CESSO CESSO CENTRA CANVILLE AND FRAN GRANVILLE AND FRAN GRANVILLE READERS SOLVE DESIGN PROBLEMS

Create a swept-sine function in LabView with just one virtual instrument

Sean McPeak, University of California-San Diego

Swept sine waves are useful when you want to test a product over a wide frequency range. A large research project included the requirement to determine wave propagation in the open ocean. This application required the generation of a swept sine wave to drive an acoustic transducer. Although many waveform generators have a built-in function for this requirement, you must program it yourself if you want to implement a swept sine with a multifunction data-acquisition card. You can create a sweptsine function in National Instruments' (www.ni.com) LabView with just one VI (virtual instrument). Using this function, you can control start and stop frequencies, sample rate, and the overall duration of the sweep (Figure 1).

The LabView software calculates an array of numbers that represent the swept-sine-wave time series at each sample point as the frequency either increases or decreases, depending on the direction of the sweep. You must handle the frequency change of the output on a point-by-point basis. The basic form of the equation is $Y(I) = V \times sin((A \times I^2)/2 + B \times I)$, where Y(I) is the amplitude of the swept sine wave as a function of the sample point, I is the integer that steps through the time series, V is the peak voltage, and A and B are variables. You define A as $2 \times \pi(f_{\text{STOP}} - f_{\text{START}})/N$, and you define B as $2 \times \pi f_{\text{START}}$ where N is the number of samples, f_{START} is the normalized start frequency, and f_{STOP} is the normalized stop frequency. To normalize the start

DIs Inside

- 44 Charlieplexing at high duty cycle
- 48 Serial port tests digital circuits
- 51 DAC calibrates 4- to 20-mA output current
- 51 Alarm tells you to close the refrigerator door
- ► To see all of *EDN*'s Design Ideas, visit www.edn.com/design ideas.

and stop frequencies, you must change the unit to cycles per sample. You accomplish this task by dividing the f_1 and f_2 frequencies in hertz by the sample rate. You determine the sample rate by deciding how smooth of a transition you want to represent your swept sine wave. A good rule of thumb is to have at least 10 samples/cycle at the high-

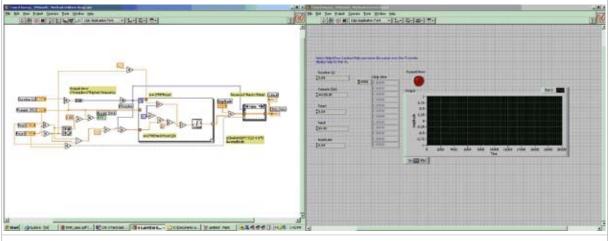


Figure 1 With just one LabView virtual instrument, you can control start and stop frequencies, sample rate, and the overall duration of the sweep.

est frequency. When setting the sample rate, you need to take into account the overall frequency span you are sweeping and the duration of the sweep itself. It is also helpful to compare the results and performance of the LabView dataacquisition-system implementation of the swept sine wave with those of an AWG (arbitrary-waveform generator).

You use two methods of comparison. First, you compare the output of both the data-acquisition and the AWG swept sine wave on a spectrum analyzer. Second, you run them both through an audio-amplifier/speaker system and simply listen to the output. This method is useful in determining sweep rate, duration, and stop and start frequencies. This type of comparison is valid only if the frequencies involved are in the audible range. The LabView VI employs simple array manipulation and uses a "for" loop. The input duration is in seconds, the sample frequency is in samples per second, and the starting and ending frequencies are in hertz. Dividing the sample rate immediately converts the start and end frequencies to cycles per sample. A maximum/minimum block takes the normalized ending and starting frequencies as its inputs and uses the maximum output of the input pair. You use this method to determine whether your design meets the

Nyquist criteria, given the sample rate and highest frequency you require.

This approach drives a simple Boolean variable to alert the user about whether the design meets the Nyquist criteria. You set the "for" loop to run for the total number of samples you want to calculate. You determine this value by multiplying the duration in seconds by the sample rate in samples per second. To guarantee that the loop processes all of the generated samples, you must add one, because the loop stops at N-1.

You implement the output function in the "for" loop with simple algebraic operators and the sine block. The output is an array that reaches the perimeter of the "for" loop. It is important to enable indexing at this node. This action allows the circuit to individually handle each element in the array at the output of the "for" loop. You can also add a simple gain stage to set the peak-to-peak value to whatever point you want. Finally, you use the "rotate-1D-array-block" case structure to flip the array if the ending frequency is lower than the starting frequency. This approach handles cases in which you want a frequency sweep that starts out in a higher frequency and descends to a lower frequency.

You can easily modify and expand

this simple program. One idea would be to use the output array, which is nothing more than the time series representing a predescribed frequency sweep, to feed a loop that would drive a data-acquisition module. The output of the module should accurately represent the frequency sweep, as long as the module's output sample rate is the same as the sample rate you use for generating the frequency-sweep time series. You should then be able to track the output samples and, when they are complete, reverse the frequency-sweep array. You then again feed this new flipped array to the data-acquisition module. Depending on the maximum and minimum frequencies, sweep duration, and sample rate you use, it may be difficult to flip the array and configure the module quickly enough to not miss a sample. In that case, you can prefill a frequency-sweep array for a set number of passes.

These modifications allow the sweep to continue back and forth for a set period. Another improvement would be to add some real-time FFT (fast-Fourier-transform) capability so that the user can see the sweep in the frequency domain. This approach also adds an increased level of insurance that the circuit properly meets the sweep definition.EDN

Charlieplexing at high duty cycle

Luke Sangalli, Digital Designs, Melbourne, Australia

A few articles have recently appeared describing novel ways to increase the number of LEDs a microprocessor can drive with a limited number of pins (**Reference** 1). The standard multiplexing technique made popular by multidigit seven-segment displays has, in pin-scarce designs, given way to "Charlieplexing."

Charlie Allen devised this technique while working at Maxim (www.maxim-ic.com), which has since introduced

TABLE 1 NO. OF LEDs AND DUTY CYCLES				
No. of pins	No. of LEDs	Charlieplexing duty cycle (%)	Standard multiplexing duty cycle (%)/no. of pins	
Three	Six	33.3	50/five	
Four	12	25	33/seven	
Five	20	20	25/nine	
Six	30	16.6	20/11	
Seven	42	14.2	16.6/13	
Eight	56	12.5	14.2/15	
Nine	72	11.1	12.5/17	
10	90	10	11.1/19	

ICs using the technique (**Reference 2**). Allen used the high-impedance third input state available to most microprocessors for turning off LEDs in a

matrix; the LEDs' respective microprocessor pins' high or low states individually turn on these LEDs. Using this method, you can drive nine seven-segment LED displays using only nine microprocessor pins rather than the usual 17. For N pins, you can individually address $N \times (N-1)$ LEDs using Charlieplexing.

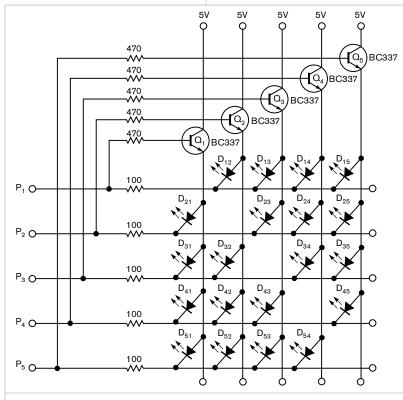
One of the gripes people often level at Charlieplexing regards its poor duty cycle. A previous Design Idea com-

pares the standard multiplexing method with Charlieplexing (Reference 3). Using Charlieplexing, the maximum duty cycle for a 20-LED display is only 5%. The poor duty-cycle figure is due not to the method, however, but rather to the driving capability of the microprocessor and the parasitic-leakage paths. A single pin cannot usually sink the current a number of LEDs require to effectively light up, so these designs often require one source pin and one sink pin to light only one LED at any time. However, adding a transistor or two resistors allows you to circumvent these issues.

If you rearrange the LEDs in the familiar cross-point array and add a transistor to each column to carry the common current, you'll see the duty cycle of the Charlieplexing method does not differ much from standard multiplexing (**Figure 1**). For a 20-LED, five-column matrix, each LED remains on for 20% of the time compared with 25% for standard multiplexing, but now using only BY THE TIME YOU GET TO 90 LEDs, THE PCB REAL ESTATE AND COST OF THE 10 TRANSISTOR/ RESISTOR SETS PALE IN COMPARISON TO THE DISPLAY ITSELF.

five pins instead of nine (Table 1).

One of the drawbacks of adding the transistor and resistors to each column is that you need additional components to achieve a reasonable LED brightness when a large number of LEDs is involved. This approach, however, is a better alternative to using a costly IC and no worse than standard multiplexing or "Gugaplexing," which also requires additional transistors and resistors. From a cost and benefits point of





view, consider that, by the time you get to 90 LEDs, the PCB (printed-circuitboard) real estate and cost of the additional 10 transistor/resistor sets pale in comparison to the display itself.

Examining the circuit in detail, you'll notice that it has five microprocessor pins, P_1 through P_5 , available, for a total of $N \times (N-1) = 20$ LEDs. When P_3 , for example, is high, the emitter of Q_3 is at approximately 4.4V, and you can turn off D_{13} , D_{23} , D_{43} , or D_{53} if you make P_1 , P_2 , P_3 , or P_5 low. Any pin that you set to input, or high impedance, alternatively turns off the corresponding LED. When P₁ and P₄ are low, P₂ is high, and P, and P, are in high-impedance states. With P_1 high, transistor Q_1 biases on, all the other transistor bases are either low, which ensures that no current will flow, or high-impedance, which supplies no current into the base to allow the transistor to conduct. All the diodes in the third column can turn on, but only D_{13} and D_{43} have a path directly to ground through P_1 and P_4 , which are low and through the 100Ω current-limiting resistors.

 D_{23} and D_{53} connect to the high-impedance input pins and can conduct only through the 100Ω resistors attempting to turn on Q_2 and Q_5 . Because of their forward-voltage drop typically, 2.2V—the emitters of Q_2 and Q_5 will be less than 1.6V, as the following **equation** shows: $5V_{CC}$ = 0.6V (Q_3) = 2.2V (D_{23} or D_{53}) = 0.6V (Q_3 or Q_5) = I_{LED} × 100Ω < 1.6V, where I_{LED} is the current of the LEDs. This scenario does not allow any LED in Column 2 or Column 5 to light up to any level that would have an undesirable effect.**EDN**

REFERENCES

Lancaster, Don, *Tech Musings*, August 2001, www.tinaja.com/glib/ muse152.pdf.

Charlieplexing-Reduced Pin-Count LED Display Multiplexing," Application Note 1880, Maxim, Feb 10, 2003, http://pdfserv.maxim-ic.com/ en/an/AN1880.pdf.

Gupta, Saurabh, and Dhananjay V Gadre, "Multiplexing technique yields a reduced-pin-count LED display," *EDN*, Oct 16, 2008, pg 68, www. edn.com/article/CA6602447.

Serial port tests digital circuits

Yury Magda, Cherkassy, Ukraine

A PC's serial port provides signal lines that you can use to read voltage levels of digital circuits. You can use the port to test digital TTL (transistor-to-transistor-logic)level circuits. You just need to convert the TTL levels to RS-232 voltages, and you can add a multiplexer to increase the number of signals that the serial port can sense.

The circuit in Figure 1 uses a MAX232 IC from Maxim (www. maxim-ic.com) to convert RS-232 voltage levels to TTL levels (Reference 1). A 74HC4051 from Texas Instruments (www.ti.com) lets you select any of four digital inputs and route them to the serial port (Reference 2). Listing 1, which is available with the online version of this Design Idea at www.edn. com/090625dia, lets you control the RTS (ready-to-send) and DTR (dataterminal-ready) pins in the serial port that selects the signal under test. The CTS (clear-to-send) pin then reads the signal under test into the PC.

The four digital-input signals, A0 through A3, from your device under test connect to the first four inputs, X0 through X3, of the multiplexer. Only one of those signals can pass through to the X output, Pin 3, at a time. By setting the appropriate binary code on the serial port's RTS and DTR lines, you can select the signal to pass through the multiplexer (Table 1).

The PC software, running on Windows XP, sequentially sets those binary combinations on the port's RTS and DTR lines and reads the digital signal on the CTS line. The software then reads the status of the selected bit and displays it when you press the "checkstatus" button (**Figure 2**). The code is written in Microsoft C# 2008, but it

TABLE 1 INPUT SELECTION					
Signal to X pin	RTS bit	DTR bit			
AO	0	0			
A1	1	0			
A2	0	1			
A3	1	1			

will run on the 2005 version, as well.

To create the application, select the "Windows Form Application" from the templates in the project wizard. Place the text-box, label, and button components on the project's main form and assign titles for them. You should place the serial-port component on the design area of the project. Then, set the appropriate parameters for the serialport component, including the port number, baud rate, data bits, parity, and stop bits.

When you build the circuit, follow all precautions concerning the MAX232 and 74HC4051 wiring according to

the manufacturers' data sheets. Place bypass capacitors as close as possible to the IC's power and ground. You can replace the MAX232 with a MAX225 or MAX233.EDN

REFERENCES

"MAX220-MAX249 +5V-Powered, Multichannel RS-232 Drivers/Receivers," Maxim, January 2006, http:// datasheets.maxim-ic.com/en/ds/ MAX220-MAX249.pdf.
"CD54/74HC4051, CD54/
"CD54/74HC4051, CD54/74HC4052, CD74HCT4052, CD54/74HC4052, CD74HCT4053 High-Speed CMOS Logic Analog Multiplexers/Demultiplexers," Texas Instruments, 2004, http://focus.ti.com/lit/ds/symlink/ cd74hct4053.pdf.

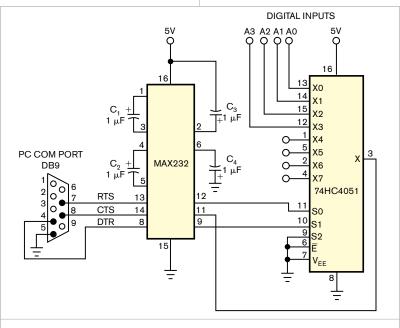
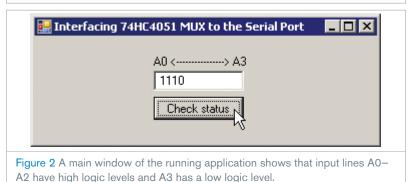


Figure 1 This circuit lets you pass up to four TTL-level signals to an RS-232 port to read their status.

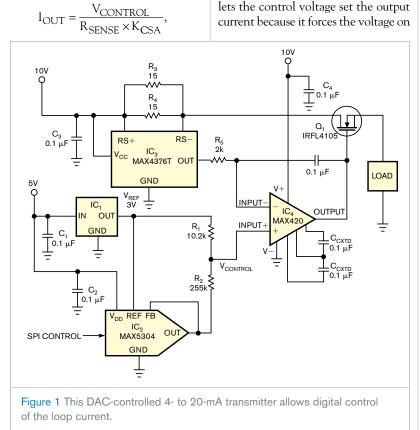


DAC calibrates 4- to 20-mA output current

Ronald Moradkhan and Steven Lau, Maxim Integrated Products, Sunnyvale, CA

Industrial controls make heavy use of 4- to 20-mA current loops to transmit process measurements because current loops retain information in the presence of noise and changes in loop voltage. The loop circuit requires proper calibration to ensure accurate readings. The circuit in **Figure 1** calibrates the loop by generating a current in response to a control voltage:

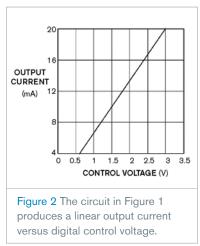
where I_{OUT} is the output current, $V_{CONTROL}$ is the control voltage, R_{SENSE} is the sense resistance, and K_{CSA} is the gain of the current-sense amplifier—20 in this case. The circuit comprises IC₂, a Maxim (www.maxim-ic.com) MAX5304 DAC; IC₃, a MAX4376T current-sense amplifier; IC₄, a MAX420 op amp; and Q₁, an N-channel IRFL4105 MOSFET. The op amp lets the control voltage set the output current because it forces the voltage on



the negative input equal to that on its positive input. The output current depends on the value of the sense resistor, the gain of the current-sense amplifier, and the control voltage.

The DAC provides the control voltage that lets you automate the calibration procedure. By selecting the right value for the sense resistor and by using a suitable resistor divider for R_1 and R_2 at the output of the DAC, you can adjust the circuit's output to 4 mA when the DAC's digital input is zero-scale and 20 mA when the digital input is full-scale. **Figure 1** shows the component values you need to achieve that condition.

With a zero-scale digital input, the DAC output is 0V and the resistor divider produces 0.6V at the op amp's positive input, forcing the output current to 4 mA. With a full-scale digital input, both the DAC output and the midpoint of the resistor divider are at the 3V reference voltage, forcing the output current to 20 mA. A transfer curve relates the output current to the control voltage (Figure 2).EDN



Alarm tells you to close the refrigerator door

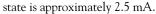
Boris Khaykin, TRW Automotive, Livonia, MI

The circuit in **Figure 1** is a simpler and safer device than a sim-

ilar one I recently read about (Reference 1). A few years ago, I built the

circuit that this Design Idea describes, and the gadget still operates with the original 9V battery. The circuit operates by sensing a decrease in resistance of photocell PC_1 that results from light in the refrigerator when its door is open. A counter is in a reset state when PC_1 is in the dark, and its resis-

tance is greater than 30 k Ω . Usually, the dark resistance is greater than 200 k Ω , and current consumption at this state is less than 40 μ A. Oscillatorcounter IC₁ starts counting when PC₁'s resistance is lower than $15 \text{ k}\Omega$ —that is, when the light bulb in the refrigerator is on. Then, in 20 seconds, it turns on a buzzer for 20 seconds or until someone closes the door. The current at this



You can use almost any photocell, such as the Jameco (www.jameco. com) 202403 CDS0018001 with 200 $k\Omega$ dark and 3- $k\Omega$ light resistance.

This circuit uses a RadioShack (www.radioshack. com) 273-074 buzzer. You can use any similar piezoelectric buzzer with an operating dc voltage of 1.5 to $15V. V_1$ can be as low as 3V.The trade-off is that using a voltage this low gives you longer battery life but lower volume of sound.EDN

REFERENCE

Babu, TA, "Alarm Sounds When Refrigerator Door Remains Open Too Long," *Electronic Design*, March 26, 2009, pg 46, http://electronicdesign. com/Articles/ArticleID/ 20806/20806.html.

