

CMPE 663 Project 5

Signal Generator

Andrei Tumbar

Instructor: Wolfe
TA: Nitin Borhade

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Analysis/Design

This project looked at generating certain waveforms with varying parameters. The usage of the DMA in tandem with the DAC was crucial to the operation of this exercise. A system was designed in which the DAC could be utilized to its maximum operating speed when enough RAM was available.

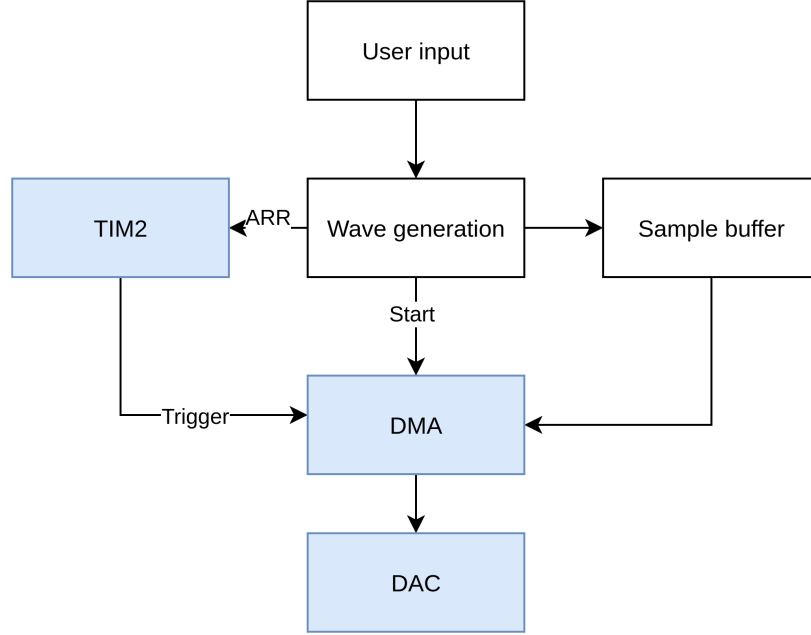


Figure 1: Overview of system operation

Hardware Operation

The signals that must be generated for this project need to reach a maximum of 100 kHz. Unlike the previous project, operating the DAC using a timer interrupt will not work properly at this frequency. Interrupting the CPU causes inherent delay that will far exceed the minimum response time at the maximum frequency. To circumvent this issue, the DMA may be used to write to the DAC directly. The DMA peripheral must be triggered by a hardware source to operate in the way we intend. TIM2 was used again to trigger the DMA. Like the previous project, we can represent the frequency of a discrete periodic wave given the trigger frequency and number of buffer samples.

$$F_{sysclk} = F_{wave} \cdot N_{samples} \cdot (ARR + 1)(PSC + 1) \quad (1)$$

For now we will assume that the DAC may operate at an arbitrarily high frequency and we will account for this later on. F_{sysclk} , F_{wave} , and $N_{samples}$ are known constants. ARR and PSC are variables. In the previous project we set, $PSC = ARR$. However this method will not allow the accuracy at higher frequencies. We are now setting $PSC = 0$.

$$ARR = \frac{F_{sysclk}}{F_{wave} \cdot N_{samples}} - 1 \quad (2)$$

We assumed previously that the DAC may operate at an arbitrarily high rate. This is not an accurate assumption as the DACs on the this STM32 board have a maximum sample rate of 1 MSPS. Using the wave frequency, a known constant, we can calculate the maximum number of samples we can use inside a single period to reach the maximum sample rate of the DAC peripheral.

$$N_{samples} = \frac{1 \text{ MSPS}}{F_{wave}} \quad (3)$$

Because we don't have infinite memory and our sample buffer is statically allocated, we must cap the number of samples to the chosen length of the sample buffer. A sample buffer length of 2000 was chosen because it creates a smooth curve at the minimum operating frequency and isn't too large.

Wave Generation

To generate the waveforms for the triangular, square, and sinusoid, differing methods were used. The square wave was the simplest with half of the buffer being the maximum voltage and the other half being the minimum voltage. The triangle wave could be generated by calculating the slope between the min and max voltages over half of the usable buffer length. Then use this slope to interpolate a rising line in the first half and a mirrored falling line in the second half. Lastly, the sinusoid is generated by scaling the `sin()` function from the `math` library to the length of buffer. The move and scale this function to fit within the desired min/max range.

Test Plan

To test operation, an oscilloscope was used to monitor the wave generation. The oscilloscope can measure periodic wave frequencies and was able to verify the correct operation of the system. To test, low frequencies were used first to verify that the chosen buffer length was long enough to create smooth curves. The highest frequencies were tested last to verify that the algorithm to determine the buffer length operated properly.

Project Results

Results shown on the oscilloscope are expected. Low frequency waveforms have very smooth profiles while higher frequencies (higher than 10 kHz) start to show more jagged edges. This is expected because the maximum sample rate of the DAC peripheral begins to have significant impact waveforms profile. At the highest frequency (100 kHz), we can see that only about 10 samples are used which create step like function for the triangular and sinusoidal waveforms. The square wave does not suffer at all from the lower sample count as it only requires two samples to fully model the waveform.

Lessons Learned

This project explored the operation of the DMA in conjunction with the DAC. When starting the project, following a DMA/DAC setup guide for STM32 did not work at all. There turned out to be a completely unrelated peripheral accessing the GPIO bus causing the DMA to not play nicely with the DAC. Starting the project from scratch and not initializing peripherals to their defaults allowed the DMA to work as intended. The second major obstacle was finding out that the DAC had a maximum sample rate. Deriving an algorithm to limit trigger frequency of the timer by limiting the number of samples used was vital in making this project as robust as possible.