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Teaching Communication Systems with Simulink and the USRP

Abstract

This paper discusses a system to demodulate and decode radio signals in real-time using the universal software radio peripheral (USRP) in combination with Simulink in an undergraduate teaching environment. Several previous papers have shown the value of using real world radio signals to inspire and motivate students in traditional communication systems courses and laboratories. Recent versions of Simulink now support a direct connection to the newer versions of the USRP, and Simulink can connect to the original USRP1 through an existing open source software package. This capability allows users to rapidly prototype radio systems that run in real-time on the host computer, which enables more realistic and interesting projects, laboratory experiments, and demonstrations. To illustrate this capability, a Simulink model is presented that implements a real-time FM radio that demodulates and plays the audio signal, and it also decodes the digital Radio Data System (RDS) signal and prints out the RadioText messages which usually convey the name of the artist and song that is currently playing. The use of this receiver as a demonstration in an undergraduate communication systems course is discussed, and its effectiveness is assessed. The source files for this receiver are available for free from the author.

Background

Communication systems classes usually contain highly mathematical and theoretical material which is sometimes difficult for students to learn. It can be very helpful to motivate the students by showing how the material is used in real systems. Simulations are an effective and easily deployed method, but they do not have as direct connection to the real world as real systems. However most real communication systems are very complex, and it is often difficult to utilize them in a class. Some laboratory experiments have been based on the USRP because they can be used to build communication systems that run in real-time without the need for building low level hardware and software.

The USRP is an open source hardware platform provided by Ettus Research. It converts radio frequency (RF) signals to digital signals that can be processed by a computer, and also converts digital signals from a computer to RF signals. Its function is analogous to a sound card, except that it can capture and generate high frequency RF signals. Using the USRP allows the students to work with RF signals from existing communication systems and to generate their own RF signals.
Simulink is a programming environment that works in conjunction with MATLAB\(^4\). It allows the user to build a program by connecting blocks that represent basic functions into a block diagram. This programming environment is often used to run simulations, but it can also process data on a standard computer in real-time as long as the system does not require too much computation.

Simulink now has the ability to take data directly from the USRP and deliver data to the USRP, which allows a real-time communication system to be constructed using a high level language like Simulink and commercially available hardware. Since it greatly simplifies the process of implementing a communication system, this platform is very useful for demonstrating communication systems in a lecture format and for experimenting with communication systems in a laboratory.

The Radio Data System (RDS), which is also called Radio Broadcast Data System (RBDS) in the United States, is a digital signal that is sent along with the analog audio signals in FM radio broadcasts. This signal is usually used to transmit the title of the song, the name of the artist, the name of the program, and other information. The RDS signal, which is added to the analog audio signals before frequency modulation, uses a carrier frequency of 57 kHz, bi-phase (or Manchester) encoding, a raised cosine pulse shape, and a bit rate of 1187.5 bits/sec\(^5\),\(^6\). The RDS signal is useful in teaching communication systems because it can be demodulated and decoded with a reasonably simple receiver, and because it is a signal that is familiar to most students\(^7\).

**Hardware**

In order to demonstrate the abilities of Simulink and the USRP in an educational setting, an FM receiver was developed that not only demodulates and plays the audio signal from an FM radio station, but it also demodulates the digital RDS signal and prints out the RadioText messages that usually contain the title of the song, name of the artist, and other information.

The hardware consists of an antenna, USRP1 with an appropriate daughterboard such as the TVRX or the WBX, and a standard PC or laptop as shown in Figure 1. The antenna can be a length of wire or an FM antenna. The author used the USRP1, which costs about $700 and is one of the inexpensive versions of the USRP\(^3\), but it should be possible to use the other versions of the USRP as well.
Software

The Simulink-USRP Project is an open source project that created software that allows Simulink to interface with the USRP1 (Simulink can interface directly to the newer USRP models, in which case it is not necessary to install this software.) Once this software is installed, models that use data from the USRP1 can be built and run using MATLAB/Simulink from Mathworks. The top level Simulink model for the FM mono receiver with RDS decoder is shown in Figure 2.

Figure 1. Hardware (USRP1, Laptop, and Antenna)

Figure 2: Top Level Simulink Model
The block on the top left of Figure 2 is labeled usrp_source, and it represents the USRP1. Double-clicking on this block opens a dialog box for configuring the USRP. The Simulink model was designed for a sampling rate of 250 kHz. Since the analog-to-digital converters on the USRP1 run at 64 MHz, the USRP was configured for a Decimation Factor of 256 which leads to a sampling rate of $64 \text{ MHz}/256 = 250 \text{ kHz}$. The sample width was set to 16 bits, and the vector length was set to 1024 samples. The TVRX or WXB daughterboard was set to a data type of INT16 Complex, and the frequency was set to the carrier frequency for the desired radio station. The gain is set depending on the signal strength of the station. The data from the USRP is a complex signal that represent the complex envelope of the sampled RF radio signal from the FM station.

Some values, such as filter coefficients, that are used in the model are computed when the Simulink model is initialized. These values can be examined by opening the model with Simulink and then clicking on File, Model Properties, Callbacks, InitFcn.

The FM Demodulation block, which is shown in Figure 3, demodulates the FM signal, which results in the stereo multiplex signal, whose spectrum is represented in Figure 4.

**Figure 3: FM Demodulation Block**

**Figure 4: Spectrum of the Stereo Multiplexed Signal**
After the FM signal is demodulated by the FM Demodulation Block, the analog mono (L+R) signal is isolated by a lowpass filter, which is then down sampled to reduce the amount of computation required, passed through the FM de-emphasis filter, and sent to the computer’s speakers (see Figure 2).

The bottom branch of Figure 2 contains the RDS digital receiver. First the RDS signal is isolated from the other signals by a bandpass filter centered at 57 kHz. Next the signal is down-sampled by a factor of 14 to greatly reduce the amount of computation that is required by the rest of the receiver. The down-sampling operation intentionally undersamples the signal, which causes aliasing, and one of the aliases falls at $57000 - 3 \times \frac{250000}{14} = 3428.6$ Hz. So, after the down sampling operation, the RDS signal is centered at 3428.6 Hz and the new sampling rate is $\frac{250000}{14} = 17857$ Hz.

Next the signal is shifted down to baseband by a Costas Loop, which estimates the actual carrier phase from the received signal in order to account for the carrier frequency and phase offsets that occur in real systems. Although FM radio transmissions include a pilot carrier signal that can be used instead of the Costas Loop (see Figure 4), most digital communication systems do not have a pilot carrier, and so using the Costas Loop makes it easier to modify the receiver for use in other digital systems. The contents of the Costas Loop block are shown below in Figure 5.

![Costas Loop Block Diagram](image-url)
The output of the Costas Loop is passed through a matched filter to remove as much noise as possible, and then the signal is passed to the block labeled “Timing Recovery, etc.” which is shown in Figure 6.

**Figure 6: Timing Recovery, etc. Block**

A timing recovery algorithm is necessary to determine when to sample the baseband signal to locate the peak of the pulse representing each bit. A modified timing recovery algorithm was used for this receiver because the standard timing recovery algorithm based on maximizing the power of the sampled signal would sometimes converge to a local maximum, resulting in it sampling at the incorrect times. The modified method is based on the current sample and the sample from one-half symbol period earlier, and the slope of the signal at these times. The delay line block in Figure 6 generates the samples needed by the timing recovery algorithm (see Figure 7).
The Compare Time block, which is shown in Figure 8, enables the Enabled System block when the current time is greater than the next sample time.

Since the rest of the processing is performed only when the system samples the signal, the rest of the processing is enclosed in the Enabled Subsystem, which is shown in Figure 9.
Most of the blocks in Figure 9 are used to compute the next sample time for the timing recovery algorithm. The top branch of the model in Figure 9 contains the slicer (quantizer) and the differential decoder, and the call to a MATLAB function which decodes the RDS data and prints the RadioText messages.

The MATLAB function first buffers up 104 bits and then calls a function that searches for the beginning of a valid RDS group by checking the CRC checksum bits for each possible starting point. Once a valid RDS group is found, it is checked to see if it is a RadioText group, and if so, the ASCII characters are extracted from the group and printed to the MATLAB command window.

This system runs in real-time, so the students can listen to the music from a radio station while watching the RDS RadioText messages scrolling by in the command window. The signal at any point in the system can easily be captured and graphed in MATLAB, which is a valuable technique for debugging the system as well as for learning how the system works. For example, the inputs to the slicer, which are called the soft decisions, are often used to determine how well a system is working, and they can be captured by connecting a “To Workspace” block to the input of the Slicer Block in Figure 9. After the model is run for a short time, the soft decisions can be plotted using the MATLAB command below.

```
>> z = double(simout.signals.values); plot(z(:),'.')
```
An example of the soft decisions plot is shown in Figure 10.

![Plot of Soft Decisions](image)

**Figure 10: Example Plot of the Soft Decisions**

**Use in the Classroom**

The author currently uses the USRP/Simulink receiver as a demonstration in an undergraduate communication systems lecture course, which requires a course in signals and systems as the prerequisite. Although this communication systems course does not have a companion laboratory class, the students are required to complete a single laboratory experiment in pairs on their own time where they use a real-time spectrum analyzer (Tektronix RSA3408A) to capture the RF signal from a local FM radio station of their choice. Then over the course of the semester, the students learn how to write a MATLAB program to demodulate and listen to the mono audio signal, and to demodulate and decode the digital RDS signal and print out the Radio Text messages.

The goal of the demonstration is to increase the students’ interest in communication systems by showing how a working receiver demodulates and decodes analog and digital signals. The demonstration took about 80 minutes and was split between two class periods. First the receiver was run to allow the students to listen to the audio and watch the Radio Text messages being printed out in real-time. Next the purpose of each of the major blocks of the receiver was briefly discussed, the spectrum of the signal at each point in the receiver was plotted using the Spectrum Scope in Simulink’s Signal Processing Blockset, and the resulting baseband digital signal was plotted using the Time Scope. These plots were used to show how the receiver isolates the
signals and shifts them back to baseband. With the author’s setup, the computer is not fast enough to demodulate the audio and RDS signals while simultaneously displaying these graphs, but this limitation had little effect on the demonstration.

Assessment

In an attempt to measure the impact of the demonstration, an anonymous survey was distributed to the students prior to and after the demonstration. Neither survey was given on the same day as the demonstration. The survey asked the following questions:

1. How would you rate your current knowledge about communication systems?
2. How would you rate your current level of motivation to learn more about communication systems?
3. How interesting do you currently find communication systems?
4. List the major parts of a receiver in a digital communication system.
5. Do you have any suggestions for how the course could be improved?

The first three questions were scored on a 1 to 5 scale, with a score of 5 indicating very knowledgeable, motivated, or interested, and a score of 1 being not at all knowledgeable, motivated, or interested. The answer to Question 4 was graded by the instructor on a scale of 1 to 5, with 1 indicating no answer or no correct answer, and 5 being a complete list of the major parts of a digital receiver (carrier recovery, matched filter, timing recovery, and quantizer). The results of the survey are summarized in Table 1.

<table>
<thead>
<tr>
<th></th>
<th>Before USRP/Simulink Demo (N=20 people responded)</th>
<th>After USRP/Simulink Demo (N=16 people responded)</th>
</tr>
</thead>
<tbody>
<tr>
<td>STD</td>
<td>0.6584</td>
<td>0.6387</td>
</tr>
</tbody>
</table>

As shown in Table 1, the mean values of the students’ self-reported knowledge increased slightly after the demonstration, but the mean value of the students’ self-reported motivation and interest
decreased slightly. The instructor’s score for Question 4 increase slightly after the demonstration. However, none of these differences was big enough to be considered statistically significant at the 5% significance level according to the t-test for difference of population means for unknown and not necessarily equal variances.

In response to Question 5 in the survey before the demonstration, three of the students (out of 20) mentioned that they liked real-world examples and/or applications. In the post-demonstration survey, three of the students (out of 16) stated that they liked the demonstration or wanted more demonstrations. One student wrote, “More demos! I love em!”

Although the numerical results of the survey were inconclusive, the responses to Question 5 seem to indicate that at least some of the students thought that the demonstration was worthwhile.

Discussion

This system demonstrates that the Simulink/USRP platform can be used to build real-time digital and analog communication systems without low level hardware and software development. The digital RDS receiver is a much more complicated system than the relatively simple mono FM receiver, and the platform can run both in real-time. This platform can be used in a communication systems course for demonstrations or lab experiments, or to allow students to test a subsystem that they create inside a complete system. For example, in a course that teaches the Costas Loop, the students could create their own Costas Loop model in Simulink, insert it into the receiver, and test it as part of a real receiver running in real-time.

The source files for the receiver described in this paper can be obtained for free by sending email to the author.

Conclusion

This paper presented an example analog and digital communication system built using Simulink and the USRP that runs in real-time. The advantages of this relatively low cost platform for education is that it allows communication systems to be built that run in real-time using a high level programming environment and commercially available hardware.
References

2. Welch, Thad B.; Shearman, Sam; “Labview, the USRP, and their Implications on Software Defined Radio,” ASEE Annual Conference and Exposition, Conference Proceedings, 2011.