at various places on the Web including [4], must be saved in the ROOT:\ Program Files\LTC\LTspiceIV\lib\sub folder. There it is available to all simulations. Alternatively, you can select an opamp type available in the standard library.

After you start the simulation, you can use a virtual probe to measure voltages, currents, power levels and other quantities at various points in the circuit. The corresponding waveforms are shown graphically.

Figure 3 shows a virtual oscilloscope display of the signals at three test points:

a: the inverting input of the opampb: the non-inverting input of the opamp

out: the output of the opamp From the measured output waveform, you can see right away that the 741 (as is well known) is not a rail-to-rail type; the output saturation voltage is around 20% lower than the supply voltage. In addition, the slew rate of $0.2 V/\mu s$ is depicted realistically. Test point 'a' shows the voltage on the 2.4-nF capacitor; it is a slightly bowed triangle waveform with alternating positive and negative slopes. A square wave signal can be seen at test point 'b' (non-inverting input). Due to voltage divider R2/R3, its amplitude is exactly half that of the signal at the 'out' test point. With the aid of this simulation, circuit operation can be explained to relatively inexperienced users. When the voltage on the capacitor rises above or falls below the reference voltage at the node of voltage divider R2/R3, the opamp output changes state (switches from the positive saturation level to the negative saturation level or vice versa). As a result, the capacitor is periodically charged and discharged.

Example 2: Percussion generator

The behaviour of the second circuit (Figure 4) is less obvious, even for experienced designers. The reverse feedback is provided by a network in this circuit. If you view the

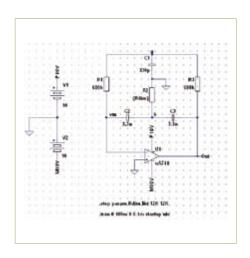


Figure 4. The behaviour of this circuit is not as obvious as that of the first example.

TEST & MEASUREMENT

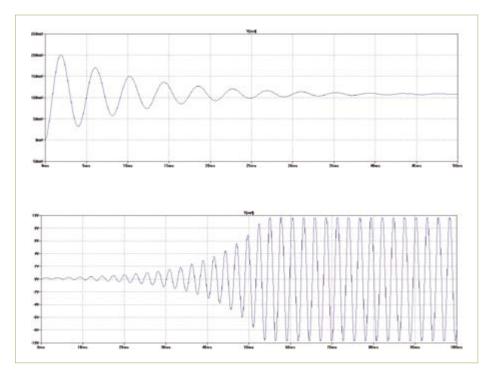


Figure 5. The attenuation of the oscillation waveform depends on the value of resistor R2 (see Figure 4). Values below the critical value result in a waveform with increasing amplitude.

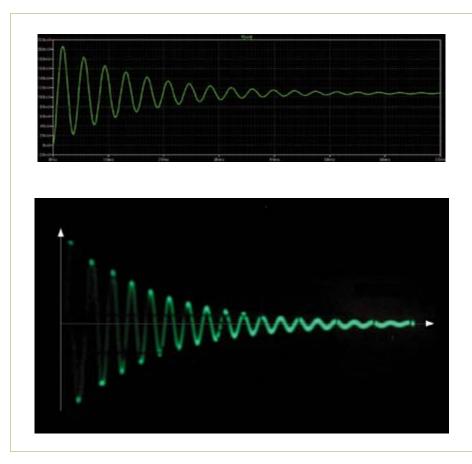


Figure 6. A perfect fit: simulation result versus waveform measured with real hardware.

simulation results, you can clearly see that this network has a rather unusual property. Two simulation results are shown in Figure 5. For the upper curve, the value of resistor R2 was set above the critical value (around 17 k Ω with the indicated component values). This circuit generates an exponentially decaying sine wave signal after the supply voltage is switched on. This also explains the name of the circuit: if the output signal is converted into an acoustic signal, it sounds like a light tap on a drum head. However, if the value of R2 is reduced below the critical value, the circuit produces an exponentially increasing sine wave signal as shown in the lower curve, which is ultimately limited by the maximum output voltage of the opamp.

If the value of R2 could be set to exactly the critical value Rcrit, the circuit would generate a sine wave signal with constant amplitude. However, this cannot be achieved in practice.

Figure 6 shows a comparison between the simulation result and an oscilloscope signal recorded with an actual implementation of this circuit. These images demonstrate the outstanding quality of the simulation software.

To further illustrate the extensive features of the simulation tool, a parametric plot is shown in Figure 7. It represents the output voltage versus the voltage at test point 'b'. You can immediately recognise the phase shift between the two signals and the time decay of the signal amplitudes.

Other useful functions

In addition to transient analysis (time domain analysis), you can generate an AC analysis with LTSpice. This is useful for examining the frequency response of a network.

For this analysis, a tuneable signal generator must be connected to the input of the network. Here again, test points can be defined with the virtual probe. Figure 8 shows the frequency response of the feedback network of the percussion generator circuit. The middle curve illustrates the situation when R2 has the critical value, which yields the narrow impedance notch clearly visible here. The curves to the left and right illustrate the situations for oscillation with decaying or increasing amplitude.

Even more complex circuits can be studied in detail with the aid of a simulation program such as LTSpice. The virtual measurement results are usually astonishingly close to the results measured with the real hardware.

However, you should always be aware of the limitations of computer simulations. Although programs such as LTSpice have reached a high level of maturity, aberrations can always occur in practice. In particular, it takes considerable effort to fully model ambient conditions such as electromagnetic interference, radiated high frequency interference and noise components in simulations.

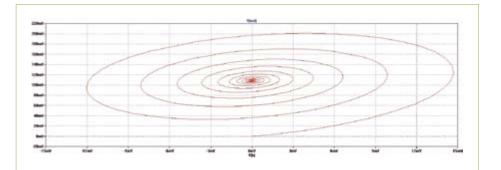
Finally, you should always bear in mind that Murphy's Law also applies to simulations: a circuit that works perfectly as a simulation can always fail in practice.

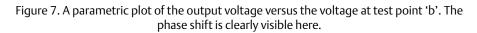
On the other hand, a circuit that doesn't work properly as a simulation is unlikely to prove satisfactory in real hardware.

(081006-I)

Internet Links

[1] www.linear.com/designtools/software/ ltspice.jsp





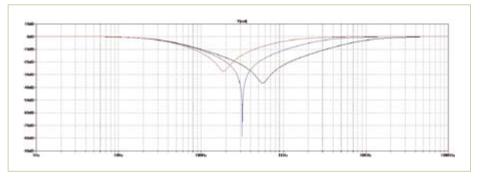


Figure 8. LTSpice can also show the frequency response of a network (in this case the feedback network of the circuit in Figure 4).

[3] www.elektor.com/0602

[4] http://focus.ti.com/docs/prod/folders/ print/ua741.html#technicaldocuments

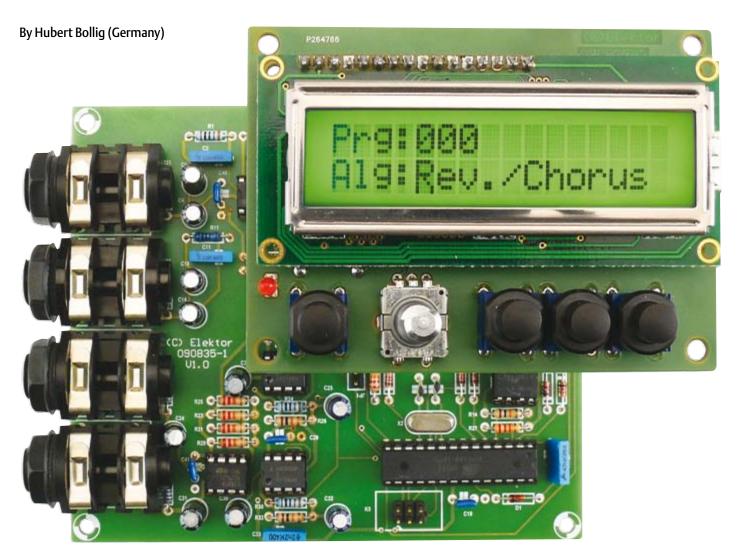


[2] www.elektor.com/060207[3] www.elektor.com/060206



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Digital Multi-Effects Unit With 15 configurable sound effects and a user-friendly HMI



It's a simple fact: every recording sounds better with the right sound effects. Here we prove that it's possible to generate a variety of effects digitally, including reverb, chorus and flanger effects, without having to work yourself to the bone with DSP programming. The circuit is built around a highly integrated effects chip and features an intelligent user interface with an LCD. The result is a treat for the eye and the

ear.

Sound effects are what gives music that special touch. An effects generator is an indispensable element of the kit of every

modern musician, both professional and amateur. In the mid-1980s, the only components available to designers of effects units were bucket-brigade ICs and opamps, but in the course of the digital revolution these components have increasingly been replaced by digital signal processor (DSP) ICs. The growing market presence of these

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highly integrated ICs spelled an end to the DIY era for electronic enthusiasts with an interest in music, as the components and the necessary software simply became too complex. In addition, DSP ICs are not readily available and are almost impossible to solder by hand.

Everything in one IC

Fortunately, this unsatisfactory situation has changed in the last few years. This probably results from the fact that sound effect capability is being built into more and more mixing panels, stomp boxes and other musical equipment. Application-specific ICs such as the FV-1 IC [1] from Spin Semiconductors [2] represent a relatively economical option for equipment manufacturers.

All the components necessary for building an effects unit are already integrated in the IC: high-quality audio A/D and D/A converters, delay RAM (for delaying audio signals), four LFOs (two sine wave and two ramp), and three auxiliary audio inputs. As you can see from the block diagram (Figure 1), these components are grouped around a DSP core with a 24-bit ALU. The DSP does not require user programming, since seven algorithms (for corresponding effects - hall, chorus, flanger, etc.) are stored in ROM by the manufacturer. Another eight effect algorithms can be held in an external EEPROM and input via the I2C port. Users can generate these algorithms themselves with the aid of a special assembler (SpinAsm) [3]. A development environment and various examples can be downloaded free of charge.

Now you may be thinking that such a special IC is virtually impossible to obtain for personal use, but that's not the case. There are three sources that also sell to private parties: the online shops of Profusion PLC in the UK [4], Small Bear Electronics in the USA [5], and Das Musikding in Germany [6].

Instead of spending several months wrestling with DRAM timing, special setup registers and so on, you can obtain the full functionality of an effects unit in the form of a single IC. This IC is intended to be used with a user interface consisting of a rotary switch for selecting the effect algorithm

Features

- Multi-effect generator with 15 effect algorithms
- Based on the FV-1 effect processor IC
- Frequency response 20 Hz to 15 kHz
- Maximum delay time 700 ms
- 64 memory locations for preset configuration profiles
- Separate user interface
- MIDI In port
- Effect selection and parameter control via MIDI port
- Integrated ramp generator for producing attack/decay effects
- Easy assembly using standard components (except the FV-1 IC)
- Free assembler program for generating user-programmed effect algorithms

and three potentiometers for adjusting the effect parameters. However, our aim here is to achieve more than the basic circuit has to offer.

User-friendly operation

We want our effects unit to have the following convenience features:

- LCD display
- an EEPROM for holding eight supplementary effect algorithms (as previously mentioned)
- a ramp generator for attack and decay effects
- 64 preset profiles for user-defined effect settings
- and of course, a MIDI interface for musicians who shudder at the thought of

buttons (which includes the author)

For this reason, in our design a microcontroller looks after the tasks that would otherwise be handled by a rotary switch and a set of three potentiometers. Digital selection of the effect algorithm and generating the analog voltages for configuring the effects are easy tasks for a microcontroller. For this job we chose an Atmel ATmega8. It has an integrated EEPROM, which is ideally suited to storing the preset configuration settings.

The user interface with its LCD module and buttons is housed on a separate PCB, which provides several advantages. If you want to install the effects unit in an existing synthesizer or amplifier, it's easy to fit the user

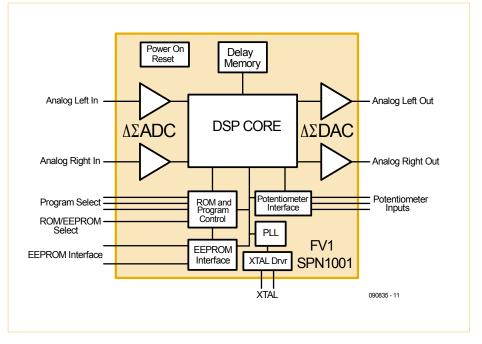


Figure 1. The FV-1 IC is sound effects processor featuring a DSP core, ADC, DAC and additional components. Seven predefined effect algorithms are stored in ROM.

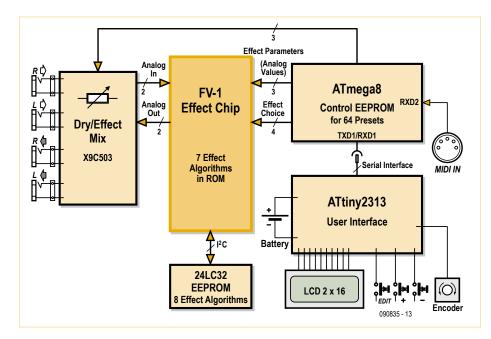


Figure 2. The ATmega microcontroller controls the effects processor IC via four digital lines and three analogue lines. The user interface has its own microcontroller.

interface board in a separate enclosure. The connection in this case is provided by a three-way cable, and there's no need to disfigure the front panel of your sound gear. This approach also simplifies the layout of the main PCB, but the key consideration is that we gave the user interface its own intelligence in the form of an ATtiny microcontroller. Here the author made a virtue out of a necessity: the ATmega did not have

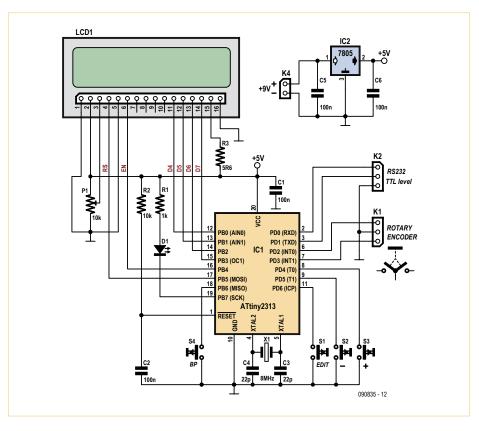


Figure 4. Schematic diagram of the user interface. The LCD module is operated in 4-bit mode, and connector K2 is connected to K3 on the main board.

any I/O pins left for the controls and indicators, so some form of I/O expansion was anyhow necessary. The end result is that the overall circuit offers maximum flexibility for changes and extensions, and if you wish you can also use the user interface for other purposes.

The EEPROM for the additional effects can also be seen in the block diagram of the circuit (Figure 2). The block at the left in this diagram allows the original (dry) signal to be mixed with the signal emerging from the effects processor. In line with the design philosophy of the unit, dry/effect mixing is also controlled by the ATmega with the aid of two digital potentiometers.

The circuit

Armed with the previous description, it's fairly easy to understand the operation of the circuitry on the main board (Figure 3). The audio signals from the input connectors are fed directly to the inputs of the FV-1 IC via coupling capacitors, and to the inputs of the buffer amplifiers for dry/effect mixing (IC1a and IC1b). The outputs of IC1a and IC1b are fed to the inputs of the electronic potentiometers (X9C503). The output signals of the effect processor are fed to the other inputs of the signals from the potentiometers. The signals from the potentiometers pass through buffer amplifiers on their way to the output connectors.

The external circuitry of the FV-1 IC corresponds to the recommend circuit of the manufacturer's data sheet. Capacitors C3 (100 nF) and C17 (10 uF) must be located as close as possible to corresponding pins of the IC. LED D2 is a clipping indicator, which lights up when the internal ADC or DAC is close to saturation or in saturation [1]. It should also be mentioned that the DSP operates with a supply voltage of 3.3 V. The effect algorithm is selected by a 4-bit signal applied to pins T0 and S0-S2. The level on T0 determines whether an internal effect algorithm (hard-coded in ROM) or an external algorithm (stored in the 24LC32) is to be used. Diodes D1 and D5–D7 together with resistors R7 and R19, R20, R21 are simple level converters from 5 V to 3.3 V. Diodes D9 and D10 are reserved for future

extensions.

The three analogue inputs POT0–POT2 can be used to adjust the specific effect parameters of each effect algorithm. The analogue voltages for POT0–POT2 are generated by the ATmega in the form of PWM signals. For this purpose, the 5-V PWM signals are first reduced to 3.3 V by voltage dividers (R9/ R10, etc.) and then converted to DC voltages by low-pass filters.

The effect type and effect parameter settings, along with the dry/effect mix setting, are combined to form a preset profile and stored in the internal EEPROM of the ATmega8. The 512 bytes of EEPROM memory provide exactly enough space for 64 preset profiles. JP2 is provided for write protection (write protection is enabled when pin 7 is tied to VCC and disabled when pin 7 is not tied to VCC).

Effect selection

There are two options for sending commands to the ATmega to select a preset profile:

- The serial port (K3) and associated user interface
- The MIDI port (K7)

•

The implementation of the MIDI port with an optocoupler is standard. The only thing that should be mentioned here is that the optocoupler used for this (a 5N139) is simpler and less expensive than the PC900 stipulated by the MIDI specification. The serial MIDI data is clocked into the INT0 input at 31,250 baud.

The user interface is connected to the other serial port (RxD/TxD). Diodes D3 and D4 allow the main PCB and the user interface PCB to be powered from separate supply voltages. For example, the user interface may operate under battery power. Of course, it is also possible to power the user interface from the main PCB.

The circuit of the user interface is straightforward. The LCD module is operated in 4-bit mode. The input components consist of four buttons and a rotary encoder, which is connected to K1. Pay careful attention to

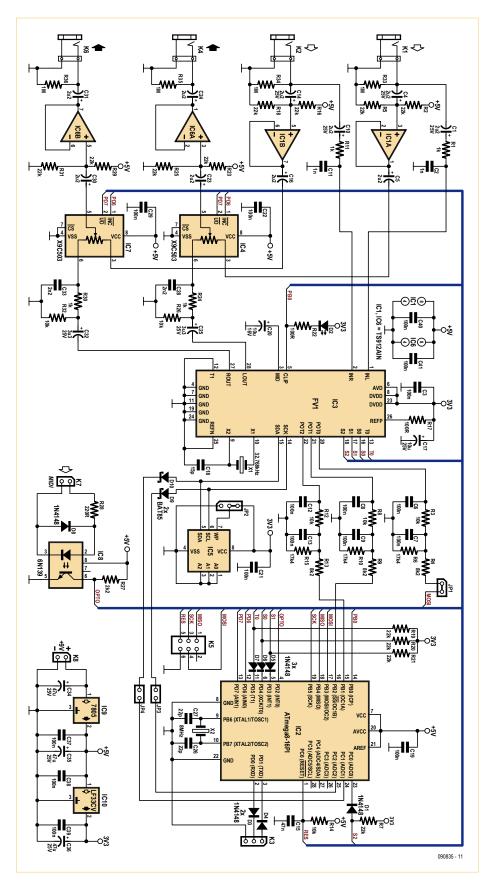


Figure 3. Schematic diagram of the main board, with the buffer amplifiers for the signal inputs and outputs and the digital potentiometers for dry/effect mixing shown on the left. The three resistor/capacitor networks in the middle of the diagram convert 5-V PWM signals into DC control voltages with a range of 0 to 3.3 V.

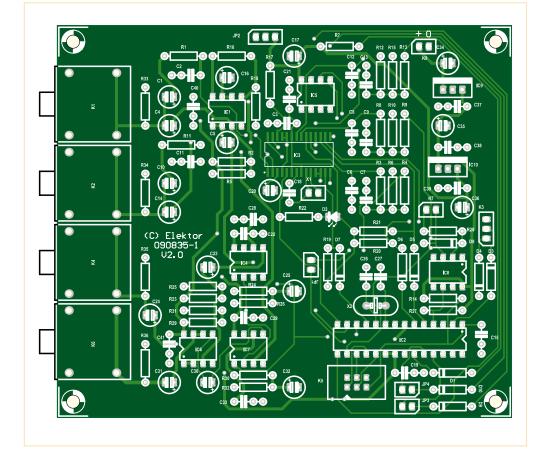


Figure 5. Component layout of the main board. The FV-1 IC is the only SMD component.

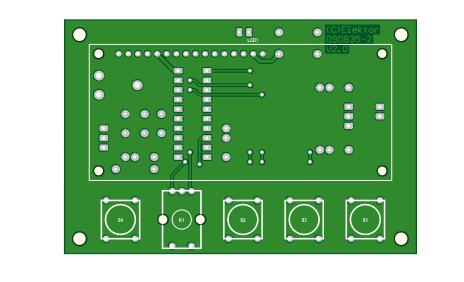


Figure 6. User interface PCB.

COMPONENT LISTS

Main Board

Resistors $R1,R11,R24,R30 = 1k\Omega$ R2,R5,R7,R16,R18,R19,R20,R21,R $23,R25,R29,R31 = 22k\Omega$ R3,R8,R12,R14,R26,R32 = 10kΩ $R4, R9, R13 = 8.2 k\Omega$ $R6,R10,R15 = 17.4k\Omega (1\%)$ 250 mW) R17,R22 = 100Ω R27 = 2.2kΩ R28 = 220Ω $R33, R34, R35, R36 = 1M\Omega$ Capacitors C1,C4,C5,C10,C14,C16,C23,C24,C 30,C31 = 2.2µF 16V C2,C11,C28,C33 = 1nF 400V C3,C6,C7,C8,C9,C12,C13,C19,C21 ,C22,C29,C37,C38,C39,C40,C41 = 100nF, ceramic C17,C20 = 10µF 16V C15 = 47 nFC18 = 1nF, ceramic) C25,C32 = 3.3μF 16V C26,C27 = 22pF C34,C35,C36 = 47µF 16V Semiconductors D1,D3,D4,D5,D6,D7,D8 = 1N4148 D2 = LED, 3mm, low current D9,D10 = BAT85 IC1,IC6 = TS912, dual opamp, rail to rail (ST Microelectronics TS912IN) IC2 = ATmega8-16PU, programmed, Elektor order # 090835-41 [8] IC3 = SPN1001-FV1, Spin Semiconductor IC4,IC7 = X9C503 electronic potentiometer (Xicor X9C503P, Farnell: 179485) IC5 = 24LC32 (Microchip 24LC32A-I/P), programmed, order # 090835-31[8] IC8 = 6N139 Optocoupler (Vishay Semiconductor) IC9 = 7805 IC10 = LF33CV (ST Microelectronics) Miscellaneous X1 = 32.68kHz quartz crystal X2 = 8MHz quartz crystal K1,K2,K4,K6 = 2-way screw terminal block, PCB mount K3 = 3-pin pinheader, lead pitch 0.1 in. K5 = 6-pin (2x3) pinheader, lead pitch 0.1 in. K7,K8 = 2-pin pinheader, lead pitch 0.1 in. |P1,|P3,|P4 = 2-pin pinheader, lead pitch 0.1 in., with jumper IP2 = 3-pin pinheader, lead pitch 0.1 in., with jumper

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PCB # 090835-1, see [8] or

Kit of parts # 090835-71, contains all components including boards and programmed microcontrollers, EEPROM, see [8]

User Interface

Resistors R1 = $1k\Omega$ R2 = $10k\Omega$ R3 = 5.6Ω P1 = $10k\Omega$ trimpot Capacitors C1,C2,C5,C6 = 100nF C3,C4 = 22pF

Semiconductors D1 = LED, 3mm, low current IC1 = ATtiny2313-20PU, programmed, Elektor order # 090835-42 [8] IC2 = 7805

Miscellaneous X1 = 8MHz quartz crystal EC11E15204aE K2 = 3-pin pinheader, lead pitch 0.1 in. K4 = 2-pin pinheader, lead pitch 0.1 in. S1,S2,S3,S4 = pushbutton, Multimec type 3FTL6 LCD1 = LCD 2x16 characters, Displaytech type 162C PCB, order # 090835-2, see [8]) or Kit of parts # 090835-71, contains all components including boards and programmed microcontrollers, EEPROM, see [8]

K1 = Rotary encoder, e.g. Alps type

the required pin assignments when selecting a component for this (check the data sheet).

Assembly and programming

To simplify assembly, a kit with all the necessary components (including the PCBs, pre-programmed microcontrollers and a programmed EEPROM) is available from the Elektor Shop. The PCBs (Figures 5 and 6), the pre-programmed microcontrollers and the programmed EEPROM are also available separately from the Elektor Shop. All components except the FV-1 IC are leaded types. The only SMD is the effects processor, but it is relatively easy to solder by hand since the pin spacing is 1.27 mm. If you cannot find 17.4 k Ω resistors for R6, R10 and R15, you can use 15 k Ω and 2.2 k Ω resistors connected in series.

The source and hex files for programming the microcontrollers and the hex files for programming the EEPROMs (external and internal to the ATmega) are available on the web page for this article [8]. The EEPROM file for the ATmega contains initialisation values for 63 preset profiles. As downloading program code to the flash memory of the ATmega also erases the internal EEPROM, the EEPROM must be programmed last.

The flash memories of the AVR microcontrollers can be programmed with an STK500 development board (among other options), which can also be used for writing data to

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Operation

A total of 15 effect algorithms (effect types) are available. The name of the currently selected effect algorithm is shown on the first display page. Each effect algorithm has three specific effect parameters, which are combined to form a preset profile. The unit can store up to 64 preset profiles in memory.

The dry/effect mix ratio and ramp generator setting are also stored in each preset profile.

The manual included in the download file [8] contains a table of all effect algorithms and their configurable parameters.

After the unit is switched on, profile '000' is loaded and shown in the first line of the display. The currently selected effect algorithm is shown in the second line of the display.

The next or previous preset profile can be selected by pressing the '+' and '-' buttons.

The Bypass (BP) button takes the effect unit out of the audio signal path.

The rotary encoder can be used to adjust the parameter values. The currently selected parameter is indicated on the display by a blinking cursor. Clockwise rotation increases the parameter value, while anticlockwise rotation reduces the value.

Press the Edit button to jump to the next parameter page or select the next parameter. If you have already reached the final parameter, pressing the Edit button takes you back to the first display page.

To save a preset profile, first press and hold the Edit button and them press the '+' button.

If you do not wish to save a setting, simply press the '+' or '-' button. Any changes you have made to the profile up to this point will be lost.

The downloadable manual also includes information on controlling the unit via the MIDI interface.



the internal EEPROM. An ISP port (K5) is also provided for the ATmega to enable in-circuit programming and downloading of firmware updates. Jumper JP1 must be removed in order to program the microcontroller via the ISP port.

You need an EEPROM programmer for the external EEPROM, but you can also manage with a bit of supplementary hardware and the PonyProg program [9]. The Pony-Prog website has a design for a simple parallel port to I2C adapter circuit, which can easily be assembled on a piece of prototyping board. Incidentally, this handy program is also suitable for programming the AVR microcontrollers.

The zip file, which (as usual with Elektor projects) can be downloaded free of charge, also includes a Read Me file with information to help you sort out the hex files, as well as several screen shots showing the correct fuse settings.

Initial operation

As usual, you should power up the circuit

for the first time with the IC sockets empty. Check the supply voltages (5 V and 3.3 V) at the IC sockets. Next, fit the ICs in their sockets and connect the user interface to the main board. The welcome message should appear on the LCD module after power is switched on. Press the Edit button to navigate to the settings screen for the first parameter. Measure the voltage on the POT0 terminal. It should be possible to adjust it over the range of 0 to 3.3 V with the rotary encoder. It should never exceed 3.5 V under any conditions.

The Clip LED (D2) should light up briefly when power is switched on. The circuit is now ready for use, and you can connect a line level signal to its input.

The operating instructions are summarised in the inset. A detailed user guide (in German) is included in the download file [8].

Internet Links

- [1] www.spinsemi.com/Products/ datasheets/spn1001/FV-1.pdf
- [2] www.spinsemi.com
- [3] www.spinsemi.com/Products/ datasheets/spn1001-dev/ SPINAsmUserManual.pdf
- [4] www.profusionplc.com
- [5] www.smallbearelec.com/home.html
- [6] www.musikding.de/
- [7] http://ww1.microchip.com/downloads/ en/DeviceDoc/21713g.pdf
- [8] http://www.elektor.com/090835
- [9] www.lancos.com/prog.html

(090835-I)

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The Elektor DSP Radio (2) Antennas and PC software

By Burkhard Kainka (Germany)

An important advantage of the Elektor DSP radio over other world receivers is the flexibility the user has to change its characteristics, both by modifying the antenna and front-end circuit configuration and by modifying the software. The unit can be used in its various configurations as a

۲ -81 (c) Elektor 100126-1 VI.2

PC-controlled radio or as a portable world receiver.

The circuit diagrams shown here illustrate the input circuit of the receiver we published in the first article in this series. For each possible antenna configuration the relevant jumper settings appear in red and the connected antennas and their accompanving coils appear in green.

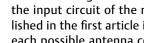
Figure 1 shows the standard configuration of the front-end circuit, with all jumpers in their default positions. Switching diodes D6 and D7 are represented symbolically as shortwave and mediumwave switches. You can see from the diagram how in each case a part of the inductance in the circuit is shorted. A common antenna is used for all frequency ranges: one example of a suitable design is a 50 cm whip. Another possibility is to use an outdoor antenna: the author has successfully used a 10 m longwire antenna connected via 30 m of coaxial cable. Although this antenna is not ideal for FM reception, it nevertheless gave good results over all frequency bands from longwave to FM.

If separate antennas are to be used for FM and AM reception, IP1 can be fitted to take the bottom of FM coil L2 to around. The open pin at IP1 (connected to L3 and C15) then forms the new AM antenna input for long-, medium- and shortwave (Figure 2). For indoor operation on long- and mediumwave ferrite antennas are often better than wire antennas as they have greater immunity to the types of electric field interference often found inside houses. Figure 3 and Figure 4 show two alternative possibilities. In the first example the coils on the ferrite rod form an additional part of the receiver circuit, while in the second example they replace the fixed inductors on the receiver board, which are taken out of circuit by removing jumpers JP2, JP3 and JP4.

The suggested turn counts given are suitable for use with a 10 mm diameter ferrite rod with a length of between 9 cm and 15 cm. The automatic tuning function of the receiver means that the exact inductance value is not critical. Experiments show that the ferrite rod provides good signal amplitude and interference rejection at frequencies of up to 10 MHz, and this is why both circuit variations also include a shortwave coil on the ferrite rod.

A further interesting option is the tuned loop antenna shown in Figure 5. A wire loop with a total length of 4 m takes the place of JP2. It is possible to make the loop smaller, and the wire length and shape can also be varied. Again, because of the automatic tuning, high signal levels can be achieved, approaching the performance possible with longwire antennas. The radio's display will show the capacitance in the tuned circuit:

RF



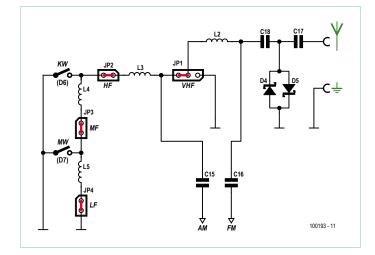


Figure 1. Standard configuration of the front-end circuit.

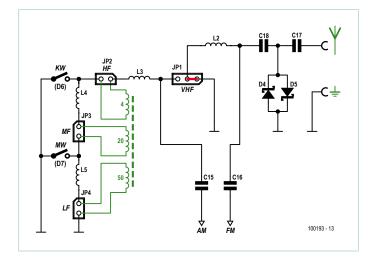


Figure 3. Connecting a ferrite rod antenna for use up to around 10 MHz.

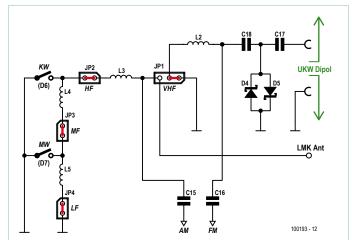


Figure 2. Separate connection of FM and AM antennas.

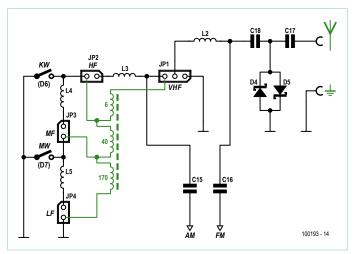


Figure 4. In this variation the coils on the ferrite rod antenna replace the inductors on the receiver board.

as long as this remains between 10 pF and 500 pF the antenna will operate optimally. It is worth noting that the same antenna can be pressed into service for medium- and longwave reception, in each case forming a part of the total inductance in the tuned circuit.

Finally, Figure 6 shows a possibility for shortwave DXing. Jumpers JP2 to JP4 are not fitted and instead we fit a small transformer at the AM input. The turn counts given are intended for use with a 5 mm former with a tuning slug suitable for operation up to 15 MHz. The same coil was used in the 'Preselector for Elektor SDR' article in the December 2009 issue of Elektor and is available from Conrad Electronics (order code 516651) or from ModulBus (RF coil kit T1.4) [1].

The input coil replaces the inductor in the tuned circuit on the receiver board, and

opens up the possibility of changing the turn count on the antenna side to optimise matching to the antenna. For example, a small coupling winding as shown, consisting of just one turn, will give good matching to a 50 Ω antenna.

PC control

PC control opens up a whole new dimension of possibilities to the Elektor DSP radio beyond its use as a portable receiver. The radio can be controlled over a USB interface using a specially-developed program called ElektorDSP1, written in Visual Basic. The program, including source code, is available for free download from the Elektor web pages accompanying this article [2].

The first step on starting the program is to establish which COM port is to be used: the default is COM1. If, for example, COM4 is to be used, edit the 'COM' text box in the user interface (Figure 7) appropriately and click on 'Open'. When the program is closed the COM port setting is saved in the file COM. ini, and the value will be restored when the program is next started. All frequencies and station names are stored in the file DSPfreq. ini, and these are also loaded when the program starts.

The program allows the user to choose a frequency from 20 presets for the FM band and a further 20 presets for the AM bands. The frequencies can be edited individually. For each preset a station name can be given for reference: in the case of an FM preset this string will be shown on the PC's screen, but the display on the receiver will show the station name as transmitted by RDS. In the case of AM presets the station names are sent to the receiver along with the frequency data, and the receiver will display this string when the preset is selected

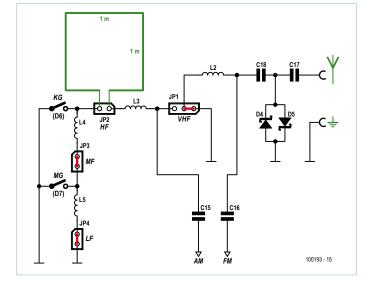


Figure 5. Connecting a loop antenna for AM.

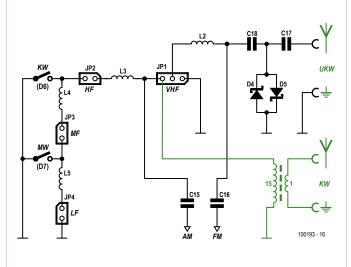


Figure 6. Antenna coupling using an input coil for DX reception.

(by pressing S5 for longer than 0.5 s). This makes tuning to AM presets a little more user-friendly.

The settings can be stored in the receiver as power-up defaults by pressing S5 for more than two seconds. All settings, including any changes to the various parameters described below, will then be copied into EEPROM. To erase all these stored settings, hold down S5 when turning the receiver on: everything will then be reset to standard values.

The Si4735 device in the DSP radio can be configured in a wide variety of ways that determine its reception characteristics. At power-up all these parameters are set to

Table 1. FM properties				
Property Valid range		Effect		
De-emphasis	0 to 2	Treble cut		
Stereo	0 to 80 dBµV	0 to 80 dBµV Stereo mode above this signal strength		
Mono	0 to 80 dBµV	Mono mode below this signal strength		
Mute rate	0 to 100 dB/s	0 dB/s Soft mute rate of volume change		
Mute max	0 to 32 dB	D to 32 dB Soft mute maximum attenuation		
Mute SNR	ute SNR 0 to 25 dB Soft mute SNR threshold			
Seek SNR	0 to 20 dB	Automatic search SNR threshold		
Seek RSSI	RSSI 0 to 60 dBµV Automatic search signal strength threshold			

Table 2. AM properties			
Property Valid range		Effect	
De-emphasis	0 to 1	Treble cut	
Filter	1 to 6 kHz	Filter bandwidth	
Mute rate	0 to 100 dB/s	Soft mute rate of volume change	
Mute slope	0 to 100 dB/s	Soft mute attenuation function slope	
Mute max	0 to 32 dB	Soft mute maximum attenuation	
Mute SNR	0 to 25 dB Soft mute SNR threshold		
Seek SNR	0 to 20 dB	Automatic search SNR threshold	
Seek RSSI	0 to $60 \text{ dB}\mu\text{V}$	V Automatic search signal strength threshold	

sensible default values, but it is possible to alter their values to suit particular reception requirements or conditions. This possibility was kept in mind when developing the embedded software for the radio, and it is possible to control eight important parameters for AM and FM reception from the PC. If desired, it is possible to set up the receiver with an entirely fresh batch of settings and then save them for when the radio is used as a portable receiver.

FM parameters

In the interests of reducing noise, FM transmitters add 'emphasis' (boost of high frequencies) to the signal. This must be compensated for in the receiver ('de-emphasis'). The Si4735 features a switchable de-emphasis filter.

When signal strength is low many FM receivers produce a poor stereo output with considerable interference; some radios automatically switch to mono mode in these conditions. The DSP radio offers a better solution, smoothly and almost imperceptibly transitioning between stereo and mono modes. The upper and lower signal strength limits for the transition are adjustable.

In very weak signal conditions the DSP radio chip uses a 'soft mute' function rather than the sudden muting provided in some receivers, gently attenuating the output and thus also the noise. The parameters controlling this are the 'mute rate' (speed of volume change), the maximum attenuation ('mute max') and the SNR (signal-to-noise ratio) threshold below which muting is triggered. It is worth experimenting with these parameters, especially when tuning in to weak stations.

The 'seek SNR' and 'seek RSSI' parameters affect the behaviour of the automatic station search function. Stations will only be found if they exceed the specified signal strength (RSSI) and SNR thresholds.

Table 1 gives an overview of the FM parameters.

AM parameters

The parameters governing AM reception (Table 2) are similar to those for FM. An extra is that the bandwidth of the receiver is adjustable to one of a number of predefined values. The parameter can take on the value 0 (6 kHz), 1 (4 kHz), 2 (3 kHz), 3 (2 kHz) or 4 (1 kHz). Here '2 kHz' (the default value) means that the will accept signals out to 2 kHz away from the tuned frequency on either side, and thus corresponds to an actual IF filter bandwidth of 4 kHz. For strong stations, using a wider bandwidth can improve audio quality, while for DX reception a narrower bandwidth will help reduce interference. The deemphasis control has a similar effect, and can be used as a crude treble control.

In AM mode there are four parameters that control the soft mute function. The highly effective automatic level control in the receiver IC increases gain when the signal level falls, which as a consequence also increases noise. If the input level falls below a preset threshold, however, the volume will be reduced in proportion. The signal level threshold, the slope of the muting function, the speed with which the function acts and the maximum degree of muting are all adjustable. The default values are designed to work well when

Listing 1

Low-level subroutines

Declare	Sub	Get_int_status()
Declare	Sub	Rx_volume()
Declare	Sub	Power_down()
Declare	Sub	Init_am()
Declare	Sub	<pre>Init_fm()</pre>
Declare	Sub	Am_tune_freq()
Declare	Sub	Fm_tune_freq()
Declare	Sub	Fm_seek_freq_up()
Declare	Sub	Fm_seek_freq_down()
Declare	Sub	Am_seek_freq_up()
Declare	Sub	Am_seek_freq_down()
Declare	Sub	Fm_tune_status()
Declare	Sub	Fm_rsq_status()
Declare	Sub	Am_tune_status()
Declare	Sub	Am_tune_status_stop()
Declare	Sub	Am_rsq_status()
Declare	Sub	Fm_start()
Declare	Sub	Am_start()
Declare	Sub	Am_seek_step_9khz()
Declare	Sub	Am_seek_step_5khz()
Declare	Sub	Am_seek_step_1khz()
Declare	Sub	<pre>Init_rds()</pre>
Declare	Sub	Fm_rds_status()
Declare	Sub	Rds()

	Vol	4		1	•	50 dB
FM			AM			
1 88300	kHz	DKULTIR	1	153	kHz	DLF
2 88800	kHz	WDR 5	2	549	kHz	DLF
3 95100	kHz	WDR 3		648	kHz	BBCWS
4 96500	kHz	0F0S	4	720	kHz	WDR 2
5 99200	kHz	WDR 2		5955	kHz	RNW
6 101300	kHz	WDB 4		6075	kHz	DW
7 102200	kHz	Essen	7		kHz	
8 102800 9 102300	kHz	DLF	9	_	kHz	
	kHz kHz	Funkhous	9	_	kHz kHz	
10,0000	kHz	Essen	11		kHz	
11 106700	kHz	ILIVE	12		kHz	
13	kHz		13		kHz	
14	kHz		14		kHz	
15	kHz		15		kHz	
16	kHz		16		kHz	
17	kHz		17		kHz	
18	kHz		10		kHz	
19	kHz		19	_	kHz	
20	kHz		20		kHz	
Memory	,			Memor	7	
Deemphasis	-	¥ 2	Deer	nphasis	4	3
Stereo	1	▶ 49 d8wV	Filter		1	> 2 kHz
Mono		> 30 dbyV	Hute	Rate		► 64 d8/s
Mute Rate	1	> 2 d8/s	Hute	Slope	1	2 d8/d8
Mute Max	1	▶ 16 d8	Hute	Max		> 16 d8
Mute SNR	1	▶ 4 dB		SNR	-	▶ 5 dB
Seek SNR	1	9 g 0		SNR	1	▶ 5 dB
Seek RSSI	1	▶ 20 dBuV	See	RSSI	1	▶ 25 dBull

Figure 7. The user interface presented by the ElektorDSP1 program.

Listing 2

AM tuning and band switching

```
Sub Am_tune_freq()
  If Fam > 500 Then
    If Fam > 2000 Then
       Portb.0 = 1
                        \SW
       Portc.3 = 0
    Else
       Portb.0 = 0
                         ۱MW
       Portc.3 = 1
    End If
  Else
       Portb.0 = 0
                         ١LW
       Portc.3 = 0
  End If
  I2cstart
  I2cwbyte 34
  I2cwbyte &H40
  I2cwbyte &H00
  H = High(fam)
  L = Low(fam)
  I2cwbyte H
  I2cwbyte L
  I2cwbyte &H00
  I2cstop
End Sub
```

Listing 3

```
Decoding serial commands
```

```
baud = 38400
D = Inkey(#1)
   If D = 102 Then F_control
                                     `f, Freq
   If D = 109 Then Mam_control
                                     'm, Memory AM
   If D = 110 Then Mfm control
                                     'n, Memory FM
   If D = 112 Then Properties
                                     'p, Property
   If D = 105 Then Pc control i2c
                                     'i, I2C command
   If D = 106 Then Rdsout = 1
                                     'j: RDS output
   If D = 107 Then Rdsout = 0
                                     'k: RDS output off
   If D = 114 Then Print Rssi
                                     'r: RSSI
   If D = 115 Then Print Snr
                                     's: SNR
```

receiving strong stations. The volume is attenuated by a maximum of 16 dB when the input signal level falls below 10 dB μ V. DXers will likely not find this behaviour ideal as weak stations will fluctuate unnecessarily in volume. So, decide whether the primary purpose of the receiver is to listen to stronger stations or DX reception, and set the parameters accordingly. Many DXers will want to disable the soft mute function entirely: the simplest way to do this is to set 'mute max' to zero.

Listing 5 **Adjustable properties** Sub Properties Print "Property" Input D If D = 1 Then Prop = &H1100 If D = 2 Then Prop = &H1105 If D = 3 Then Prop = &H1106 If D = 4 Then Prop = &H1300 If D = 5 Then Prop = &H1302 'FM SOFT MUTE MAX ATTENUATION If D = 6 Then Prop = &H1303 'FM SOFT MUTE SNR THRESHOLD If D = 7 Then Prop = &H1403 'FM SEEK TUNE SNR THRESHOLD If D = 8 Then Prop = &H1404 'FM SEEK TUNE RSSI THRESHOLD If D = 9 Then Prop = &H3100 If D = 10 Then Prop = &H3102 If D = 11 Then Prop = &H3300 If D = 12 Then Prop = &H3301 If D = 13 Then Prop = &H3302 `AM_SOFT_MUTE_MAX_ATTENUATION If D = 14 Then Prop = &H3303 AM SOFT MUTE_SNR_THRESHOLD If D = 15 Then Prop = &H3403 If N = 16 Then Prop = &H3404If N = 0 Then Prop = &H4000 Input Dat Property End Sub

`FM `FM	_DEEMPHAS _BLEND_ST _BLEND_MC _SOFT_MUT	TEREO_T DNO_THR	
`AM `AM	_DEEMPHAS _CHANNEL_ _SOFT_MUT _SOFT_MUT	FILTER E_RATE	
'AM	_SEEK_SNF _SEEK_RSS lume	—	

Program-it-yourself

The firmware in the DSP radio was developed using Bascom, and the source code is available, along with a hex file, for free download at [3]. The DSP radio printed circuit board has an in-system programming connector for the ATmega168, which allows you to modify the firmware in the radio to your heart's content. If low-level programming is not your cup of tea, the ElektorDSP1 PC control software described above allows you to customise the receiver without writing a line of code. Furthermore, it is possible to use the USB interface to communicate directly with the firmware and alter the receiver's behaviour. In most cases a simple terminal program is all you need, although it is of course possible to develop programs on the PC to control the radio in specific ways.

The Bascom software runs to over a thousand lines of source code, too much to describe in detail here. However, the program includes a large number of readymade subroutines with self-explanatory names that control some of the basic functions listed in the Si4735 datasheet (see Listing 1).

The subroutine Am_tune_freq (Listing 2) is worth a closer look. It shows how a command is typically constructed for transmission to the Si4735 over the I2C bus: the device responds to address 34. In this instance the command code is &H40. After that come four bytes of parameters, including the frequency in kilohertz as a more-significant and a less-significant byte. The AM tuning command also selects the required band via the switching diodes connected to ports B.0 and C.3.

Another aspect of the software worth a look from the point of view of developing special software for manipulating the receiver's parameters is the way the serial port is handled. The interface runs at 38.4 kbaud and appears to the PC as a virtual COM port (see Listing 3). Each command is headed by a single character. For example, a lowercase 'f' results in the subroutine F_control being called, which sets a new frequency. Commands 'm' and 'n' allow the preset

Advertisement

memory to be programmed. The 'p' command gives access to the various 'properties' or 'parameters' of the DSP IC and the 'i' command gives direct low-level access to the Si4735 via the subroutine Pc_control_ i2c. Read commands are also available. For example, it is possible to read the current signal strength or signal-to-noise ratio, and to gain access to the RDS data.

Listing 4 shows how a frequency is set. A frequency in the AM band is set by a command such as 'f5955<Enter>': you can try this out using a simple terminal program. In this example, the receiver is tuned to 5955 kHz. FM band frequencies are also given in kilohertz, such as 'f102800<Enter>' for 102.8 MHz. The firmware carries out

appropriate initialisations when switching between FM and AM modes. After a frequency command the unit returns to normal manual operation mode, and so PC commands and manual commands can be executed alternately. The source code of the ElektorDSP1 program gives an example of how to control the receiver in this way using Visual Basic.

Listing 5 shows the range of functions available via the 'p' command. A total of sixteen different receiver properties can be adjusted. For example, to increase the receiver bandwidth, use a terminal to send the sequence 'p10<Enter>2<Enter>'. To disable the soft mute function, adjust property number 13 (AM_SOFT_MUTE_MAX_ ATTENUATION) by sending the sequence 'p13<Enter>0<Enter>'. Even the overall volume can be adjusted in this way, using property number 0. For maximum volume send 'p0<Enter>63<Enter>'.

Listing 6 shows how to access the low-level features of the Si4735. Each command is introduced by the lower-case letter 'i'. For I2C data transfer the next character (which specifies the subcommand) should be 'C'. The following bytes are transferred as raw binary values. The first two bytes are counts of the number of bytes to be sent and received respectively, and the following bytes are then sent verbatim in the natural order. The microcontroller takes care of prefixing the bytes on the I2C bus with the





RF

Listing 4

Tuning over the serial port

```
Sub F control()
  Print "Tune"
  Input Fin
  If Am = 1 Then
    If Fin >= 60000 Then
      Power down
      Waitms 10
      Am = 0
      Fm start
      Waitms 100
    End If
  End If
  If Am = 0 Then
    If Fin < 60000 Then
      Power_down
      Waitms 10
      Am = 1
      Am start
      Waitms 100
    End If
  End If
  If Am = 1 Then
    Fam = Fin
    Am tune freq
    Waitms 250
    Am tune status
  End If
  If Am = 0 Then
    Fin = Fin / 10
    Ffm = Fin
    Fm tune freq
    Waitms 250
    Fm_tune_status
  End If
End Sub
```

Listing 6

Direct I₂C bus access

```
`*** PC I2C commands ****
Sub Pc_control_i2c
  Print #1 , "I2C"
 Do
    Get #1 , Command
                      'I2C write and read to Si4735
                                ייר יי
    If Command = 67 Then
       Get #1 , Bytesout
       Get #1 , Bytesin
       For N = 1 To Bytesout
         Get #1 , Di2c(n)
       Next N
       I2cstart
       I2cwbyte 34
       For N = 1 To Bytesout
          I2cwbyte Di2c(n)
       Next N
       I2cstop
       If Bytesin > 0 Then
         I2cstart
         I2cwbyte 35
         While Bytesin > 1
           Bytesin = Bytesin - 1
           I2crbyte D , Ack
           Put #1 , D
         Wend
         I2crbyte D , Nack
         Put #1 , D
         I2cstop
       End If
    End If
    If Command = 65 Then
                          'LCD Line 1
      Input #1 , Text1
      Locate 1 , 1
      Lcd Text1
    End If
    If Command = 66 Then 'LCD Line 2
      Input #1 , Text1
      Locate 2 , 1
      Lcd Text1
    End If
 Loop
End Sub
```

start command and the device address. The receiver IC is then addressed again in read mode. The requested number of bytes is then read and passed back to the host PC. The 'A' and 'B' subcommands of the 'i' command are used to send text to be shown on the receiver's LCD panel: the two commands allow the two lines of the display to be written independently.

The receiver does not automatically return to manual mode after the 'i' command is completed; instead, the device remains under PC control. In summary, the Elektor DSP radio offers a wide range of options, well beyond those offered by a conventional world receiver. In particular, it offers an unprecedented array of customizing possibilities. We hope that many readers will build the receiver and exchange their experiences on the forum on the Elektor website, and we feel sure that you will come up with ideas far in advance of anything we have thought of!

(100193)

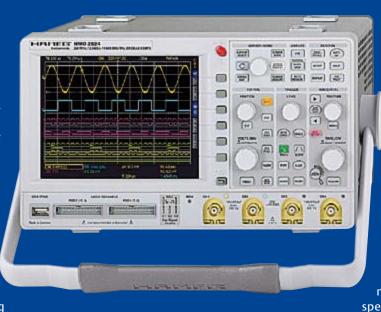
Internet Links

- www.ak-modul-bus.de
 (site only available in German: search for 'HF Spulen Bausatz T1.4')
- [2] www.elektor.com/100193(web pages accompanying this article)
- [3] www.elektor.com/100126 (first article in this series)

Examined: Hameg HMO2524

By Harry Baggen and Ton Giesberts (Elektor Netherlands Editorial / Elektor Labs)

To an electronics designer an oscilloscope is something like an extension of his brain. As soon as any trifling bit of hardware has been slapped it needs to be checked. An oscilloscope is then an indispensable tool (in addition to the equally indispensable multimeter, of course). You use it to 'just' check whether and oscillator is working



was left in the hands of the design team.

The name Hameg conjures up fond memories for many electronics engineers. This German company was one of the first companies to make affordable and usable oscilloscopes available, long before China and other Asian countries started to compete with their products. The prices were such that even for hobby use it became interesting to purchase such a scope. In the meantime the company has specialised in products with

properly or whether there is a signal at a certain place in the circuit. Alternatively you can use it to make more accurate measurements of more complex waveforms. Nowadays, all kinds of

a very good price/quality ratio and a high reliability. That the products made by Hameg have a good reputation is also shown by the fact that the renowned test equipment manufacturer

a scope that can handle both analog and digital signals

bus signals from microcontroller circuits are added to that too. You could use a special logic analyser for this, or a scope which can cope with both analogue and digital signals.

Some time ago, Hameg made an offer to Elektor, for us to try

Rohde & Schwarz bought the company in 2005 to expand their product range at the bottom end. And we probably won't have to tell Elektor readers about the quality (and also the price) of R&S gear, these are at the top of the range!

out one of their new oscilloscopes for a while. Now, the Elektor lab is always keen to do that. Especially if it concerns one of the more expensive products produced by Hameg. After making an appointment with Mr. Grimm, he arrived a few weeks later at Elektor House with a brand new scope and the necessary accessories. He gave an elaborate demonstration of the capabilities of this scope, after which it



In recent years Hameg has expanded their product range at the top end with more expensive and more powerful models. One of the reasons is a better fit with the products that its big brother R&S has to offer. And it is one of these models that our designers were allowed to test drive for a while: the HMO 2524. This is a powerful, digital, 4-channel scope with an analogue bandwidth of 250 MHz and a real-time sampling-

E-LABS INSIDE

rate of 2.5 GSamples/s (when operating in 2-channel mode). The price of the base version of this scope is around 3300 Euros (excl. VAT). The scope can also be used as a 16-channel logic analyser. Additional software options are available to test buses with protocols such as I2C, SPI and RS-232. Of course, for our practical test we had several 'pods' (adaptors to make the digital connections) available to us and the required software options were also activated in the scope. Various types of probes were obviously present as well and we also had an active probe with an input capacitance of only 0.9 pF (suitable for measurements up to 1 GHz).

The first thing you notice when switching the scope on is the bright and clear display, which has a resolution of 640 x 480 pixels and an LED backlight, it is a pleasure to look at! In the lab we have several digital 'all-round' scopes in the price range from 1000 to 3000 Euros and none of those comes even close to competing with this.

The number of options is overwhelming, you will have to invest guite a bit of time before becoming familiar with all the niceties. However, the user-friendliness does suffer a little as a result, because of all these options and with only occasional use, even the most simple settings can be a chore to find in the menus. During our first measuring attempts, Ton, who has more than 20 years experience in the lab and has worked with many different scopes, did need quite some time to dig through a number of menus to finally find the appropriate settings to give a clear test signal on the screen. The quality of the displayed signal, after some effort, is outstanding however. Moreover, the scope reacts quickly to changes in settings, something that is regrettably not the case with many digital scopes. But a scope with this many options forces you to have a good read of the user manual (something that our designers — just like other users of complex equipment — do only rarely and then reluctantly). So the number of features is extremely large. There are numerous trigger modes, there is a sizeable buffer of 4 MSamples where you can search for signals in various ways, there is a very usable FFT-function and there are many cursor measurements as well as mathematical functions that you can program yourself.

What's more, we haven't even mentioned the digital signal measurements yet. This gives you the ability of measuring 16 digital signals simultaneously, at a rate of up to 1.25 GSamples/s, also with very extensive trigger options. This does, however, require special pods to connect the digital signals under test to the connections on the scope. Depending in the type of bus you want to look at you may have to install additional software (available for I2C, SPI, UART/RS232).

With this many signals displayed on the screen at the same time, it can become very cluttered. To keep everything wellorganised, Hameg have made a very handy scroll function that allows you to move through a virtual screen which is a lot bigger than the real screen. We like it a lot!

This new series of oscilloscopes from Hameg, which are, considering the price, above their existing range of analogue/digital equipment, can easily compete on capabilities and features with comparable products from other scope manufacturers such as Agilent, LeCroy and Tektronix. The HMO2524 is an excellent and reliable test instrument for both analogue and digital measurements. But if you are considering purchasing such an instrument you should have the applications to put its many abilities to good use and you will have to invest the time to become proficient with all its features (but this is equally true for any scope in this price range).

Unfortunately the scope had to go back to Hameg. We will especially miss that crisp and bright screen. Is it really not possible to swap it for a few of those other oscilloscopes that we have sitting around here?

Internet Link: www.hameg.com

COS

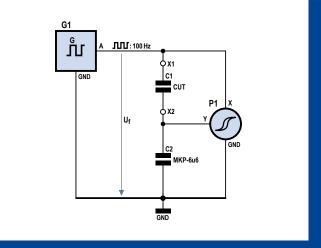
(100420-l)

Know what you're measuring

By Thijs Beckers (Elektor Netherlands Editorial)

At our editorial offices we receive quite a few submissions describing ideas and designs for electronic circuits from readers and authors. All submissions are looked at and receive a cursory assessment as to their usefulness and correct operation, as part of an initial sorting procedure, before putting some serious effort into it. This also applied to a submission for sorting the best (audio) capacitor from a collection of capacitors. At first glance the described method showed much promise: without special (expensive) test equipment it would be possible to compare the quality of capacitors with each other and select the better one(s). This looked like a very interesting design to us. The measuring method is described below. The figures show the test arrangement. The top circuit is used for testing bipolar capacitors. The bottom circuit provides polarised electrolytic capacitors with a bias voltage for testing those types. In each case the bottom capacitor (C2 and C4 respectively) is the 'reference capacitor' and the top capacitor (C1 and C3 respectively) is the 'CUT' (Capacitor Under Test). The idea behind measuring with a bias voltage is that a polarised electrolytic capacitor will distort (even) more when the voltage across the capacitor is allowed to become negative, which is potentially the case with an audio signal. By giving the electrolytic capacitor a sufficiently high bias voltage, a negative voltage across its terminals can be prevented.

The oscillator generates a square wave, which is then divided across both capacitors. The voltage across both capacitors is then visualised using an oscilloscope. By operating the oscilloscope in X-Y mode, specific figures are created which depend on the differences between the capacitors. A 'perfect' diagonal line means that both capacitors are exactly identical, meaning the CUT is of the same quality as the 'reference capacitor'. Any deviation from the diagonal, such as hysteresis, asymmetry or non-linearity means that the CUT is different from the reference capacitor. These differences could conceivably be a larger series resistance, dielectric absorption and other non-linearities. Within this context, it's obvious that the reference capacitor



And because we were also unable to find a way, without using expensive equipment, to establish whether the difference was caused be a difference in capacitance, we regrettably had to reject this article proposal. The problem with this circuit is that you do not really know what you are measuring exactly. That makes the proposed method unsuitable for detecting quality differences between various types of capacitor.

it's not possible to establish the quality of a capacitor with 'cheap' equipment

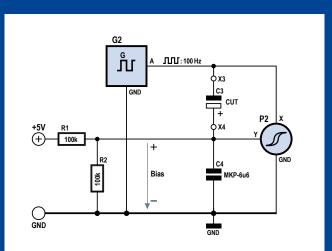
needs to be a specimen of high quality.

During our measurements in our test set-up the idea looked promising. It was indeed possible to clearly see the differences between different types of capacitors. There was an obvious non-linear distortion of the diagonal when a tantalum capacitor was compared to an MKT type, and the difference between an MKT and MKP device was clearly smaller than tantalum MKT, but still clearly visible.

What we also experienced during these measurements, is that a difference in capacitance also resulted in an alleged (quality) deviation. When comparing a 4.7 uF Wima MKP-capacitor (the reference) with a 4.7 uF MKT capacitor the difference became smaller (the 'eye' of the diagonal became narrower) when a 100 nF capacitor was connected in parallel with the MKT device... even when an electrolytic capacitor of 1 uF was connected in parallel with the MKT, the difference from the reference capacitor became smaller. And when making the measurement using a square wave at high frequency (several hundred kHz) very strange things started happening on the scope screen; at both ends of the diagonal there were strange peaks which could not be explained by any differences between the capacitors and which are potentially caused by parasitic inductances in the test set-up.

A (small) difference in capacitance results in such a large deviation that we were unable to establish whether we were dealing with a bad capacitor or just a difference in capacitance. Unfortunately we have to conclude that for the time being it is not possible to establish the quality of a capacitor with 'cheap' equipment. But we fortunately still have our ears, which we can use to test the capacitors in the signal path for what is still the most important characteristic in audio circuits: the sound quality.

(100482)





Tiny samples, big footprint

We are slowly becoming more aware of our impact on the environmental and carbon footprint but something happened in the lab the other week that made us smile: for an upcoming project one of our team, Clemens Valens ordered four IC samples from the chip manufacturer Analog Devices.

One week later the editor of our French edition of Elektor took delivery of a surprisingly large (at least 30 litre) parcel. The complete parcel weighed-in at 1.3 kg and on opening he found, surrounded by packing material, two more boxes in which were nestled some 50 cm long plastic tubes containing the miniscule samples. The total volume of the chips themselves can hardly be more than 1 cubic centimetre!

From the labels on the box the package began its journey to France in the Philippines. Needless to say, Clemens was relieved he wasn't paying for the samples or the post and packing on that one!

(100240)





And dropping by at Elektor...

Our position in the centre of Holland near the German border and close to a motorway means that we are often pleased to welcome guests from Germany. Recently Harry Sievers, Franz Ruffer and Patrick Schmidt dropped in. They represent the distributor CS&S electronic (www.csselectronic.com). Their company carries high-tech Korean products including boards from TeleChips using ARM-compatible controllers with multimedia-capable chipsets. They are intended for mass market applications such as portable devices and communications equipment. These powerful boards look interesting but the Codecs are protected by strict licences so we are unlikely to see them making an appearance in an Elektor multimedia player in the near future but stay tuned – we're working on it...

(090928)



Useful spares or toxic waste?

The Feedback form on the Elektor website is the first place to go if you have a question about any of our projects or products (as well as giving us feedback on any article that caught your eye). General technical questions can best be posted on the reader's forum on our website. There are exceptions though. Unable to find an answer in the forums a reader wrote to us recently with a question that's been bothering him: "I am sure I am not alone in having a large box of old components lying around under my bench. Is it possible to mix newer RoHS compliant components with older non-RoHS components? Any tips from the Elektor lab?"

We put the question to Antoine Authier, head of the laboratory here at Elektor:

"It depends entirely on whether the components are fitted to equipment which will then go on to be sold commercially. Here in the Elektor labs our production facilities conform to the current regulations for commercial production. With very few exceptions we use RoHS compliant components and leadfree solder. For purely personal use however there is no reason why you cannot mix RoHS with non-RoHS in the same design although the aim of course is to reduce hazardous substances in the environment".

Antoine also recommends using lead-tin solder; it's easier to make a clean joint. Lead-free solder has a higher melting point so we can assume that RoHS compliant components are designed to tolerate the higher temperature encountered during the soldering phase but sadly this is not always the case and we still occasionally get problems! On the workbench at home the average hobbyist isn't able to control the temperature and recommended soldering time to the same degree. In general the older non-RoHS component types that soldered without trouble are likewise in their RoHS variant also likely to be fairly trouble-free.

(100220)

Subscribe to Circuit Cellar now!

This magazine has a topic that keeps me busy on making prototype projects, advances in electronic design ideas, & always wondering what would be the next. For me, being informed in my field of expertise, it really adds confidence in my career.

> R. Arcilla, Celestica Philippines, Inc.

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Juggling Audio Bits Audio DSP for DIY applications

By Harry Baggen (Elektor Netherlands Editorial)

Audio hobbyists usually confine their hobby to the analogue domain, since the opportunities for doing your own experimenting in the digital domain are very limited. There is very little affordable equipment available that allows a multitude of digital-audio processing operations and getting started with DSPs yourself requires a considerable depth of knowledge of this subject matter. With the modules from miniDSP you can easily realise all kinds of audio processing functions without the need to become intimately familiar with digital signal processing.

In the audio world, DIY is most certainly not dead yet. There are still plenty of people who design/build amplifiers and speaker boxes or modify existing equipment. Despite the fact that the source material (the music) and the playback equipment are usually digital, there are still sufficient opportunities for tinkering to your heart's content with the analog stages that follow. Nevertheless, you will have noticed that more and more audio processing is taking place in the digital domain, ranging from signal processors to digital power amplifiers. This is quite a complicated world for the average audio hobbyist, where there are now few or no opportunities for doing anything yourself. Still, it is becoming increasingly interesting to familiarise yourself with the possibilities that these digital technologies have to offer. A digital crossover filter for an active loudspeaker system comes to mind, which would allow easy adjustment of the crossover frequency, slope, filter type and delay between the drivers.

It is true that there are a few systems on the market which offer you the possibility of experimenting yourself (for example the amplifier module AS2.100 from Hypex, which contains two digital power stages and a DSP for filtering and equalisation, or the semi-professional DCX2496), but overall the choice is very limited.

The young company miniDSP, established in Hong Kong, comprises several engineers who came up with the idea of marketing flexible and affordable digital audio products for both hobbyists and professional audio equipment manufacturers. The objective here was to keep the programmability of these products as simple as possible. In practice this means that you buy a board with a DSP on it and then you choose the required functionality from the software applications that they offer. This software is configured using a comfortable user-interface on your PC. So you only have to type in a crossover frequency and a slope for the filter and the DSP will do exactly what you have asked it to do. There is no need for extensive knowledge of DSPs and programming, and (almost) none for digital audio technology.

The basis of the entire system is the miniDSP board, a printed circuit board measuring 7.5 by 7.5 cm, which contains a DSP and a number of connectors. To this you can add an I/O-board, which has a number of digital inputs and outputs (miniDIGI) and a board with four

digital power stages (miniAMP). All boards have identical dimensions and can easily be stacked together and interconnected. Using the so-called audio plug-ins you can determine the functionality of the DSP board, that is, the software you put into the DSP. At the moment, plug-ins are available for 2- and 4-way crossover filters with equalizer (in simple and advanced implementations) and a mixer with 31-band graphic equalizer. Work is in progress for other plug-ins.

The plug-ins work in combination with Adobe Air, which you have to install on your computer beforehand. Double-clicking the downloaded plug-in is then sufficient to install it.

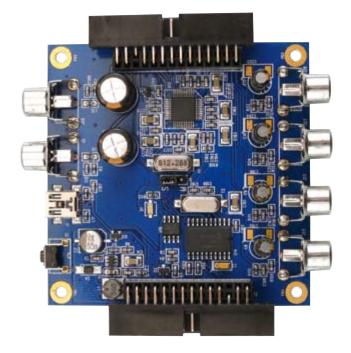


Figure 1. The miniDSP board has two analogue inputs and four outputs.



First, let's take a look at the hardware. At the core of the miniDSP board (Figure 1) is a ADAU1701 from Analog Devices. This processor was developer specifically for audio applications. Most of the calculations are carried out in 56-bit double precision mode for an accurate result. The IC, in addition to the DSP, also contains 24-bit A/D- and D/A-converters which operate according to the sigma-delta principle, which ensures a large dynamic range. The program for the DSP is automatically read from the serial EEPROM on the board when the power supply it turned on. In addition to the DSP and EEPROM the miniDSP board also contains a PIC18F14K50, which

takes care of the communications between the computer and DSP, using a USB-interface. The board can be easily connected to other audio components using the RCA connectors (2 inputs, 4 outputs). The board uses audiophile quality electrolytic capacitors from Nichicon for a minimal influence on the analogue signals. On each side of the board there is a header for interconnecting all the relevant signals to other miniDSP boards. There is also the option of connecting a potentiometer, which can be used to control the volume in the DSP.

If you require digital inputs or outputs (for example, because you want to go directly from a CD player, or you want to connect the processed signal in digital form to some other equipment), then your needs are met with the miniDIGI board (Figure 2). Here you will find two coaxial and two digital inputs and a coaxial and optical output. This board is available with and without transformers for galvanic isolation of the coax connections. The heart of this board consists of a sample-rate converter (SRC43821 from TI) which will convert all sample rates up to 216 kHz to 48 kHz, which is the operating sample rate of the DSP. Using a jumper you can select which of the 4 inputs is used and another jumper is used to select which signal is routed to the digital outputs (for example the input signal or the DSP processed signal). The board can also be used to de-jitter an S/PDIF-input signal. The last of the currently available boards is the miniAMP (Figure 3). This board contains

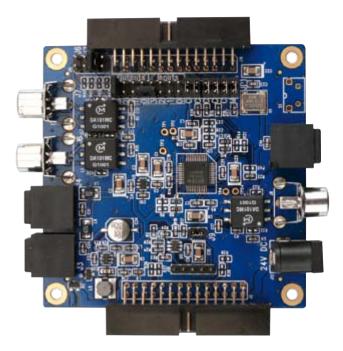


Figure 2. The miniDIGI board contains a sample rate converter and offers a substantial number of inputs and outputs.

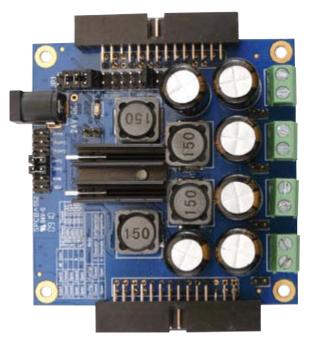


Figure 3. The third board, the miniAMP, contains a class-D amplifier IC with four output channels, which can also be bridge connected.

REVIEW: AUDIO DSP



Figure 4. The design of the software. From this overview you can select each function block.



Figure 5. Here we can see the part for the crossover filter. You can select the crossover frequency, type of filter and slope. At the same time you can choose a band-pass filter.

four digital amplifiers, which can be bridge-connected for a stereo version with a higher output power. The power amplifier IC on this board is a TAS5704 (also from TI). This can generate an audio power of 4 x 10 W into 4 Ω or 2 x 20 W into 8 Ω in bridge configuration. Thanks to the class-D design of this amplifier it has a high efficiency (90%) and a small heatsink mounted directly on the PCB suffices for the cooling. The electrolytic capacitors are made by Nichicon, but these capacitors are not necessary in bridge configuration and can be linked out using jumpers. The TAS5704 is presented with the audio signal in digital form, so the D/A converters of the DSP are therefore not used. Although the output power of this board is not all that high, it is ideal for experimenting because you have 4 power amplifiers at your disposal. Just connect the drivers via a few cables to the screw terminals and you can immediately evaluate the effect on the sound of any of the DSP settings that you make.

The printed circuit boards all have identical dimensions and can be assembled one on top of the other with stand-offs. Short ribboncable interconnects are supplied to make the connections between the boards. The power supply for the miniDSP board an be provided by the USB connector mounted on the board. The miniDIGI board and miniAMP board both contain a standard DC power supply jack for connecting a mains power supply. The power supply voltage may range from 4.5 to 24 V for the miniDSP and miniDIGI, the mini-AMP board requires 12 to 24 V.

Connecting and testing

For this test we received a complete set, comprising a miniDSP, miniDIGI and miniAMP. We selected a miniDSP version with an input range of 2 V, so that you can directly connect the output from a CD player to it (there is also a version with a 0.9 V sensitivity available, to which you could easily add a voltage divider, if need be). The boards are connected together using very short interconnecting cables (see opening photo). There are plenty of possibilities with this complete set of boards, but in practice you would only buy those boards that you need for your particular application. If, for example, you want to add a digital crossover filter to a loudspeaker box with analogue inputs and use existing power amplifiers then

you will only need to buy a miniDSP board. If you would like digital inputs with that then you will also have to buy the miniDIGI board. And should you want to experiment regularly with several different loudspeaker boxes, then the miniAMP board could be a very handy accessory.

The boards are supplied completely assembled. You effectively only have to connect a power supply and you are ready to get started. Well, that is after you have set all the jumpers on all the boards appropriately. In particular the miniDIGI board contains a large number of jumper settings, among others, for the correct interconnections of the I2S signals between the boards. This can be quite confusing initially. Fortunately the miniDSP website contains documentation with various example configurations, which show exactly which jumpers have to be placed where for a certain combination of boards or application. In addition, the developers have advised us that they are continually busy with further improvements and expansion of the documentation. Even during the time we were doing our evaluation there appeared several updates to the software and documentation.

For the software-example for this test we choose a two-way filter with built-in parametric equalisers. This is an excellent test project for such a system.

In the first instance we connected the entire module to our Audio Precision System II to check the operation and perform a few measurements. The module was powered from a large regulated mains power supply (power supply voltage was 24 V).

To get a feeling for the affect that the DSP has on the analog audio signal, all the filters in the software were set to 'straight through' and we supplied a digital signal via the miniDIGI board. We then measured the distortion of all the analog outputs of the DSP. These were very low, about 0.005% at 1 kHz at nearly 100% signal level. This value changed very little when we switched to the analogue inputs of the DSP, that is, the A/D converters are now also in the

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Figure 6. Every parametric equaliser of the two-way plug-in allows a maximum of six corrections (frequency, value, Q factor). Here a simple bass-boost has been realized.

signal path. The signal-to-noise ratio is at more than 90 dB (linear), which corresponds to the value that AD specifies for the ADAU1701. The class-D amplifier on the miniAMP board operates at a frequency of about 400 kHz (changes slightly based on the applied sample frequency). The residual of this frequency at the outputs has an acceptable magnitude of about 100 mVp. The harmonic distortion in single-ended configuration amounts to about 0.07% at 100 Hz and 1 kHz/1 W/8 Ω . The output power at 1% THD and 24 V power supply voltage turns out to be 14.5 W into 4 Ω and 8 W into 8 Ω . This is very close to the specifications from TI. We didn't measure the bridge configuration, but this will certainly deliver 20 W into 8 Ω , as reported in the data sheet for the miniAMP.

In practice

After all this it was time to play with the software and have a look at how we could realise a practical circuit. After installing Adobe Air runtime (free download from the Adobe website) and downloading the two-way crossover plug-in from the miniDSP website, you only need to double-click the downloaded .air-file after which the installation is automatically carried out and shortcuts are also created. Now we only need to connect the DSP-board, using a cable with a mini-USB connector, to the PC and we are ready to start experimenting.

When the program is first started the tabbed page 'Audio Settings' appears, which shows a block-diagrammatic overview of the design of the software. The tabbed page 'System Settings' contains a number of general settings, such as the type of input signal (analogue or digital), activating an external volume control, and saving and loading of the configurations (in xml-format). In this way you can save all the settings that you have made and easily retrieve them at some later time. There is also a button here to return everything to the default settings. Finally there is a third tab that brings you to the miniDSP website. All plug-ins follow the same general design. When clicking on one of the function blocks on the 'Audio Settings' page, the corresponding settings appear. With the two-way crossover plug-in you can select in the first block the attenuation of the



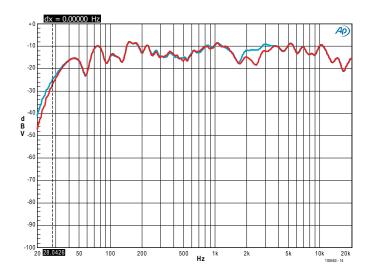


Figure 7. MLS-measurement where the time delay between woofer and tweeter was changed in small steps until the response around the crossover frequency and time delay was measured, Red = no time delay, blue = 0.2 ms time delay, about 7 cm.

input signal. This is followed by a parametric equalizer for each channel with up to 6 configuration settings. Subsequently we arrive at the crossover part where for each driver you can select the crossover frequency, type of filter and the filter slope. The available filter types are Bessel, Linkwitz-Riley (12, 24 and 48 dB/octave) and Butterworth (6 to 48 dB/octave). You can also expand each filter to a band-pass type, for example to protect a small woofer from frequencies that are too low or to combine a two-way system with an already existing active sub-woofer. This section is then followed by the 6-band parametric equalizer, which allows individual corrections to be made for each driver. In the final block you can set the attenuation and time delay for each driver. The latter is particularly important with the Linkwitz-Riley filter type because the woofer and tweeter need to have their acoustic centres perfectly above each other. The correction has quite a wide range, up to 7.5 ms, that is equivalent to more than 2.5 m.

Once all the necessary settings have been made you can click the green 'Synchronize' button and the software will make the connection with the (via the USB-cable connected) miniDSP board. The settings you made are then sent to and stored on the miniDSP board. The firmware for the DSP is also automatically updated, if that is necessary. The green button disappears after synchronising, which appears a little strange at first. Is no further synchronisation possible? There is. After synchronizing, all the changes you now make are immediately transmitted to the miniDSP board, in other words, you have a 'live' connection. So you can experiment to your heart's content and change all kinds of things. You can, for example, connect a music signal and hear immediately what the effect of each individual change is on the sound image. So while experimenting with our two-way speaker box we could very easily change the level of the tweeter while listening to it and even play with the time delay between woofer and tweeter. A fantastic opportunity for experimenting with DIY loudspeaker systems! When the program on the PC is closed the last settings are automatically stored in the memory on the miniDSP board.

When you're done experimenting, you can mount all the boards

in the loudspeaker enclosure, together with a suitable power supply and power amplifier. As already mentioned, the miniAMP is very handy for experiments, but for a final loudspeaker system it would be better to choose a few bigger (and better quality) power amplifiers.

Very handy

With the miniDSP system you can in a very simple manner and a very short time assemble an audio circuit which would be much more difficult to realize using analog components. Consider an active crossover filter with delay time correction and a few frequency corrections. With this system it is done in the blink of an eye. You do not require any knowledge of DSPs, but you will need, of course, knowledge and experience in the area of filter and loudspeaker technology. But this you will need anyway, if you are going to develop your own loudspeaker systems. The method of directly changing the settings from your computer works really well and even encourages you try out many more things than you were perhaps originally planning to do. After all the experimenting you can simply build a final version by building the miniDSP board with power supply and suitable power amplifiers in a loudspeaker box. And, if after a while you decide that these settings are not quite ideal, then it is very quick to connect a PC and make the necessary corrections.

Despite the extensive capabilities of the miniDSP system it is very affordable. For most applications you will only need the miniDSP board and you're done for \$99 plus \$10 for the plug-in. One board on its own is capable of driving two two-way speaker boxes. However in most cases it will be more convenient to build a miniDSP board into each speaker box, in addition, the board has sufficient computing power to implement a four-way filter.

The audio quality of the miniDSP board is quite good, but we can imagine that some audio enthusiasts are not satisfied with this and would prefer to use other A/D- and D/A-converters than those that are built into the DSP. This is also a possibility with this board, by using the I2S-in- and outputs on the miniDSP board. And there are many more adjustments and options... The most important thing is that with these boards you can very easily make a start with digital audio processing. And all that for a very attractive price!

(100453-l)

Internet Link

www.minidsp.com

miniDSP, rev A (0,9 V) of rev B (2,0 V): \$99.00 minDIGI rev A (less S/PDIF-trabsformers): \$55.00 minDIGI rev A (met S/PDIF-trafo's): \$60.00 miniAMP: \$60.00 Audio plug-ins: \$10.00 each

Kits of various combinations of boards are available as well.





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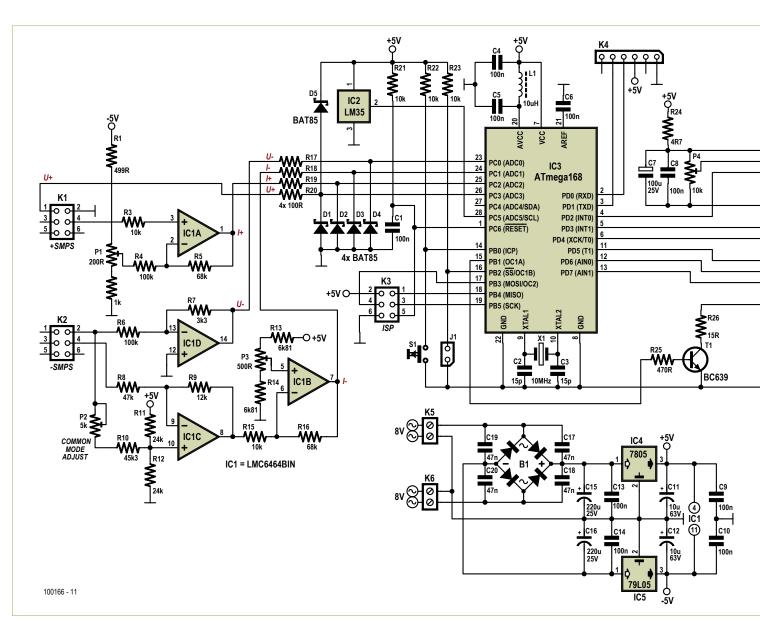
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Dual Voltage/Current Display For a symmetrical Unilab power supply

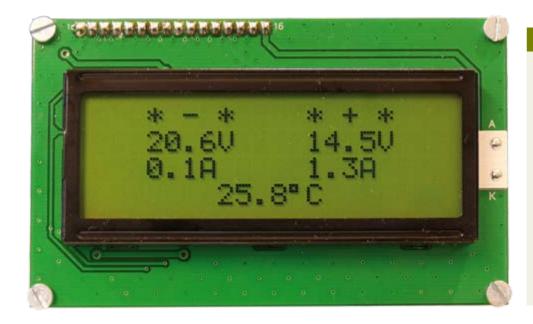
By Ton Giesberts (Elektor Labs)

The Unilab power supply module described in the April 2010 edition is a compact, adjustable switch-mode power supply unit that is very well suited to building a single or dual lab power supply. For this application, we have developed a special meter circuit in the Elektor Labs with a four-line display that shows the output voltages and currents of a symmetrical configuration with two Unilab circuit boards.

If you wish to build your own lab power supply, it's only natural to want to equip it with separate meters for voltage and current. With a symmetrical power supply, this amounts to two sets of two meters. Although a variety of small, ready-made display modules with LCD or LED readout are commercially available, they have two drawbacks. The first is that they aren't espe-



UNILAB DISPLAY



Features

- Readings shown on a large backlit LCD module with 4 lines of 20 characters
- Voltage and current of each supply output (positive and negative) displayed
- Internal temperature of PS enclosure also displayed
- Selectable °C or °F temperature indication
- Signal conditioning circuitry specifically designed for a dual Unilab power supply
- Software-aided calibration procedure
- Powered by a separate transformer (2 × 8 V / 3.3 VA)

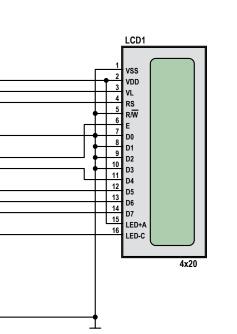


Figure 1. Schematic diagram of the dual voltage/current meter for a Unilab power supply. Naturally, it can also be used for other applications. cially cheap, which means you're looking at a tidy sum if you need four of them. The second is that connecting modules of this sort to a power supply is always difficult. The amplitude and polarity of all of the measured quantities must be adapted to obtain optimal resolution with the selected modules, and in many cases it's necessary to make certain measurements with the meter floating relative to the actual power supply. As the Unilab module is highly suitable for putting together a symmetrical DIY lab power supply, we decided to develop a separate display unit for this purpose with a large LCD module that shows the voltage and current of both halves of the supply, along with the temperature inside the enclosure. All in all, this solution is certainly no more expensive than four separate modules, and what's more it is much more attractive and guaranteed to work properly. The latter reason is probably the best argument for building your own display module.

Design

Before we explain how the circuit works in detail, it's a good idea to describe the overall design (**Figure 1**). The 'core intelligence' of the circuit is provided by an ATmega168, an Atmel microcontroller with (among other things) six A/D converters. Here we use four of them to measure the two load currents and the two output voltages. The clock signal for the microcontroller is generated using an external 10-MHz crystal. The portion of the circuit around opamps IC1a–IC1d provides signal conditioning for the current and voltage sense signals before they are fed to the A/D converter inputs of IC3 (ADC0–ADC3). This portion is discussed separately below. Resistors R17–R20 and diodes D1–D4 protect the A/D converter inputs of the microcontroller against negative voltages. Only the positive inputs need to be protected against excessive positive voltages, since the other measurement signals never exceed +5 V.

The LCD module that displays all the readings has 4 lines of 20 characters and is driven directly by the microcontroller. The contrast can be adjusted using P4. Transistor T1 (which is driven by port PB1) allows the brightness of the backlight to be adjusted or switched off. The software causes the backlight to blink if the power supply is overloaded (output current greater than 3 A). The supply voltage line to the LCD module is strongly decoupled by R14, C7 and C8 to avoid any interference on the A/D inputs. As a bonus, we included an LM35 (IC2) that can be used to monitor the temperature inside the power supply enclosure. The temperature sensor provides an output voltage of 10 mV/K and is powered from the +5 V rail. If you want to have the temperature displayed in degrees Fahrenheit, simply fit jumper [1. In this case the 'C' on the display is replaced by 'F' and the Celsius temperature signal from the LM35 is converted to Fahrenheit using the formula

$T[^{\circ}F] = T[^{\circ}C] \times 9/5 + 32.$

K3 is the ISP connector. This is handy if you want to modify the software to suit your own wishes. K4 is a debug connector. An Elektor USB/TTL adapter cable (# 080213-71) can be connected directly to K4.

COMPONENT LIST

Resistors

(0.25 W unless indicated otherwise) $R1 = 499\Omega \ 1\% \ 0.6W$ R2 = 1.00kΩ 1% 0.6W R3,R15,R21,R22,R23 = 10kΩ 5% $R4, R6 = 100 k\Omega 5\%$ $R5,R16 = 68k\Omega 5\%$ $R7 = 3.3k\Omega, 5\%$ $R8 = 47k\Omega 5\%$ $R9 = 12k\Omega 5\%$ $R10 = 45.3 k\Omega \ 1\% \ 0.6 W$ R11,R12 = 24.0kΩ 1% 0.5W R13,R14 = 6.81kΩ 1% 0.6W R17,R18,R19,R20 = 100Ω 5% $R24 = 4.7\Omega 5\%$ R25 = 470Ω 5% $R26 = 15\Omega 5\%$ P1 = 200Ω trimpot, e.g. Vishay Spectrol type M64W201KB40

- $P2 = 5k\Omega$ trimpot, e.g. Vishay Spectrol type M64W502KB40
- P3 = 500Ω trimpot, e.g. Vishay Spectrol type M64W501KB40
- P4 = 10kΩ trimpot, e.g. Piher type PT10LV10-103A2020

Capacitors

- C1,C4,C5,C6,C8,C9,C10,C13,C14 = 100nF ceramic, lead pitch 5mm
- C2,C3 = 15pF 2%, ceramic, lead pitch 5mm
- C7 = 100µF 25V, radial, lead pitch 2.5mm
- C11,C12 = 10µF 63V, radial, lead pitch 2.5mm
- C15,C16 = 220µF 25V, radial, lead pitch 3.5 mm
- C17,C18,C19,C20 = 47nF, ceramic, lead pitch 5mm

- nductor
- L1 = 10μH 10%, axial, Rs = 0.29Ω, e.g. Epcos type B78108S1103K

Halfgeleiders:

- D1,D2,D3,D4,D5 = BAT85
- T1 = BC639
- IC1 = LMC6464BIN
- IC2 = LM35CZ
- IC3 = ATmega168-20PU, programmed, con-
- tained in kit # 100166-71
- IC4 = 7805
- IC5 = 79L05
- B1 = 80V, 1.5A bridge rectifier (pinout ~ + ~ -), e.g. Semikron SKBB80C1500L5B

Miscellaneous

- K1,K2,K3 = 6-pin (2x3) pinheader, lead pitch 0.1 in. (2.54mm)
- K4 = 6-pin pinheader, lead pitch 0.1 in. (2.54mm)
- K5,K6 = 2-way PCB terminal block, lead pitch 5mm
- J1,S1 = 2-pin pinheader, lead pitch 0.1 in. (2.54mm)
- LCD1 = 16-pin pinheader, lead pitch 0.1 in. (2.54mm)
- LCD1 = 16-pin SIL connector for display, lead pitch 0.1 in. (2.54mm)
- X1 = 10MHz quartz crystal, HC-49/S, 50ppm, $C_1 = 12pF$
- LCD, 4x20 characters with background lighting, dim. 60 x 98 mm, e.g. HTDISPLAY HC200401C-YF62L-VA
- PCB # 100166-1
- Software en PCB design files: free download * Kit of parts incl. PCB, programmed controller
- and LCD: Elektor order # 100166-71 '
- * Details at www.elektor.com/100166

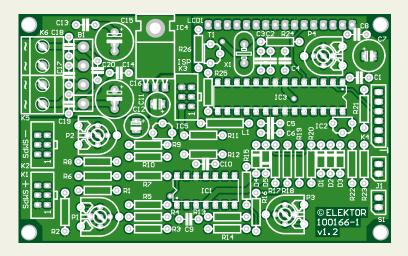


Figure 2. The double-sided PCB is exactly the same size as the display module.

Tweaking and shifting

To build a symmetrical lab power supply, you connect two Unilab power supply modules (# 090786-1 or 090786-71) in series, with the junction of the plus output of the one module and the minus output of the other module forming the neutral terminal of the overall power supply. Unfortunately, this makes it difficult to measure the output current of the negative supply, since the sense resistor for current measurement is located in the negative output line of the power supply module. This means that the voltage on the sense resistor floats on top of the output voltage. Consequently, the output voltage must be suppressed in order to measure the voltage across the sense resistor.

The Unilab circuit board has a connector (K1 on PCB # 090786-1) that makes it easier to measure the current and voltage. With the positive supply module, the voltage and current can be measured using pins 1 and 4 of this connector. Pin 2 is connected to the minus output terminal on the PCB and thus serves as the ground connection point (the neutral terminal of the symmetrical power supply) for the display PCB. With the negative supply module, pins 2 and 4 are used to measure the voltage and current, respectively. Here pin 2 is connected directly to the negative output voltage, and the voltage between pin 2 and pin 4 is proportional to the voltage across the sense resistor. In order to allow the various quantities to be measured accurately using the internal

A/D converters of the microcontroller, the full range of the voltage or current sense signal must lie within the full-scale range of the A/D converter. Here we use the internal voltage reference, so the full-scale range of the 10-bit converters is 1.1 V. This means that it is not possible to measure negative voltages.

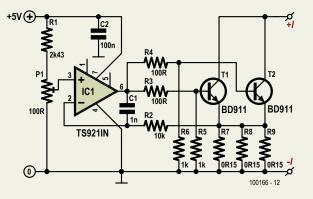
Measurement setup for the positive supply

As the positive output voltage can range from 0 V to approximately 30 V, all we need in order to measure this voltage is a voltage divider. It is located on the power supply board (R15 and P7), but it must first be adapted to our requirements. By reducing the value of R15 on the positive supply module board to 5.6 k Ω (preferably using a 0.6-watt metal film resistor), we can use P7 to adjust the voltage on pin 26 of the microcontroller to less than 1.1 V at the maximum no-load output voltage. The exact value is not especially important, since everything is sorted out by the calibration procedure.

Test Circuit

We designed a simple current source (see the schematic diagram) to help you align the signal conditioning circuitry around IC1 and perform the subsequent calibration. To allow this test circuit to operate at the lowest possible voltage, it must have its own source of 5 V power. We assume that everyone will have a suitable supply available.

This is a conventional current source design using an opamp that compares the voltage across the emitter resistor of a transistor with a set value, which in this case is taken from the wiper of P1. Here the transistor consists of two devices connected in parallel (T1 and T2), which can jointly handle a substantial current. This approach also distributes the heat load more evenly over the heat sink. If you use three standard 0.25-watt resistors with a value of 0.15 Ω for the emitter resistor, this circuit can easily supply currents as high as 4 A. The usual practice when transistors are connected in parallel is to provide each transistor with its own emitter resistor. This is not possible here, and a specific voltage drop across each emitter resistor is also necessary for proper operation. To nevertheless ensure a reasonable degree of current sharing, we use individual base resistors here. However, this only works well if the gains of the transistors are roughly the same, so you must pay attention to this. C3 and R2 stabilize the feedback loop.



The maximum amount of power that must be dissipated is 120 watts (for example, $3.5 \text{ A} \times 34 \text{ V}$). This is rather a lot and calls for a substantial heat sink. We solved this problem by using a fairly small heat sink together with a hefty fan to provide strong forced-air cooling.

The transistors are ST types in a TO-220 package. We chose the TS921IN rail-to-rail device for the opamp. It can supply enough current (80 mA) to drive the two transistors.

In theory, it should be possible to set the current to just under 4 A with voltage divider P1/R1, but the tolerance of the potentiometer may cause the actual value to vary considerably. If the maximum current is insufficient, you can increase the supply voltage a bit (but do not exceed 12 V) or adjust the value of R1.

ential amplifier using only one opamp. As we need to be able to handle voltages up to more than 30 V and the IC supply voltages are only \pm 5 V, this stage also attenuates the signal. This avoids applying voltages to the IC that exceed its input voltage range.

Component dimensioning around IC1c requires careful attention. For example, we

must bear in mind that the voltage may be as high as 33 V. At 33 V, the voltage divider formed by R10, R11, R12 and P2 causes the voltage at the inputs of IC1c to be slightly less than -4 V. To allow the signal level to be adjusted by P3 in the following stage (IC1b) so it lies within the allowable range, this stage inverts the output of the differential amplifier. This means that the volt-

Measuring the current is a bit more complicated. On the power supply board, the voltage across the sense resistor is first fed to amplifier (IC3a) with a gain of approximately 4. The output signal from this stage is available directly on pin 4 of connector K3 of the power supply module. Due to the design of the Unilab module, the output voltage of this amplifier stage is negative, but it has the same slope as the voltage across the sense resistor. If the voltage across the sense resistor increases, the output voltage of the amplifier becomes less negative. This means that we need a circuit on the display board to shift the voltage into the input range of the A/D converter and amplify it a bit more to obtain maximum precision. This task is handled by opamp IC1a, with an additional gain of 1.68. With the gain of IC3a in the power supply module and a maximum signal level of 150 mV across the sense resistor, this yields a voltage of approximately 0.97 V at the A/D converter input. As the 0 A setting can be adjusted reasonably close to 0 V with P1, the circuit also has some headroom to display higher currents (due to supply overloading), up to a maximum of around 3.4 A. The –5 V supply rail is used as the reference voltage for voltage divider R1/R2/P1.

Measurement setup for the negative supply

The only thing we need to measure the negative power supply output voltage is an inverting amplifier (IC1d), which attenuates the output voltage (maximum value 30 V or more) to match the input range of the A/D converter. An output voltage range of 0–33 V can be measured with the specified values of R6 and R7 (–3.3 V \div 100 \times –33 = 1.089 V).

In order to convert the voltage on pin 4 of K2 into a usable current reading, we need a bit more than what we use for other signal processing stages. Although the current sense signal has the same form as the signal for the positive output voltage, here it floats on top of the output voltage. Since we're already sensing the output voltage, we can use a differential amplifier (IC1c) to suppress this voltage as a common-mode signal. Here we implement a classic differ-

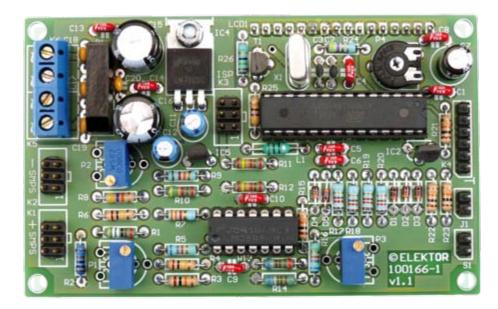


Figure 3. The fully assembled prototype. It is fitted with the trimpots specified in the components list, but there is enough space available to fit other types.

age across the sense resistor must also be inverted by the differential amplifier, so the signal from pin 4 of K2 is fed to the inverting input of IC1c. The voltage divider ratio (which can be adjusted with P2) must match the ratio of R9 and R8.

For the opamp, we chose a National Semiconductor micropower CMOS device with rail-to-rail input and output capability. The supply current, speed and bandwidth are not important here because we are only processing DC signals. However, the railto-rail properties are important, along with the low bias current (typically 150 fA, so extra compensation is not necessary) and the excellent temperature coefficient of the input offsets (only $1.5 \,\mu$ V/°C). The various input offsets (a few millivolts) are compensated by the calibration procedure.

Power supply

It's a good idea to galvanically isolate the power supply of the display board from the Unilab modules in order to eliminate potential ground loops. The implementation here is straightforward. A transformer with symmetrical secondary windings, a bridge rectifier and smoothing capacitors form the basis for the symmetrical 5-V supply for the display board. We use a standard 7805 for the +5 V supply voltage. To improve heat dissipation, this voltage regulator is fitted flat on the PCB without any insulation (the PCB has copper planes on both sides). Note that the +5 V supply must also provide the current for the backlight of the LCD module, which can draw more than 100 mA (47 mA in our prototype). The analog supply voltage for the microcontroller is separately decoupled by a choke and a capacitor (L1 and C5). All we need for the –5 V supply is a 79L05, since it only has to power the analogue circuitry of the input stage. The current consumption on this supply rail is quite low.

A standard short-circuit proof transformer rated at 3.3 VA is a reasonable choice for the supply transformer, preferably a type with dual 8-V secondary windings (or alternatively dual 9-V windings). However, we took a different approach for our prototype. The transformer for the two Unilab modules is a toroidal type, and it's easy to add a new winding to it. To determine how many turns are needed. you first have to make a winding with a few turns (such as 5 or 10) and then measure the voltage. After this you can simply calculate the number of turns needed for 8 V, since the transformer has a uniform magnetic field. You can calculate how much wire you need by measuring the circumference of the core and adding some more for the leads to the PCB. For our prototype, we needed approximately 5.5 metres (18.5 feet) of wire for each winding (32 turns). We used bifilar winding (two wires together) to obtain two identical windings. Use relatively thick stranded wire with robust plastic insulation for the windings.

Assembly

Assembling the double-sided PCB shown in Figure 2 should not present any problems, since only standard components are used (i.e. no SMDs). As already mentioned, the 7805 is mounted flat on the PCB and secured by a screw. With this arrangement, the PCB provides adequate cooling. You can choose from various types of trimpots here, either horizontal or vertical. Figure 3 shows the prototype board fitted with the types specified in the components list. The 16-way headers are fitted on the back of the PCB. The display module has a 16-pin connector on the back of its circuit board, and it can be plugged into the header of the assembled PCB. The two boards can be fastened securely together using standoffs (12 mm height).

Suitable software must be loaded into the microcontroller before the board is put into service. You can do this yourself by using the ISP port to download the software (the C source code and hex file are available free of charge on the Elektor website under product # 100166-11), or you can order a pre-programmed microcontroller from the Elektor Shop (# 100166-41).

When fitting the two Unilab modules and the display board in an enclosure, be careful to ensure that the plus terminal of connector K2 on the negative power supply board is connected to the minus terminal of connector K2 on the positive power supply board by a thick wire. This connection forms the ground reference for the display board. Be sure to replace R15 on the positive Unilab board by a 5.6-k Ω resistor. In addition, connector K3 on each Unilab board must be connected to connector K1 or K2 (respectively) on the display board using individual 6-way IDC connectors and 6-way flat cable. Be sure to get the connections right! With regard to the remaining wiring of the Unilab supply modules, we suggest that you consult the article in the April 2010 edition. The only other thing you need to do is to connect the two 8-V supply inputs of the display board to the supplementary power transformer or the additional wind-

Alignment and calibration

After assembling the boards and fitting them in the enclosure, you can start the alignment procedure for trimpots P1 to P3. Set the positive Unilab module to its maximum output voltage and measure the voltage on one of the leads of R20 of the display board. Adjust P7 to obtain a reading less than 1.1 V (you may be able to simply leave P7 at its maximum setting).

The next task is to align the current signal conditioning stage. Adjust the output of the positive Unilab supply to exactly 0 V and ensure that no load is connected. Adjust P1 to obtain a voltage of exactly 0 V (or slightly higher) at the output of IC1a. Then check whether the voltage on R19 is less than 1.1 V when the output current is 3 A. This does not have to be done at the maximum output voltage; it can be checked just as well at an output level of a few volts. This way you can use a relatively small load resistor, such as $1 \Omega / 10$ W with an output voltage of 3 V.

No alignment is necessary for the negative output voltage. Unfortunately, the alignment of the signal conditioning circuitry for the negative output current is distinctly more complex.

First the effect of the negative output voltage must be suppressed. This is adjusted with P2. In order to do this properly, we recommend that you build the current source described in the inset (tit also comes in handy during the subsequent software calibration procedure). Now connect an ammeter (with a range of 10 A or so) in series with the current source and connect this arrangement to the negative Unilab output to serve as a load. Set the current to exactly 3 A and vary the output voltage back and forth over the range of 2 V to 25 V. If everything is OK, the current through the current source will remain constant. In this case vou can disconnect the meter and then use it as a voltmeter. Measure the voltage at the output of IC1b or IC1c. Adjust P2 until this voltage remains constant when the output voltage is varied. Bear in mind that the signal is filtered by IC3a in the power supply module, so you must always wait a few seconds (or longer) after each change to allow the voltage to stabilise. Disconnect the load. With the current now zero, adjust P3 so the voltage on R18 is 0 V (or slightly higher). After this you can perform the calibration procedure.

The calibration routine allows for three possible situations when the supply voltage is

switched on:

- 1. An Elektor-programmed microcontroller by is fitted in the circuit.
- 2. A user-programmed microcontroller is fitted in the circuit.
- 3. Calibration has already been performed.

In the first case, provisional parameter values have already been stored in the EEPROM memory by Elektor, but calibration is still necessary. In this case, a message requesting you to press S1 is displayed for 5 seconds after power is switched on. If this time expires without S1 being pressed, the display will show the voltages and currents, but the readings may not be correct. As a reminder of this, '[uncal]' is shown at the start of the bottom line. This reminder disappears after calibration has been performed.

If you programmed the microcontroller yourself using the ISP port, data is also stored in the EEPROM (this is part of the available download) and the same course of events occurs as in the first situation. However, if you programmed the microcontroller in some other manner, there is probably no data in the EEPROM. In this case the display indicates that calibration is necessary after the unit is switched on, since the display cannot present meaningful readings without calibration data.

If calibration has been performed, the display shows the correct readings after first presenting a welcome message.

Calibration

Button S1 is used for the calibration procedure. It is not present on the PCB; you must connect an external pushbutton for this purpose. After S1 is pressed once, the voltage, current and temperature screen is replaced by the message screen for the first step of the calibration procedure:

CALIBRATION STEP 1 set outputs to 0V then press S1

Pressing S1 stores the values of the first our parameters, which are used to determine the four zero levels. They are labelled U1L, I1L, U2L and I2L in the software, where 'U1' represents the negative supply module.

Press S1 again to display the message screen for step 2 of the calibration procedure:

CALIBRATION STEP 2 set outputs to 25V

draw 3A from -PSU then press S1

The second step determines the values of the two parameters (U1H and U2H) for the maximum output voltages that the supply modules must be able to deliver (25 V max.) and the parameter (I1H) for the maximum specified output current (3 A). If you have built the current source described in the inset, you can put it to good use here. Otherwise you must provide the right load resistance (such as twelve 100- Ω , 10-W resistors connected in parallel) to draw a current of 3 A from the negative supply output. To ensure that the output voltage is still exactly 25 V, measure it once again with the load connected before pressing S1 again. Pressing S1 now takes you to the third and final step of the calibration procedure:

CALIBRATION STEP 3 draw 3A from +PSU then press S1

In this step you determine the value of the eighth and final parameter (I2H) by connecting a 3 A load to the positive supply output. The output voltage does not matter as long as the output current is 3 A. If you use the current source, always set the current to 0 A before connecting it to the supply or disconnecting it (i.e. do not make or break the connection at 3 A). Thanks to this third step, only one multimeter and one current source or load resistor are necessary for the calibration procedure. Press S1 again to save this parameter value. The following message screen is displayed to conclude the calibration procedure:

CALIBRATION DONE! press S1

After you press S1 the last time, the microcontroller executes a restart and then presents the calibrated readings on the display. You can check them with a multimeter. We recommend that you check the readings after the modules have been fitted in the enclosure and repeat the calibration procedure if necessary.

After calibration has been completed, the display shows the following message for one second after the power supply is switched on:

SMPS display v1.0.0

Note that the actual version number may be different.

TEST & MEASUREMENT

High-voltage Probe Home made and differential, too

By Alfred Hesener (Germany)



Many circuits employ high voltages. The two best-known examples are switch-mode power supplies and vacuum tube circuitry. A relatively new application is found in hybrid and electric automobiles, which operate with high battery voltages (and dangerously high currents) in order to reduce voltage drop and the cross-sectional diameter of cabling.

Whilst any decent multimeter is adequate for measuring high voltages, it's not so handy for measuring small fluctuations in high DC or AC voltages overlaid on these. Frequently, moreover, we are not interested so much in the absolute value of a high voltage as much as in the difference between two separate high-voltage levels, such as (for example) the differing anode voltages of a push-pull amplifier or at switching nodes in a phase shift full-bridge topology transformer (within high output level switch-mode power supplies).

Starting point

One solution would be to use two standard high-voltage test probes (not too expensive, with serviceable characteristics) together with a digital oscilloscope so as to calculate the signal difference using mathematical functions. This method has three disadvantages:

 This means using two channels of the oscilloscope, which makes it harder to observe several signals simultaneously.
 Both signals are digitised at the resolution of the oscilloscope (generally 8 bits maximum), so that errors add up. Subtracting two large, almost identical voltage values is always problematic, increasing the risk of measurement errors.

3. Since the timing correlation between the two measurement channels is related to factors such as cabling and earth (ground) loops and mathematical functions within the oscilloscopes can throw up random and deterministic fluctuations, the timing information of the signal is not very trustworthy, particularly at higher signal frequencies.

Remedy can be found in the so-called 'high-voltage differential test probe'. This is a probe set enhanced with a differential amplifier able to accept very high voltages on its inputs and amplify only the voltage

TEST & MEASUREMENT

Characteristics

- Differential attenuation switchable in two stages (-20 dB/-40 dB)
- Bandwidth 1 MHz, switchable limit at 500 kHz
- Maximum input voltage ±1000 V (peak value)
- Maximum output voltage ± 10 V (at min. terminating impedance 1 $k\Omega)$
- Common-mode suppression 55 dB at 6 kHz, 35 dB at 600 kHz

kind provided with commercial differential test probes was omitted for reasons of simplicity.

Bandwidth limitation is important for measurements with spurious high-frequency signals, as in switch-mode power supplies. The frequency sweeps are shown in **Figure 1**.

The upper of the two curves indicates the output signal at -20 dB attenuation both with (in orange) and without (in green) bandwidth limiting. The blue line is the -40 dB setting with the 500 kHz filter switched in. As expected, the frequency response is very linear, with a maximum cut-off frequency of approx. 1 MHz. The red line shows the common-mode output signal relative to the frequency. By calculation this gives a CMRR of about 55 dB at low frequencies, which falls to about 35 dB at higher frequencies. This fall-off starts at around 6 kHz and is caused mainly by parasitic coupling inside the test probe. It is further influenced at higher frequencies by parasitic coupling within the op-amp.

The maximum input voltage of about ± 1000 V was soak-tested in the lab with constant voltage over an extended period. BNC connectors and suitable HV cables are specified for these tests. The input impedance of the circuit must likewise be laid out for high voltages. The resistors used (R1 in **Figure 3**) should be rated for 1600 V (standard resistors are usable up to only 250 V). Alternatively you can wire several resistors of lower voltage rating in series to divide the voltage (assuming all of these resistors have the same value).

One more tip concerning the measurement cables: the cabling on the two differential inputs should be as far identical as possible, since any disparity will unbalance the setup and cause measurement errors. For our prototype we cut a short coaxial test lead with BNC connectors into two halves of equal length, with insulated croc clips fixed to the free ends and the transition insulated with heatshrink tubing (see **Figure 2**). Because these cables represent a capacitive load to the measurement setup, they need to be as

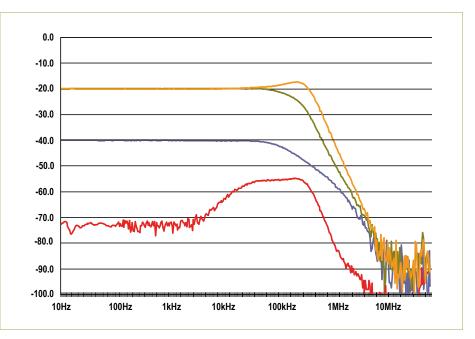


Figure 1. Frequency response at both amplification settings, each with and without the 500 kHz filter. The lower curve indicates the common mode suppression.

difference between the two input connections, whilst at the same time suppressing common-mode signals (signals of exactly the same level on each input).

It might sound as if a simple differential amplifier with voltage dividers on its inputs could do this job but life is seldom this simple. For proof just see how expensive commercial differential test probes are for this kind of measurement task. A good example is the Tektronix P5200. On the Tek website [1] you'll find an extremely readable application note [2] on using high-voltage probes. The maximum input voltage differential is given in the data sheet as ±1300 V and the bandwidth as 25 MHz. Digging deeper into the data, a value of particular interest is the CMRR or Common-Mode Rejection Ratio, which is the ability of a differential amplifier to reject the portion of the signal common to both the + and - inputs. The impressive value of 80 dB at 60 Hz drops off rapidly at higher frequencies, for instance to 50 dB at 100 kHz, which is pretty good nevertheless. A small differential signal becomes progressively more difficult to measure as the frequency of the common-mode signal rises. A frequency sweep of the commonmode signal will indicate that the output signal increases with the frequency of the common-mode signal but this is an illusion. At higher frequencies a greater part of the common-mode signal reaches the output by means of (parasitic) capacitive coupling, making it very difficult to improve the CMRR for high frequencies.

Test probes and oscilloscopes

Although the oscilloscope is a very handy instrument for taking measurements, it can easily lead you astray when you misinterpret what's indicated on the display. A whole load of measurement errors can also occur, so using a differential test probe means keeping your wits about you. The specification of our low cost, easyto-build differential test probe is shown in the panel 'Characteristics'. The switchable attenuation feature is a boon when the differential signal being measured is small and the common-mode signal is very large. Here we need to ensure that the test probe remains operating linearly and is not overdriven. An overload indicator of the

TEST & MEASUREMENT



Figure 2. Measurement cable with fully insulated croc clips.

Caution: lethal voltages!

Handling high voltages demands setting up properly, proceeding with caution and taking all necessary safety precautions — even if you're in a rush and it seems too much of a hassle. Your life is worth more than rapid results. Before testing a circuit with the type of highvoltage test probe shown here make sure you are familiar with the safety regulations. For devices operating at voltages above 50 VAC or 120 VDC testing live circuitry is allowed only if there are valid reasons why the power cannot be turned off (such as for taking voltage measurements). Under German law these operations may only be carried out by qualified electrical technicians, not by trainees and apprentices.

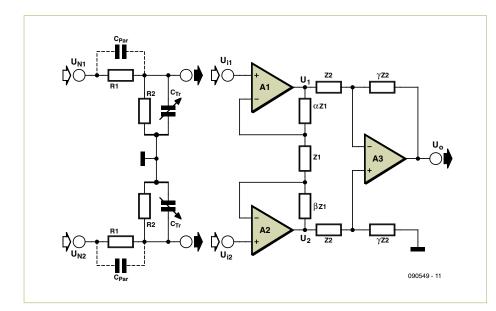


Figure 3. Block diagram with input attenuator (left) and instrumentation amplifier (right).

Input voltage division ratio: $DR_{1,2} = \frac{R_2}{R_1 + R_2}$

Output voltage (first opamp): $U_1 = (1 + \alpha + \beta) \times U_{in1} \times DR_1$

Output voltage (second opamp): $U_2 = (1 + \alpha + \beta) \times U_{in2} \times DR_2$

Output voltage (test probe): $U_{out} = (U_1 - U_2) \times \gamma$

short as practically possible.

Circuitry and input design

The circuit (**Figure 3**) is basically a differential amplifier using three op-amps, sometimes described in the literature as instrumentation amplifiers. A1 and A2 act as high-impedance inputs and amplify differentially, whilst A3 is an actual (classic) differential amplifier. Our circuit offers an interesting advantage over the classic differential amplifier, however: whilst the CMRR in a simple differential amplifier depends on the matching of the resistors Z2, it is here greater by the factor

$\gamma \times (1 + \alpha + \beta)$

This, however, creates amplification and consequently the danger of overload when a large common-mode signal is present. For this reason we must configure the amplification and input voltage divider to produce maximum CMRR in the linear operating range of the op-amp. In the circuit we selected 1 for the value of α and β , whilst γ in the two attenuator stages is 0.657 (at -40 dB) and 6.57 (at -20 dB). Z1 and Z2 are set relatively low at 1 k Ω but with the advantage that the influence of parasitic capacitance in the circuit is reduced.

The formulae for calculating the values are set out in the panel 'Formulae for Figure 3'. In your calculations ensure that the op-amp output voltages remain in the range ± 12 V.

R1 and R2 form the input voltage divider. R1 must have high resistance for a high input voltage range and low loading on the signal under measurement. Because the parasitic capacitance Cpar reduces the division ratio at high frequencies, a trimmer capacitor is fitted in parallel to R2 for compensation. We selected 10 M Ω for R1 and 51 k Ω for R2. allowing a 5-30 pF trimmer to provide compensation. The division ratio is around 198. Alignment of the trimmers is the same procedure as for other test probes, applying a square-wave signal and obtaining the optimum curve shape. Many oscilloscopes have a square-wave generator built-in for this purpose.

Construction

Figure 4 shows the construction of the

input stage. The two input resistors (for minimising leakage currents) should be insulated with heatshrink tubing. Additional ceramic capacitors are wired in parallel with the two trim-caps, to extend the compensation range.

The complete circuit (apart from the two 10 M Ω input resistors) is shown in **Figure 5**. On the left-hand side the two extra ceramic capacitors can be seen in parallel with the trimmers, also the second divider resistor. The diodes at the inputs (D11/12 and D21/22) are for over-voltage protection; they operate much faster than Zener diodes and their capacity is far lower. The high input resistors (10 M Ω) reliably clip any over-voltages from the diodes plus any current leakage from the power supply. For good frequency response it is vital to minimise parasitic capacitance at nodes in the layout.

The differential amplifier formed by U31A follows after the amplifiers U11A and U11B with their negative feedback resistors. To

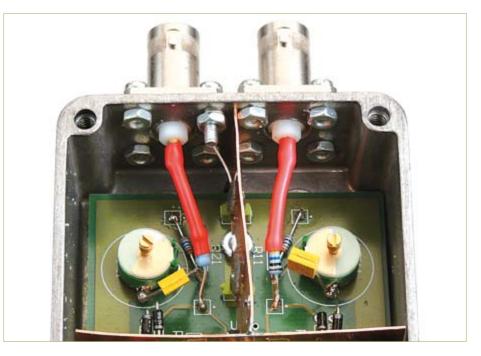


Figure 4. Constructing the input stage.

achieve the best match between amplification and CMRR, the negative feedback resistors for U31A take the form of precision trimpots. The output is terminated with a 50 Ω resistor.

Switch S1 enables the amplification factor

and the attenuation of the test probe to be toggled. S2 switches the bandwidth limiting using a 330 pF capacitor in parallel with the negative feedback resistors in the first stage. The only special feature of the power supply using voltage regulators U1 and U2 is

Calibration

Calibration is carried out as follows:

1. Connect power supply, observe red LED is lit. Check operating voltages $(\pm 15 \text{ V})$ and opamp output voltages (a few mV).

2. Connect oscilloscope to the output (output terminated in 50 Ω).

3. Set R31 and R31b in the middle of their range and connect a 1-kHz square-wave signal with an amplitude of around 10 V_{PP} to one of the two inputs – the exact value is not so important.

The square-wave signal should be visible on the oscilloscope. Adjust the trimmer capacitor on the relevant input for best squarewave curve shape. The best method is to apply the square-wave signal to trigger the second channel of the oscilloscope and use this for comparison. If the adjustment range is too small you can connect some more small ceramic capacitors in parallel with the trimmer capacitor. If the trimmer capacitor is too large and the input voltage divider behaves like a low-pass filter, you can give the voltage divider a lower impedance (e.g. using 3.3 M Ω for R1 and 16 k Ω for R2). If you find it impossible to achieve a proper square-wave signal you will need to change the physical construction (the parasitic capacity is too large).

4. Adjust the second input in the same way (square-wave signal on this input only).

5. Now adjust R31 and R31b so that the output voltage corresponds exactly to the input voltage attenuated by the factor you have set. For an input signal of $10 V_{pp}$ the output signal should be exactly $1.0 V_{pp}$ with S1 set for -20 dB and 100 mV_{pp} in the -40 dB setting.

6. Now apply the same signal to both inputs and adjust R32 and R32b to make the output signal as small as possible (the smaller the better).

7. Repeat steps 5 and 6 several times, as they are mutually interactive. Take care to achieve maximum common-mode suppression with R32/R32b — this is more important than producing the precise attenuation preset by R31 and R31b.

It is important to complete steps 3 and 4 accurately before proceeding with the remaining alignment. Without good frequency response at the input voltage divider stage all the following steps are meaningless. Signal measurement first on one and then on both inputs should be carried out repeatedly at various frequencies from DC to HF. This will also give you a good understanding of how the test probe behaves with differing signals.

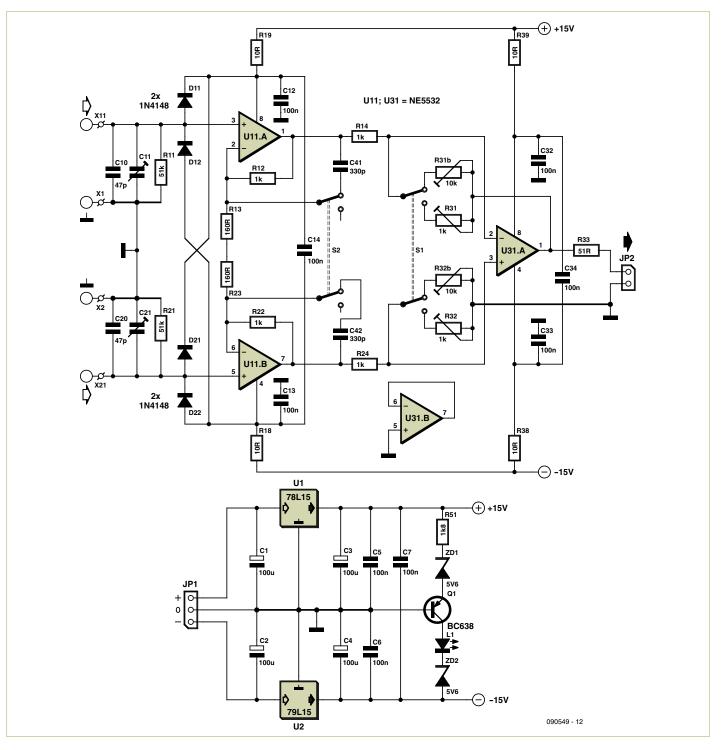


Figure 5. Full circuit (omitting the two 10 M Ω input resistors).

the little module using Q1 and LED L1 for monitoring the two voltages. Close to the opamps the voltages are decoupled once more using 10Ω and 100 nF.

The PCB layout can be downloaded from the Elektor website, where you will also find the component list.

Figure 6 shows how the circuitry is con-

structed on a PCB built into a diecast aluminium box. The input stage (left) has copper foil laid below it and shielded from the main differential amplifier on the right, so as to minimise capacitive coupling. The copper strips are soldered to the PCB using earthing (grounding) pins. Take care around the inputs to provide adequate insulation and space between the high voltage conductors. For this reason the $10 \text{ M}\Omega$ input resistors and their connecting leads should be made safe with heat-shrink tubing.

The two switches are mounted on the underside of the PCB and are operated from the lower side of the case housing. At the upper right-hand corner of the PCB are the power supply components and on the right the BNC output socket and power supply



Figure 6. Completed circuitry with the printed circuit board inside the diecast aluminium box.

connector (around ± 18 to ± 20 V).

Parasitic capacitance and inductance are minimized by the use of surface mount (SMD) resistors and capacitors. The disadvantage of the wider tolerances is offset by precise adjustment of the trimmers. Our lab sample, using 5 % tolerance components, was successfully aligned to a CMRR of more than 60 dB at low frequencies, which corresponds to a resistor tolerance of below 0.1 %.

Applications and options

Commissioning and calibration are explained step by step in the panel 'Calibration'. Remember constantly when taking measurements that high voltages are lethal! Before altering or touching anything on the test bench, switch off the volts and make sure everything is completely dead (capacitors can retain their charge for a long time!). Always work using one hand alone (keep the other one in your trouser pocket!).

The high input impedance simplifies use of the test probe and minimizes signal loading at the point of measurement. Signal level readings taken from the oscilloscope display should always take into account the attenuation level selected (-20 or -40 dB). The selected bandwidth (0.5 or 1 MHz) determines the extent to which high frequency oscillations will be visible on the oscilloscope. The input capacity of the test probe is certainly small but can nevertheless, like any capacity, lead to signal degradation or even oscillation at particularly critical points in a circuit.

Should the signal under measurement appear to be flattened or smoothed off above and below, you need to check whether the test probe is still operating linearly or is already being overdriven, because the attenuation is inadequate and/or the input signal is too large. Meaningful measurement is impossible with overloading. The effective range of the differential test probe can be adjusted by altering the division ratio, for instance to ± 100 V for lower voltages. For higher voltages the mechanical design must be changed and the insulation improved (e.g. voltage rating of the input connectors).

With a capacitor in series with R14 and R24 the DC component of the signal is removed entirely, which would enable greater amplification for measuring small signals overlaid on high DC voltages. Capacitors at this stage influence the CMRR very little. You can also add capacitors between the input divider node points and the opamp inputs but then you will need an additional resistor to ground at the opamp input for the bias current, which increases the outlay and worsens the CMRR. A capacitor in series with the 10 M Ω input resistor must possess adequate high-voltage rating, which means bulky form factor and large inductance.

Overall this differential test probe is straightforward to construct and produces a cost-effective alternative to expensive commercial products that will serve you well at frequencies up to 1 MHz. The test probe finds regular use in the author's lab and is particularly useful for audio applications.

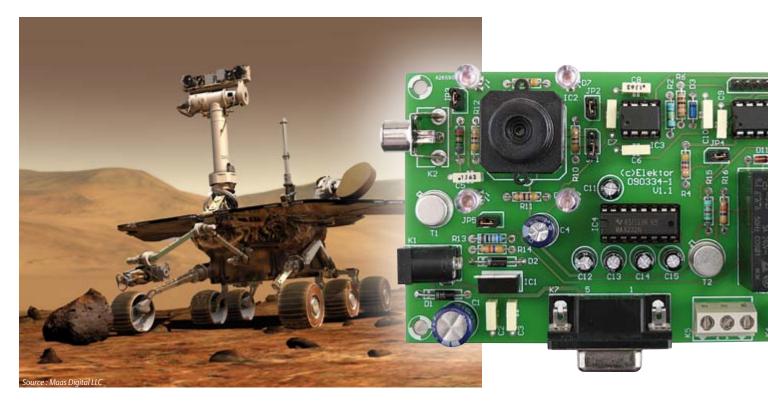
(090549)

- [1] www.tek.com
- [2] www2.tek.com/cmswpt/tidetails. lotr?ct=TI&cs=apn&ci=2343&lc=EN

The author

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Vision System for Small Microcontrollers Few pixels – clear sight



By André Thiriot (France)

The originality of the vision system described in this article lies in the method developed for acquiring video images — a method that is entirely compatible with the low calculating power of the microcontroller used and the slow sampling speed offered by its built-in analog/digital converter.

The circuit combines a microcontroller and a miniature black-&-white video camera, and acquires and processes very low resolution video images in real time. This characteristic makes it possible to use a cheap microcontroller. Very low image resolution doesn't mean mediocre performance, and in fact proves amply sufficient in applications such as:

 monitoring the presence of an object by comparing an image with a reference image stored in EEPROM

- detecting movement by comparing two successive images stored iån RAM
- recognition of simple shapes by correlation with images stored in EEPROM
- determining the co-ordinates of a bright spot (sorting and selection of the brightest group of pixels) for system position steering or servo-control

• etc.

The black-&-white video signal

The video camera is a sensor that breaks the image down into 625 lines, or to be precise,

into two interlaced fields of 312.5 lines each — an arrangement that helps reduce image flicker on the screen. Each line is scanned by an electronic 'paintbrush' that modulates the amplitude of a voltage depending on the brightness. Synchronizing signals are added to indicate the start of a new field and the start of scanning a new line (Figure 1).

The amplitude of the video signal is 1 V. The voltage range between 0 and 0.3 V is reserved for coding the synchronizing signals (line and field sync). The range between 0.3 V and 1 V allows coding of the brightness level: a voltage of 0.3 V corresponds to black, while a voltage of 1 V corresponds to white.



The usable line scan duration is 52 μ s, to which must be added the 5 μ s duration of the line sync pulse and the durations of the front and back porches (5 μ s and 2 μ s) which frame the usable signal and make it possible to delimit each line. Hence the total duration of one line is 64 μ s.

A succession of lines, separated by their synchronizing signals, form the fields, themselves separated by field synchronizing signals. These field synchronizing signals comprise a succession of line sync pulses, with

the special feature of having a different mark/space ratio, making it possible to differentiate them (Figure 2).

Out of each 312.5-line field, only 288 lines are usable (lines 23–310 for odd fields and lines 336–623 for even fields), the other lines being used for encoding the field sync pulses.

A few calculations before getting started

The operation of acquiring an image consists of sampling the video signal, with the sampling sequence synchronized to the field and line syncs.

Acquiring, storing, and processing video images require substantial resources. Let's take as an example a VGA standard image (480 lines of 640 pixels) produced from a video signal. Capturing 640 pixels in 52 μ s implies a sampling frequency of 640 / 52 μ s, or over 12 MHz.

Storing a 16-color VGA image coded using 4 bits requires memory space of 640 × 480 × 4 = 1228800 bits, or 153,600 bytes. And finally, digital image processing in real time requires a high calculation speed. Thus, in order to perform a simple comparison between two VGA images (pixel-wise subtraction), we need to carry out 640 \times 480 = 307,200 operations. If we want to perform this processing in less than 10 ms, for example, this implies a basic calculation speed of 10 ms / 307,200, or 33 ns

An image processing system normally requires resources that a simple microcontroller is unable to offer. However, if we can

Technical Characteristics

- PIC16F690 microcontroller
- 18 × 16 pixel B&W video image
- 4 gray levels
- LED illumination
- Relay control
- Image comparison
- Movement detection
- Serial link
- Constructional difficulty: medium

make do with lower performance, a microcontroller like the PIC16F690 will do. Running at 8 MHz, it offers an instruction cycle time of $0.5 \,\mu$ s, and can perform an A/D conversion in 24 μ s. It has 256 bytes of RAM, 256 bytes of EEPROM, and 4,096 words of

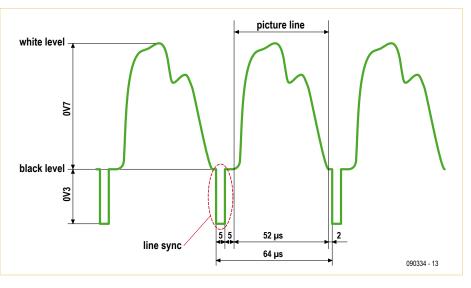


Figure 1. The B&W video signal.

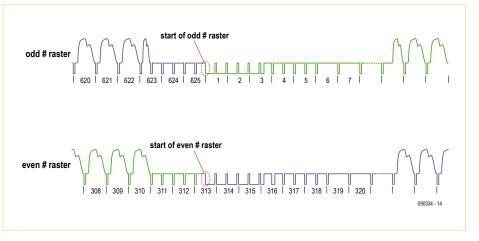


Figure 2. The difference between two lines.

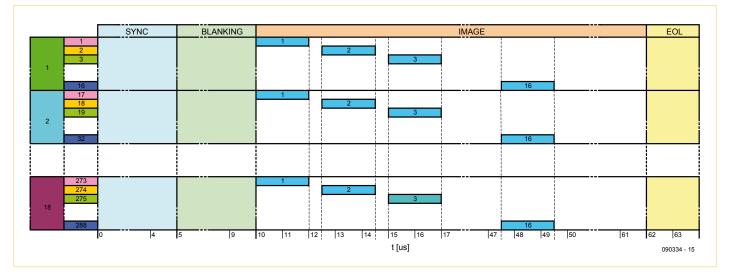


Figure 3. Sampling timing diagram. Each cell corresponds to one of the points in the image. The light blue cells symbolize the duration of the analogue/digital conversion and thus represent the minimum time to be left between two successive samples. The under-sampled grid that will form the very low resolution image can be made out.

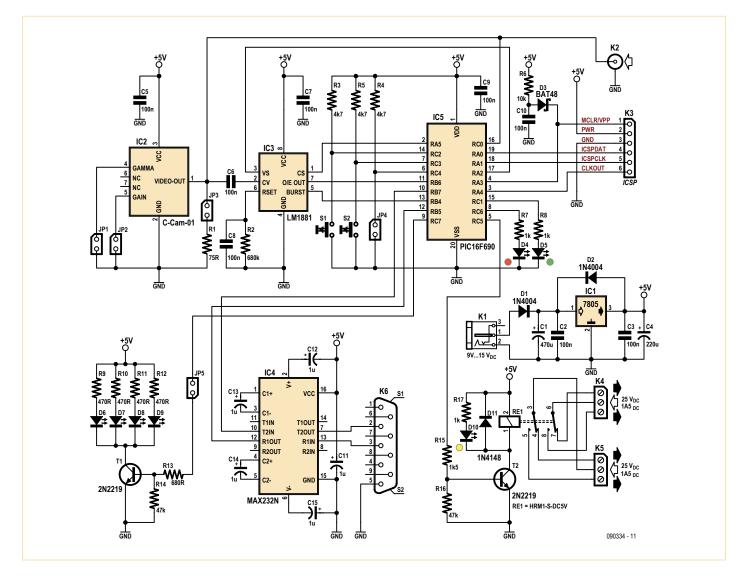


Figure 4. The circuit diagram of the project. A few suitable ICs and there you have it!

flash memory. Given these characteristics, we can see that the images are going to have to be of modest size for us to be able to process them using this microcontroller.

Suggested method

The 24 μ s conversion time means sampling at a frequency of 12 MHz is not possible. However, this time is of the same order of magnitude as the 64 μ s line period. So out of this simple observation comes the idea of acquiring only a single point per line, regularly offsetting the sampling moment with respect to the start of the line. This under-sampling then makes it possible to form a very low resolution image, compatible with the resources available to the microcontroller.

Sampling one point per line implies a number of points in the image less than or equal to the number of lines, i.e. 288 pixels. An image of 18 lines of 16 points ($18 \times 16 =$ 288) lets us maintain roughly the same proportions between image width and height (Figure 3).

The memory space required to store the image depends on the number of pixels, and also on the pixel brightness resolution. Using 2 bits to code the brightness level allows us to distinguish four grey lev-

els, which are enough for the applications considered. Thus an image will occupy 2 × 288 = 576 bits, i.e. 72 bytes. This small size allows three images to be stored in the RAM memory and another three in the FEPROM.

Circuit description

The circuit (Figure 4) is based around the PIC16F690 microcontroller (IC5) from Microchip [2], chosen for its low cost and the very affordable price of the Starter Kit PICkit 2 (or 3) used for developing the software. The 6-way connector K3 is provided for connecting up the development or microcontroller programming tool. Diode D3 protects the circuit against the programming voltage Vpp of 13 V. Before connecting up the programmer, uncheck the 'Power target circuit from MPLAB ICD 2' option (Figure 5) in the programmer configuration. This lets you power the circuit normally and use the programmer at the same time.

The image detector IC2 is a miniature B&W CMOS camera, powered from 5 V. The signal produced conforms to the description given above. Video signal acquisition is performed by the microcontroller's internal ADC via pin 16 (AN4).

Jumper JP1 lets us enable gamma correction, while JP2 lets us select the camera's image amplification gain.

The phono socket (K2) lets us connect a video monitor so as to display the image produced by the camera, or to connect an external camera: this latter option makes it possible to evaluate the circuit very cheaply, by saving the cost of the CMOS camera, which in this case will not be fitted to the PCB. When K2 is not used, JP3 must be fitted in order to load the camera video output with a 75 Ω resistor.

IC3, a classic LM1881 [3], in conjunction with the network R2 / C8, has the task of extracting the field (VS) and line (CS) syncs from the video signal, along with the parity signal (O/E OUT) used by the microcontroller to synchronize the sampling data of the video image. These signals are connected to microcontroller ports RA2, RA5, and RB6 respectively. LEDs D6–D9, driven by port RC7 (if JP5 is fitted), make it possi-

Movement detection by image comparison

ble to illuminate the zone where the image is taken.

Advantage is taken of the asynchronous series interface available in the microcontroller to send the image information via a serial link. IC4 takes care of the electrical conversion to the RS-232 standard.

Relay RE1 offers the possibility of controlling an external system — for example, an alarm. It is driven by port RC5. LED D10 tells us if the relay is energized or not.

The circuit is powered from a mains power supply (9–15 V / 200 mA), connected to connector K1. The power from the PSU is



Figure 5. The programmer parameters. Uncheck the 'Power target circuit from MPLAB ICD 2' box.

smoothed then regulated by linear voltage regulator IC1.

Software

Using a microcontroller means there's bound to be software. For performance reasons, the software has been written in assembler throughout. The detail of the

> functions (parameters and procedures) is given in the program source code, which contains plentiful comments, available for free download from [1].

> The main functions of the software are broken down into macros and subrou-

tines: the use of subroutines was favoured in order to save program memory, but the use of macros (see Table 1) proved necessary where the execution time turned out to be critical, in particular during the image acquisition phase, where the grey level coding steps take a bit of time.

The video signal sampling and pixel grey level coding steps have been optimized: the grey level of each pixel is coded while the ADC is converting the next pixel, which saves the 24 μ s wait time of the conversion operation.

The basic software illustrates the possibili-

AUDIO, VIDEO & PHOTOGRAPHY

Table 1. List of functions available for managing the vision system.				
Туре	Name	Function		
Macro	MACRO_LineSync	Wait for line sync		
Macro	MACRO_DPixelAcquisition2Bits	Launches acquisition of pixel n and grey-level conversion of pixel n-1		
Function	Start	Main program comprising the initialization phase and the primary loop		
Function	BinASCII	Decimal to ASCII conversion: returns the ASCII codes of the three digits of the decimal value to be converted (hundreds, tens, units)		
Function	Bin2Colour2Bits	Converts a grey level $(0-3)$ into the ASCII code used to represent it $(32, 176, 177, 178)$		
Function	CharTrans	Transmits a character over the serial link		
Function	PictureRamToEeprom	Copies the image stored in RAM to the EEPROM		
Function	PictureEepromToRam	Copies the image stored in EEPROM to the RAM		
Function	PictureCompare	Compares the two images stored in RAM and in EEPROM		
Function	PixelGroupDistCalc	Calculates the distance (absolute value of difference in brightness level) between four groups of two pixels		
Function	PictureAcquisition	Acquires an image of 288 points organized as 18 lines of 16 pixels coded using four gray levels		
Function	DecimalValueTransmission	Transmits the three ASCII codes representing the three digits of a decimal value over the serial link		
Function	PictureTransmissionBIT	Transmits the image stored in RAM over the serial link, with each pixel represented by the ASCII character corresponding to its gray level.		

COMPONENT LIST

Resistors (5%, 0.25W)

R1 = 75Ω R2 = $680k\Omega$ R3,R4,R5 = $4.7k\Omega$ R6 = $10k\Omega$ R7, R8, R17 = $1k\Omega$ R9-R12 = 470Ω R13 = 680Ω R14, R16 = $47k\Omega$ R15 = $1.5k\Omega$

Capacitors

- C1 = 470µF 25V radial electrolytic, 10mm diam.
- C2,C3,C5–C10 = 100nF, polyester (MKT), lead pitch 0.2 in.
- C11–C15 = 1µF 16V, radial electrolytic, 5mm diam.
- C4 = 220µF 25V, radial electrolytic, 8mm diam.

Semiconductor

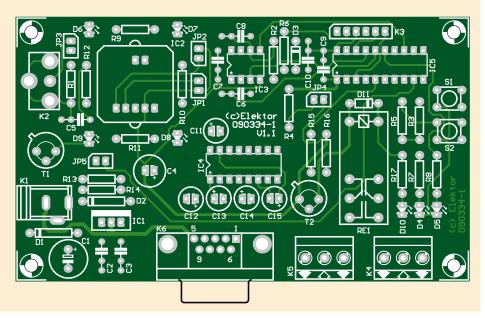
- D1, D2 = 1N4004
- D3 = BAT48, Schottky diode, DO-35 case
- D4 = LED, red, low current, 3mm
- D5 = LED, green, low current, 3mm D10 = LED, yellow, low current, 3mm
- D6–D9 = infrared LED, QED222 (e.g. Farnell #
- 1652526)
- D11 = 1N4148
- IC1 = 7805, TO-220 case
- IC2 = C-MOS IR camera module (CCIR*) e.g. Conrad Electronics # C-Cam-01 150001
- IC3 = LM1881N/NOPB, DIP-8 case (e.g. Farnell # 1564700)

- IC4 = MAX232N, PDIP-16 case IC5 = PIC16F690-I/P, DIP-20 case (e.g. Farnell
- IC5 = PIC16F690-I/P, DIP-20 case (e.g. Farnell # 1103406)

Miscellaneous

- RE1 = Relay, Multicomp type HRM1-S DC5V (e.g. Farnell # 9479937)
- JP1–JP5 = 2-pin pinheader with jumper, 0.1 in. lead pitch
- K1 = power supply connector, 2.1mm, horizontal, PCB mount
- K2 = RCA (phono) connector, PCB mount

- K3 = 6-pin pinheader, 0.1 in. lead pitch K4, K5 = 3-way PCB screw terminal block,
- 5mm lead pitch
- K6 = 9 way sub-D socket (female), horizontal, PCB mount
- S1,S2 = pushbutton, e.g. Tyco Electronics type FSM4JH (Farnell # 1555982)
- Sockets for IC3, IC4 and IC5 (8-pin, 16-pin, 20pin respectively)
- PCB, Elektor # 090334-1 (see [1])
- * substitute NTSC version where applicable.



ties offered by the vision system. This software performs the following operations:

- acquires and stores a 288-pixel image in RAM
- transmits the image in ASCII format over the serial link
- compares the image stored in RAM with the image stored in EEPROM
- sends back the result of the image comparison over the serial link
- controls the lighting LEDs

The two images stored in RAM and in EEPROM are compared pixel by pixel. The comparison method used consists in accumulating the differences in the pixel brightness values. The sum total of these differences is then compared with a threshold in order to decide whether to set off an alarm or not.

The image is sent back over the serial link in ASCII format (19,200 baud, 1 start bit, 8 data bits, no parity, 1 stop bit) so that a simple terminal (like a HyperTerminal) can be used to view the image (Figure 6). The four grey levels that each pixel can have are coded using the four extended ASCII codes with values 32, 176, 177, and 178.

Pressing button S2 lets us store the latest image acquired and stored in RAM into the EEPROM; pressing button S1 lets us send the image stored in EEPROM back up the serial link. LED D5 flashes whenever either of the buttons is pressed. LED D4 indicates the start and end of image acquisition. D10 lights when the relay is energized.

JP4 allows the software to use the lighting LEDs. JP5 allows the hardware to use the lighting LEDs — the difference is a subtle one.

Conclusion

Despite its modest features, this circuit opens the door to a great deal of experimentation and even practical applications. For example, the author experimented with the object presence detecting software while looking for a way of detecting the presence of letters and packets in a letter-box, rather than the event of their being put into the box. To achieve this, a patterned background (checkerboard or concentric squares) is placed at the back of the letter-box, monitored at regular intervals by the vision system mounted in the top of the box. The image is compared with a reference image stored in EEPROM when the system is installed. There's no daylight in the letter-box, so the inside is illuminated by the circuit's LEDs while a new image is being acquired. Inserting a letter or packet modifies the image seen by the system, setting off an alarm which continues for as long as the object inserted remains in the box. This alarm goes off when the post is collected and the box is empty again. Now it's your turn to dream up other applications, which you can share with Elektor

(090334)

readers!

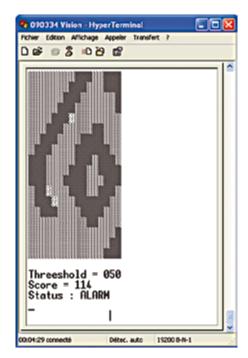


Figure 6. The video image in HyperTerminal. Can you guess what it's of?

(.Aran e nithin a larger square)

Internet Links

- [1] www.elektor.com/090334
- [2] ww1.microchip.com/downloads/en/ DeviceDoc/41262E.pdf
- [3] www.national.com/ds/LM/LM1881.pdf

ROOMBA

Make Room for the Roomba! Elektor Teardown Session

By Wisse Hettinga (Elektor Editorial)

Without question, one of the duties we enjoy the most at Elektor is organizing teardown parties, where the only objective is to take something apart. There's nothing more fun or exciting than attacking a piece of equipment (preferably something expensive) with a screwdriver and a hobby knife, and with the vague expectation that we we'll be able put it all back together afterward. This time we wanted to have a closer look at a robot vacuum cleaner, and one of our colleagues in Customer Service just happened to have one. Yolanda! How about loaning us your hoover for a little experiment?

Fortunately, she was willing to sacrifice her Roomba to the cause of science, and – believe it or not – she was even excited at the prospect. With a good supply of beverages and a keen photographer on hand, we were able to document the whole process for posterity. The iRobot Roomba is around 1 foot in diameter and stands 4 inches

high. It (or should we say 'she', since it moves so gracefully?) travels around the room sucking up everything in its path. It comes complete with a charging unit, and it has an extra device you can use to create a virtual wall. With its bumper, it automatically detects any obstacles in its path, and when it



encounters an obstacle it takes a different route.

After finishing the job (all fully configurable), it returns to its charging station. It has sensors to make sure it doesn't fall down the stairs. If you turn it upside down, you can see that it's fitted with a roller brush, along with a small rotary brush (side brush) for the corners. Finally, there's an optical sensor on top of the Roomba that enables it to dock smoothly with its base station.

Teardown

When teardown time comes, there's only person on the team who can wield a screwdriver with no sign of hesitation: Jan. Even the threat of docking his pay to cover any resulting damage doesn't hold him back. Now we can see what's inside: a battery (3 A, 12 cells, 15.57 V) and sensor contacts, the spring-mounted wheels with their own motor (RS 540), a bunch of screws and bushes, and – of course – lots of dust. The front (steering) wheel has a black and white pattern, which makes it easy to see whether the Roomba







ROOMBA





is moving. We also see another four IR sensors that help keep the Roomba on the right path.

Some of the dust manages to get past the filter. Apparently it's a good idea to clean the Roomba from time to time.

Now we come to the electronics. There's a sizeable PCB, with as its most prominent features a WiFi daughterboard with a Freescale MC13202r, an ST ARM7 master microcontroller that looks after all the operating functions, and a connector for updating the software. Of course, there's also a 'kinda' control panel

with LED display for setting the time and other functions. In order to maintain a good working relationship with its owner, the Roomba can report faults and problems in various languages. Now we have a big pile of parts, and it's time for the miracle: putting it all back together and make Yolanda happy again.

All's well that ends well, as they say.

With its update connector, the Roomba naturally lends itself to updating or reconfiguring the software. You can find all sorts of tweaks and hacks on the Web that let you use the Roomba for other purposes, such as a musical instrument. You can even turn it into a fighting machine, as illustrated at

- http://roomba.pbwiki.com

- www.makezine.com/blog/archive/2006/03/videos_from_roomba_fights.html - http://todbot.com/blog/2006/05/03/roombamidi-roomba-as-midi-instrument

(100015-l)





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Test Equipment

By Gerard Fonte (USA)

If we were all as rich as Bill Gates we'd each have home laboratories with all the equipment we could imagine. Unfortunately, most of us (including me) never have enough money for half of the things we really need. Here are some tips for stretching that dollar.

Play it Again

Used equipment doesn't mean used up. Sure, it's exciting to open up a box and get that great smell of new plastic. But you're paying real money for that. Used test equipment, like used cars, can give you a much better price/performance value. But, like used cars, you have to take some steps to prevent a sour deal. To achieve the best bargain it's important to know two things: 1) what you're buying and 2) who's the seller. (One additional point: look to the future. It's cheaper to buy more performance than you need now, than it is to buy another piece in the future.)

Most used equipment simply gives the manufacturer's model number and brief description. For example: "Tektronix model 2465, 300 MHz Oscilloscope". You will need to do some research to find out other important features (like if it has two channels, delayed sweep, etc.). However, this is fairly easy to do with the aid of the internet. It is important to realize that slight differences in model number can have a huge difference in performance. So, always be sure of exactly what you are buying.

Who you buy it from matters, too. There are some scrap dealers which will give you a great deal. But there is no return policy and the seller may not really know if the unit really works at all. The best that they can do is turn it on and verify that the fuse doesn't blow. These deals are a gamble, but the price is great. If you have the time and skills to repair it, if necessary, this may be a good choice for you. But for most people, it's a bad bet.

Buying from individuals is questionable. If the person actually used the equipment you have a better chance of determining its quality. And, you will speak the same language. But, you must remember that many individuals (like used car dealers) buy equipment to sell at a profit. They may not be technically adept. They can determine if the device functions properly. And they usually have some sort of return policy.

Then there are companies that resell used equipment. They too, buy equipment to sell at a profit. However, they have a substantial infrastructure to support them. They can test the equipment and verify it works. Some can "calibrate to factory specifications". This means that the piece functions like new. Obviously, this is the thing to look for, and just as obviously, it costs a bit more. But usually, it's worth it. (It's never good to doubt your test equipment.). This is probably the best choice for hobbyists. The prices are about 30% to 40% of the original price but there is a return policy and the sellers have to sell quality products in order to stay in business. However, as always, it's buyer beware.

The last group are companies that lease equipment. After a lease of a year or so, the device is paid off and they sell it. Unfortunately, there's only about a 20% discount from original price. Often the original warranty is still in effect and, of course, it works like new. This makes more sense for businesses than for enthusiasts. One important note. Always (that's ALWAYS!) get a service manual when you buy your equipment. If something goes bad in a few years you will have to fix it yourself (or spend a lot of money you don't have, to get it repaired). Sam's PhotoFacts used to



be the main source for these manu-

als. But, again, the internet is making things easier. Some manufacturers will provide them as free downloads. Alternatively, there are various companies that sell CD's for about \$20 (for something complicated, like an oscilloscope).

What's in a Name

You should stick with the major brands (Hewlett/Packard, Tektronix, etc) rather than off brands. There are two basic reasons for this. The first is that name brands are quality-built and last a long time. (I have a 30 year old H/P Spectrum Analyzer that still works great.) Old Heathkit and Eico instruments were usually assembled by the buyer. So, their quality is extremely variable. They should be avoided. The second reason is parts replacement. You will have a much better chance of locating some special part if it was made by H/P rather than Dyno-Dyne. Clearly, having a frugal unit is no bargain if it can't be make to work.

Who's Your Daddy

You can always build your own test equipment. While you may not be able to construct your own 500 MHz oscilloscope, you may be able to build your own 2-GHz RF Spectrum Analyzer. Check out the "Poor Man's Spectrum Analyzer" (www.science-workshop.com). It uses TV tuners. I assembled mine in a weekend without needing to know much about RF techniques (a long, long time ago in a galaxy far, far away). While it's not a great instrument (even compared to my old H/P) it certainly may be good enough for basic testing. The starting price for the kit is around \$200.

Building your own test equipment gives you great construction experience that transfers to many other areas. There are lots of new ICs from the major houses (Analog Devices, Maxim, National Semiconductor, Linear Technology) that provide the basic core functions. You probably have to add a controller (read micro-computer) and some signal conditioning. But, if you're reading this, that's probably not a huge task. For example there's the Analog AD5934 that measures the complex impedance (real and imaginary parts) for under \$20.00 in unit quantities. Look around, there are a lot of inexpensive and amazing parts available.

There are a many more options than paying full price for your test equipment. Putting in a little up-front research can keep some money in your wallet.

(100548)

Hexadoku Puzzle with an electronics touch

No matter if your summer was hot and dry, a cool head is required to solve this new edition of our monthly brain teaser. Enter the right numbers in the puzzle, send the ones in the gray boxes to us and you automatically enter the prize draw for four Elektor Shop vouchers. Have fun!

The instructions for this puzzle are straightforward. Fully geared to electronics fans and programmers, the Hexadoku puzzle employs the hexadecimal range 0 through F. In the diagram composed of 16 × 16 boxes, enter numbers such that **all** hexadecimal numbers 0 through F (that's 0-9 and A-F) occur once only in each row, once

Solve Hexadoku and win!

Correct solutions received from the entire Elektor readership automatically enter a prize draw for one Elektor Shop voucher worth \$140.00* and three Elektor Shop Vouchers worth \$70.00* each, which should encourage all Elektor readers to participate.

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in each column and in each of the 4×4 boxes (marked by the thicker black lines). A number of clues are given in the puzzle and these determine the start situation. Correct entries received enter a draw for a main prize and three lesser prizes. All you need to do is send us the numbers in the grey boxes.

Participate!

Before October 1, 2010, send your solution (the numbers in the gray boxes) by email, fax or post to Elektor Hexadoku – 4 Park Road – Vernon CT 06666 USA. Fax 860 8751-0411 Email: hexadoku@elektor.com

Prize winners

The solution of the June 2010 Hexadoku is: 6B310.

The \$140.00 voucher has been awarded to: Zigor Gómez Arias (Spain).

The \$70.00 vouchers have been awarded to: Erik Jansen (Netherlands), Taina Paavilainen (Netherlands) and William Fackrell (USA). Congratulations everyone!

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В	3	9	F	2	4	С	Е	D	0	5	8	7	А	6	1
D	8	1	2	7	F	0	А	4	Е	9	6	С	В	5	3
С	А	6	0	D	5	9	3	В	F	1	7	8	Е	2	4
7	4	5	Е	6	8	В	1	А	3	С	2	F	9	D	0
А	5	8	7	F	D	6	В	3	1	0	С	Е	4	9	2
2	В	Е	9	1	7	3	С	6	4	D	F	0	5	8	А
6	D	0	3	А	Е	4	5	2	8	7	9	1	F	В	С
1	F	4	С	9	0	2	8	Е	А	В	5	D	3	7	6
4	6	А	8	3	В	D	7	1	9	Е	0	2	С	F	5
3	7	С	В	4	9	8	F	5	2	А	D	6	1	0	Е
9	0	D	1	5	6	Е	2	F	С	8	4	3	7	А	В
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RETRONICS

Delay-line Digital Memory (ca. 1968)

By Jean Herman (Belgium)

In November 1968, the Belgian company MBLE (Manufacture Belge de Lampes et de Matériel Electronique) published an article in its "Special new techniques" magazine presenting a new device: a delay line capable of memorizing digital information for logic and calculation. The device was similar to the delay lines used in TV color decoders, but complete with all the electronics needed to make it work. This was not a new discovery, as digital computers were already using bulky delay lines that employed a column of mer-

cury or any other substance intended to delay information. But MBLE's innovation was to have made it into a reliable, stable, and quite accurate device, in a standard, interchangeable module. Three modules were available: the GDM11 with a capacity of 256 bits, a binary data rate of 0.5 MHz and a delay of 515 µs; the GDM12 (256 bits, 0.5 or 4 MHz, 515 or 64.5 µs); and the GDM21 (256 bits, 4 MHz, 64.5 µs). The GDM11 and GDM21 modules were master mod-

ules, the GDM12 was a slave module. To increase memory capacity, several modules could be used in series. This was the simplest solution, but in practice, one soon found oneself limited by

the accuracy of each delay line. In series, the

delay time error is cumulative. So the commonest method was to use the modules in parallel, where there is no limit to the capacity. To achieve perfect synchronization of the modules, they used a common clock pulse from the single master module, making it possible to use just one bit-rate frequency converter for all the modules. This type of memory has been used in calculating machines or small computers, as an analog/digital converter buffer, in digital machine tools, as a screen memory for CRT displays, etc.

The material used for transmitting and holding the information is

ARRANCING MODULE

a special type of glass that has a high lead content, to obtain the lowest possible acoustic (ultrasonic) propagation speed. Two ultrasonic transducers are bonded to the same 40 mm end of an $80 \times 40 \times 8$ mm oblong slab of glass. Under the transducers, the glass is angled at 7.5° so that the transducers are perpendicular to the reflection paths. One piezoelectric transducer emits pulses and the other receives them.

Internal reflection takes place between the two 40 mm ends of the slab of glass, so the distance between the transducers is 2×80 mm = 160 mm. An ultra-

sonic bit travels this distance in 64.5 μ s. Calculating back from this, the sound waves travel at a speed of 0.16 m/64.5 μ s = 2,480.6 m/s.

Obviously, the long 160 mm path through the glass attenuates the signals. The voltage attenuation measured between the piezo transducers is around 6 dB. A BSX20 transistor amplifies the echo received and brings it up to TTL level.

The 256-bit word is continuously re-injected into the glass and is controlled by memory management logic. The signal attenuation is an advantage, since the initial pulse is propagated

in several ways throughout the whole of the glass slab, but is no lon-

ger of sufficient amplitude to interfere with the working of the system.

In reality, the emitting transducer generates a longitudinal wave perpendicular to the

transducer, but also a transverse wave at a different speed (around half the speed) which might interfere with the main wave

In 1968, one had to be very sparing with the precious memory capacity. I have worked with large computers that managed a whole great rolling mill and yet only used 16 KB of memory! I have also had occasion to repair Schneider calculating machines that used this delay line.

(100081-l)

Retronics is a monthly column covering vintage electronics including legendary Elektor designs. Contributions, suggestions and requests are welcomed; please send an email to editor@elektor.com



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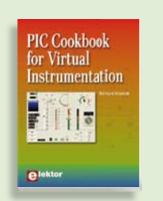
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ISBN 978-0-905705-76-7 • \$100.00



The Elektor DSP radio (July/August 2010)

Many radio amateurs in practice use two receivers, one portable and the other a fixed receiver with a PC control facility. The Elektor DSP radio can operate in either capacity, with a USB interface giving the option of PC control. An additional feature of the USB interface is that it can be used as the source of power for the receiver, the audio output being connected to the PC's powered speakers. To allow portable 6 V battery operation the circuit also provides for an audio amplifier with one or two loudspeakers.

PCB, assembled and tested

Art.# 100126-91 • \$240.40

InterSceptre (June 2010)

In our March issue, we introduced Sceptre, a fast prototyping system fitted with a 32bit microcontroller. Even on its own, this board will let you produce some great results, but if we add an extension board to make it easier to access all its peripherals, the Sceptre platform becomes downright powerful. What's more, if you fit this extension board into a suitable case, you'll be able right from the start to develop a prototype that you can use 'properly' in a installation, with no trailing wires or bits of sticky tape holding everything together. Now that's what you call fast, convenient prototyping!

Kit of parts, contains PCB and components

Art.# 100174-71 • \$187.10



dsPIC Control Board (May 2010)

This control board has been designed for incorporation into typical industrial electronics applications like controlling motors or adjustment of static up- or down-converters. The objectives were to obtain a board with a large number of pulsewidth modulation (PWM) generators, which enables us to control several motors and static converters at the same time. The cost of the control board needed to be as low as possible too. In addition, it must be possible to construct the board using a soldering iron, without requiring use of a reflow oven.

PCB, populated and tested

Art.# 090073-91 • \$225.90



Reign with the Sceptre (March 2010)

This open-source & open-hardware project aims to be more than just a little board with a big microcontroller and a few useful peripherals — it seeks to be a fast prototyping system. To justify this title, in addition to a very useful little board, we also need user-friendly development tools and libraries that allow fast implementation of the board's peripherals. Ambitious? Maybe, but nothing should deter you from becoming Master of Embedded Systems Universe with the help of the Elektor Sceptre.

PCB, populated and tested, test software loaded

Art.# 090559-91 • \$143.60

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Current Clamps Compared

Although a multimeter is an extremely useful and versatile instrument, the range of the ammeter section usually ends at about 10 A. For larger currents you should use a so-called clamp that can easily handle 100 A or even 1000 A. In the October 2010 edition we describe our experiences with twenty current clamp meters from different price ranges, varying from tens of dollars to about 500 dollars.

Photo Timer

Many modern digital cameras allow all features to be controlled remotely. Using the extensive photo timer published in the October 2010 edition, accurately timed photos can be made with Canon EOS or compatible cameras. A 4-line display shows all settings, which are also stored in an EEPROM. Optional pre-focusing is available and sound effects announce the actual 'shot'.

NE5532 Power Amplifier



In this article Elektor's 'off the beaten track' approach to electronics flourishes. Sure, the NE5532 opamp is widely used in audio applications but it's less well known that an interesting power amplifier can be made by connecting enough of them in parallel. For this project 32 NE5532s are bundled to make a quality amplifier with an output power of about 15 watts into 8 ohms. The first installment of the article describes the design philosophy and the schematics of this remarkable audio power amp.

Article titles and magazine contents subject to change; please check the Magazine tab on www.elektor.com Elektor UK/European edition: on sale September 23, 2010. Elektor USA edition: published September 10, 2010.

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