

Article

Design and Implementation of Scalable and Parametrizable Analog-to-Digital Converter on FPGA

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Abstract: The flexibility provided by FPGAs permits the implementation of several ADCs, each one configured with the required bit resolution and sampling frequency. The paper presents the design and implementation of scalable and parametrizable analog-to-digital converters (ADC), based on a successive approximation register (SAR), on FPGAs (field programmable gate arrays). Firstly, the work develops a systematic methodology for the implementation of a parametrizable SAR-based ADC from a set of building modules, such as the pulse-width modulator (PWM), external low-pass filter (LPF) and the analog comparator. The presented method allows choosing the LPF parameters for the required performance (resolution bits and sampling frequency) of a SAR-based ADC. Secondly, the paper also presents several optimizations on the PWM module to enhance the sampling frequency of implemented ADCs, and the method to choose the LPF parameters is adapted. The PWM and SAR logic are synthesizable and parametrizable, using a low number of resources, in order to be portable for low-cost FPGA families. The methodology and PWM optimizations are tested on a Zynq-7000 device from Xilinx; however, they can be adapted to any other FPGA.

Keywords: analog-to-digital converter (ADC); successive approximation register (SAR); FPGA; PWM; LPF



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1. Introduction

In the field of control systems, a set of signals are collected from sensors in order to perform the necessary actions to obtain the desired output from a reference indicated by the user [1,2]. A digital controller is usually implemented on a programmable device, such as a microcontroller, DSP, FPGA, etc. [3,4], which relies on ADCs to perform the required conversion of data acquired [5]. Important features of ADCs are the bit resolution (N), sampling frequency ($f_{sampling}$) and range of the input voltage (V_{in}).

There is a growing interest in MD (mostly digital) systems where most of the ADCs are implemented in the digital domain, simplifying its integration with digital cores (micro-processor, DSPs, Hw-accelerators, etc.) to build a SoC (system on a chip). The design flow provided by FPGA vendors allows the integration on embedded systems of digital circuits, such as FIR filters that might need external ADC [6]. Using discrete ADCs, such as LTC1406 (8-bit ADC 200 MSPS), in applications that require a high number of ADCs can inhibit its connection due the limited quantity of I/O pins in low-cost devices. In this case, a possible solution is to avoid the need for ADCs as shown in [7], where a digital signal is connected to a FPGA pin and its temporal width is measured by internal counters. Otherwise, FPGAs allow the implementation of ADCs, which, in the simplest case, requires a pin connected to a passive resistance-capacity (RC) filter and another pin to an internal logic gate acting as a analog comparator when the threshold voltage is exceed. However, the implementation of ADCs on FPGAs presents limitations, such as external analog circuits (integrators or filters), performance that cannot achieve discrete ADCs, and higher power consumption due to programmable technology.

Traditionally, digital controllers are implemented on microcontrollers or DSPs due to the development ease provided by programming languages and compiler [8,9]. However, the computing performance is limited by the sequential execution of instructions. FPGAs are programmable logic devices, which can implement a digital circuit using its hardware resources. The circuit is described in a hardware description language, typically Verilog or VHDL [10], and it is synthesized and implemented on the target device through automated design tools. Typically, the FPGA fabric is composed from a large set of logic resources, such as lookup tables (LUT) [11], flip-flops (FF), input–output blocks (IOBs) and others, that are interconnected through configurable routing resources. The available hardware resources allow the implementation of highly parallel circuits for multiple controllers that can greatly accelerate computations on FPGAs compared to a microcontroller or DSP.

An application such as SDR (software-defined radio) needs ADCs with high f_{Sampling} on the IF (intermediate frequency) stages to enable the digital sample processing. Most advanced SDRs are able to remove conversion stages to IF, reducing operational amplifiers, filters and other analog elements, using high-performance ADCs (GSPS). The high-performance ADCs are usually implemented on full-custom integrated circuits and use time-interleaving techniques that require several ADCs in parallel to convert the same input [12–15].

ADCs are not usually integrated on low-cost FPGAs. Some high-performance FPGAs may integrate an ADC, which can be insufficient for multiple controllers converting simultaneously several signals at different sampling frequencies, resolutions and voltage ranges. For example, in on-board battery charges for electrical vehicle, working in one or three phases, a large amount of signals need to be read for control and protection purposes [16–18]. In [19,20], reading several signals is needed to control simultaneously multiple independent power converters by a single DSP, where the processor is implemented on a FPGA. In biomedical applications, multi-channel acquisition are simultaneously needed [21], while in battery management systems, several signals are used at the same time for monitoring and protection purposes [22]. In such cases, external ADCs can be attached to the FPGA at the expense of additional cost.

The paper proposes a methodology to implement scalable and parametrizable ADCs on FPGAs. The scalability permits the implementation of several ADCs on a single FPGA device, each one configured according to a set of parameters to individually adjust the required sampling features (N , f_{Sampling}). The ADC architecture is based on a SAR controlling a DAC (digital-to-analog converter), which is built using a PWM and external LPFs. The SAR and PWM circuits can be implemented in the LUTs and FFs available in almost any FPGA family. The circuits do not rely on special hardware resources for a particular device (delays on carry chain), external input, such as space vector PWM [23], sinusoidal PWM [24], or carrier phase shifted PWM [25] used in other applications or specific tools from a device vendor that would limit the porting to different FPGA families. Moreover, the presented work also describes several optimizations of the PWM devoted to increase the maximum selectable f_{Sampling} .

Section 2 presents the state of the art where recent works related to fully synthesizable ADCs, ADC fully implemented on FPGAs, and SAR-based ADCs, among others, are shown. Section 3 introduces the SAR-based ADC, focusing on the analysis of the PWM module and the attached low-pass filter (LPF) that built the DAC. It finally presents a methodology to design a parametrizable ADC. Sections 4–6 present and analyze three optimizations devoted on improving the f_{Sampling} : double data rate PWM, paralleled PWM, and distributed duty-cycle PWM. The next section analyzes the combination of the three optimizations. The experimental results are reported in Section 8. Sections 9 and 10 present the discussion and conclusions, respectively.

2. State of the Art

Flash-type ADCs are based on a large set of analog comparators and a priority encoder. Each comparator generates a digital channel from comparing V_{in} against a scaled voltage

from V_{REF} . Although they provide high conversion rates, the power consumption, implementation area and linearity errors are important issues. Stochastic flash ADCs improve the linearization of the conversion by further increasing the number of comparators and adding an interpolation circuit at each digital channel. They are mixed-signal circuits that are usually implemented on custom CMOS integrated circuits [26] (8190 comparators, 6.2 ENOB, 100 MSPS) or synthesized on standard cells [27] (2040 comparators, 5.2 ENOB, 320 MSPS), [28] (2047 comparators, 5.7 ENOB, 210 MSPS). Nevertheless, flash ADCs are not suitable on FPGAs due to the large number of required external comparators and I/O connections. VCO-based ADCs can achieve a high sampling frequency. They are composed of an oscillator, a quantizer and a differential module. The works [29,30] showed a VCO-based ADC, which is fully synthesizable and described using a digital HDL, implemented in standard cell 65 nm CMOS. An architecture to eliminate the non-idealities of the coarse and fine VCO-based quantizers by replacing them with a PWM and filters is presented in [31]. The described VCO-based ADCs are implemented in CMOS technology and require a significant amount of I/O connections and methods to improve linearization. A sigma-delta ADC presented in [32] is implemented with simple analog circuit on a FPGA Virtex-4, using differential input, and requiring more complex digital circuitry. A high precision and low sampling frequency sigma-delta-based ADC on a A3PE3000 flash FPGA is used in [33] to balance a dynamically tuned gyroscope. Using the low voltage differential signal (LVDS) receiver inside the FPGA and an integrator, ref. [34] implemented a sigma-delta ADC for a current sensing application.

Most of the high-performance ADCs in FPGAs are based on a time-to-digital converter (TDC) architecture [35–37]. They are based on the temporal measurement of charging a RC circuit until V_{in} is reached. A comparator generates a digital pulse proportional to V_{in} , and the TDC circuit measures the width of this pulse. Although the TDC circuit can be easily implemented by a digital counter, these implementations greatly improve the $f_{sampling}$ by a delay network. The delay network is composed of a large number of sections of a specific FPGA resource devoted to transmitting the input to the output with a known delay by estimation or measurement. The conversion code is generated from the large number of signals retrieved from the delay network. The precision stability of TDC-based ADCs is greatly affected by the tolerances of R and C values, and by temperature or voltage fluctuations. Therefore, they usually require periodic calibration.

The 600 MSPS 7-bit ENOB (effective number of bits) ADC presented in [38] is a high-speed converter implemented in an UltraScale+ FPGA from Xilinx, with a periodic calibration circuit. The ADC implements a tapped-delay line, an edge detection circuit attached to an encoder to generate the conversion result, and a periodic calibration circuit. In this particular case, the delay is measured through a tapped-delay line, which is composed of 426 multiplexor-XOR sets from the UltraScale+ carry logic. The ADC does not require external components, since an output buffer (OBUF) and a differential input buffer (DIFFINBUF) from UltraScale+ replace the capacitor, resistance and the analog comparator. Xilinx allows configuring the slew rate and the desired impedance of the OBUF, which is externally connected to a DIFFINBUF that provides a parasitic capacitance. The DIFFINBUF is used as an analog comparator, connecting one input to the OBUF and the other to V_{in} , generating a digital output when the input difference crosses the threshold voltage. Nevertheless, the tolerance on the CMOS manufacturing process of integrated circuits can significantly affect the electrical values, such as the parasitic capacitance and the threshold voltage of the DIFFINBUF and the transistor impedance of the OBUF. Moreover, the non-linear relationship between V_{in} and time requires a higher number of signals retrieved from the delay line and its encoder, which requires periodic calibration to maintain the precision stability due to variations.

The 200 MSPS, 6-bit ENOB ADC presented in [35] is a bit simpler. It implements a tapped-delay line but is based only on the multiplexors of the carry chain from Spartan-6 FPGA. Moreover, this ADC requires an external resistance to charge the parasitic capacitance from a DIFFINBUF. Another example is the 100 MSPS 6-bit (3.8 ENOB) ADC presented

in [36], which must be maintained in cryogenic temperatures by a liquid helium environment at 4 °K. It also uses a DIFFINBUF attached to an external resistance, although it uses ISERDES (input serializer/deserializer) from Artix-7 to implement the tapped-delay line.

Using very specific hardware resources from a FPGA device to implement ADCs makes portability to other families very difficult. The hardware resources that implement the tapped-delay line cannot be placed and routed by automated tools, as they must be manually located and interconnected in the FPGA to achieve the desired delays. Moreover, the differences in the electric parameters between FPGA families may require redesigning the encoder and delay line. Finally, these ADC are designed and tested for specific ADC requirements, meaning they are not parametrizable to other resolution bits and sampling rates. Changing these requirements makes it necessary to change the delay line, calculate or measure the new propagation times and the encoder tuning.

The ADC proposed in this work is a SAR-based ADCs. Although SAR-based ADCs cannot reach the f_{Sampling} of TDC-based ADCs, they can provide several advantages, such as portability to other FPGAs and adaptability to different ADC requirements. The ADC is composed of a SAR, a DAC (digital PWM and an external filter) and analog comparator. The PWM and filter is the most economic option to implement the DAC, as described in [39], and the external components, such as resistance and capacitors are chosen with 1% tolerance to avoid periodical calibration. The SAR and PWM circuits are described in portable VHDL descriptions since the synthesized result only requires the flip-flops and look-up tables available on any FPGA. The circuits are also parametrizable to the ADC requirements (f_{Sampling} and resolution bits), and scalable allowing to implement several ADCs on a device due to the low number of devoted resources. Moreover, depending on the ADC requirements, several PWM optimizations can be applied to improve f_{Sampling} .

The SAR-based ADC presented in [40] is a more similar solution to the proposed one. The converter is built from an internal DAC [39,41,42], composed of a Dyadic PWM and a passive RC filter, which is fully-synthesizable on standard cell 40 nm CMOS technology. However, this solution uses instances to the library of the CMOS technology in order to permit the filter implementation by using the automatic translation tools. This way, the R of the filter is implemented on the high-resistivity polysilicon layer, and the C by using the metal–insulation–metal capacitance provided in the 6th metal layer of the target technology. Obviously, these type of instances are not available in FPGA libraries, and the filter must be externally implemented, as in our case. The ADC performance is very limited, providing $f_{\text{Sampling}} = 2.8$ kSPS and 7-bit resolution (6.4 ENOB). Compared to other ADCs—completely or partially synthesizable on a standard cell, with flash architecture, sigma-delta or based on a VCO—its main advantage is the important reduction of the implementation area and power consumption and being fully synthesizable for an easy integration.

A very recent SAR-based ADC is presented in [43], which improves the previous ADC, achieving 10 MSPS and 7.5 ENOB. The ADC is synthesizable in standard cell 65-nm CMOS technology, and it is based on an inverter-based RDAC (Resistive DAC), a compensating LUT and a OAI-based (or-and-inverter) comparator. The OAI-based comparator replaces the single logic gate used as an analog comparator, allowing the OR to reset the transistor nodes of the AND preamplifier in order, eliminating residue charges at nodes that undermine ADC linearity. The inverter-based RDAC permits to eliminate the LPF, reducing the t_{Settling} to enhance the f_{Sampling} , but increases the I/O connectivity and requires compensation circuitry due to the non-linear resistance offered by the transistors from inverters. The compensating LUT is tuned according to the simulation results to independently switch on/off inverters of the RDAC, improving conversion linearity. Implementing an internal RDAC in FPGAs is not possible, as they do not provide internal resistors, except for pull-up/pull-down transistor-based impedance available only at I/O blocks. The inverter-based RDAC would be theoretically possible, but the resistance values are very dependent on the FPGA family, as in the tapped-delay line, and consequently, this approach would not be easily portable or parametrizable.

3. SAR-Based ADC

A SAR-based ADC compares the voltage at its input (V_{in}) with a continuous voltage generated by a DAC (V_{DAC}) from the binary number (B) driven by the SAR. The ADC evaluates a set of voltage levels converging to input voltage after several iterations [44,45]. Starting from the MSB (most significant bit) of the result, it computes a new bit by comparing the V_{DAC} with the V_{in} at each iteration, increasing or decreasing the V_{DAC} to converge to the input voltage V_{in} [46].

At the start of the conversion, the SAR initializes the MSB of the conversion result (B) to '1' and the rest of the bits to '0'. The SAR output attaches to the DAC to generate the V_{DAC} , which is compared with V_{in} . During each iteration, the SAR writes the currently evaluated bit of B by reading the comparator output $V_{Comparator}$, generating a new V_{DAC} for the next iteration. If $V_{Comparator}$ equals to '1', this means the V_{DAC} voltage is lower than V_{in} , and, consequently, the current bit is written to '1'. Otherwise, the SAR stores '0' in the evaluated bit since the V_{DAC} is higher than V_{in} . Then, the following bit of B is initialized to '1' in order to generate the V_{DAC} for the next iteration, as depicted in Figure 1. In an N-bit ADC, the process finishes when the LSB (least significant bit) is evaluated, after N iterations, obtaining the value of B that represents V_{in} .

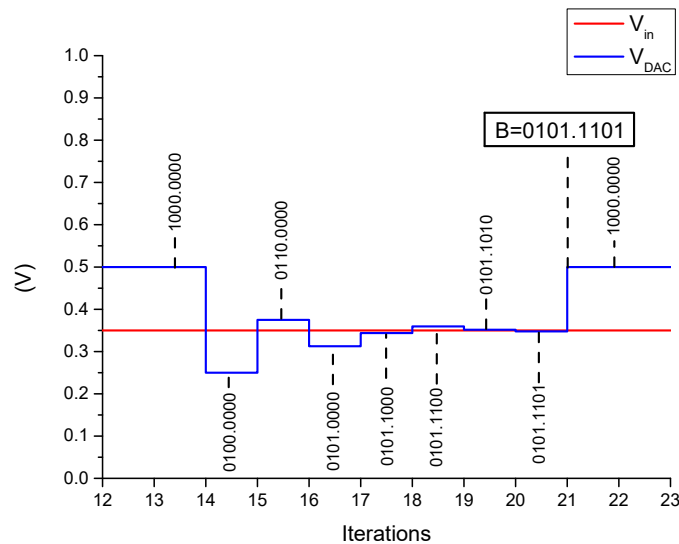


Figure 1. V_{DAC} and V_{in} for an 8-bit SAR-based ADC.

Each bit requires a settling time ($t_{Settling}$) due to the analog components involved in the DAC and the comparator. Therefore, in a N-bit ADC, the frequency sampling is computed as

$$f_{Sampling} = \frac{1}{N \cdot t_{Settling}} \tag{1}$$

3.1. DAC Implementation with PWM and LPF

The ADC replaces a resistor ladder DAC architecture with a PWM [47], which is implemented in the FPGA, and an external LPF, as shown in Figure 2. The output voltage of the LPF (V_{LPF}) is proportional to the duty cycle (D) of the PWM. The range of D is $0 \leq D < 1$ according to Equation (2), where d denotes the value at the PWM input. Finally, the V_{LPF} voltage is shifted $-\frac{1}{2}LSB$ to generate the V_{DAC} required by the analog comparator, according to Equation (3). An analog adder performs the voltage shifting in order to reduce the ADC offset error from $\pm LSB$ to $\pm \frac{1}{2}LSB$.

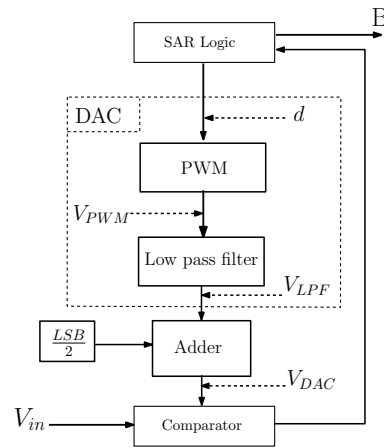


Figure 2. Block diagram of SAR-based ADC with DAC implemented with PWM and LPF.

$$D = \frac{d}{2^N} \tag{2}$$

$$0 \leq d \leq 2^N - 1$$

$$V_{DAC} = V_{LPF} - \frac{LSB}{2} \tag{3}$$

The allowed range of voltages at the input V_{in} of the ADC is related to the output voltages at the FPGA ports. V_{OH} and V_{OL} denote the output voltages at a FPGA port driving the logic levels ‘1’ and ‘0’, respectively. The V_{LPF} is obtained according the Equation (4), where the average voltage ($\overline{V_{LPF}}$) is related to D and $\Delta V_O = V_{OH} - V_{OL}$. However, the V_{LPF} also carries a sinusoidal wave $\Delta V_{LPF}(t)$ due to the limited attenuation of the LPF. The peak-to-peak voltage of $\Delta V_{LPF}(t)$, denoted as ΔV_{LPF} , depends on the the cutoff frequency (f_C) and order (O_{LPF}) of the filter and must be lower than the ADC analog resolution (LSB), as expressed in Equation (5).

$$V_{LPF}(t) = \overline{V_{LPF}} + \Delta V_{LPF}(t) \tag{4}$$

$$\overline{V_{LPF}} = D \cdot \Delta V_O$$

$$\Delta V_{LPF} < LSB$$

$$LSB = \frac{\Delta V_O}{2^N} \tag{5}$$

3.1.1. PWM Module

The PWM schematic is presented in Figure 3. The PWM module is implemented on the FPGA by an ascending N-bit counter and a FSM (finite state machine). Each PWM period is composed of 2^N counts, from 0 to $2^N - 1$, by incrementing the counter register at each rising edge of the input clock. Therefore, the PWM frequency f_{PWM} is obtained according to Equation (6), where f_{clk} denotes the clock frequency of the FPGA.

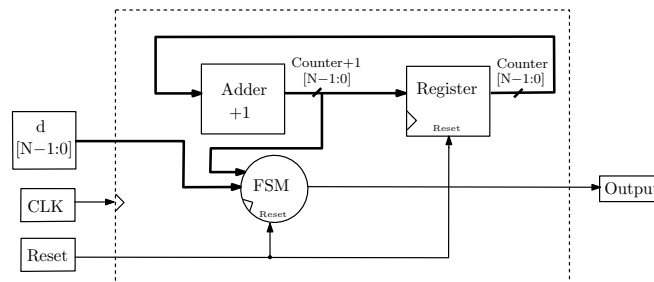


Figure 3. Schematic of the N-bit PWM.

$$f_{PWM} = \frac{f_{clk}}{2^N} \quad (6)$$

The FSM state changes when the counter is restarted to 0 or when it is going to achieve the input d , asserting or deasserting the PWM output, respectively. The duty cycle at the PWM output is $D = \frac{d}{2^N}$, where the range of the digital input d is from 0 to $2^N - 1$.

Fourier transform is applied in order to analyze the ΔV_{LPF} from the filtered PWM output. The PWM output is a square waveform characterized by the duty cycle D , and its transform is composed of the continuous coefficient A_0 and the set of harmonic coefficients A_n and B_n [48,49], as shown in Equation (7).

$$\begin{aligned} A_0 &= D \cdot \Delta V_O = \frac{d}{2^N} \cdot \Delta V_O \\ B_n &= 0 \\ A_n &= 2 \cdot \frac{\Delta V_O}{n\pi} \left[\sin(n\pi D) \right] \end{aligned} \quad (7)$$

The A_n harmonics depends on D . The first harmonic A_1 is the most significant coefficient affecting ΔV_{LPF} [50], which is maximal when $D = \frac{1}{2}$ [51], as expressed in Equation (8).

$$A_{1(max)} = 2 \cdot \frac{\Delta V_O}{\pi} \left[\sin\left(\frac{\pi}{2}\right) \right] = 2 \cdot \frac{\Delta V_O}{\pi} \quad (8)$$

The maximum peak-to-peak voltage ripple at the LPF output is mainly obtained from filtering the first harmonic of the PWM, as shown in Equation (9).

$$\Delta V_{LPF} = A_{LPF}(f_{PWM}) \cdot 4 \left(\frac{\Delta V_O}{\pi} \right) \quad (9)$$

$A_{LPF}(f_{PWM})$ represents the filter magnitude gain at the PWM frequency. Therefore, from Equations (5) and (9), it can be obtained an upper limit for the LPF magnitude gain at f_{PWM} , as in Equation (10).

$$A_{LPF}(f_{PWM}) < \frac{\pi}{4} \cdot \frac{1}{2^N} \quad (10)$$

3.1.2. Low-Pass Filter (LPF)

In order to avoid an ADC conversion error due to the LPF, Equation (10) must be satisfied to ensure the ΔV_{LPF} at the LPF output is lower than the ADC analog resolution. Since the f_{PWM} is constant, the f_C and O_{LPF} of the filter are calculated to properly reduce the ΔV_{LPF} . Reducing f_C can achieve the goal, because ΔV_{LPF} is proportional to the f_C/f_{PWM} ratio. Furthermore, increasing the filter order O_{LPF} also reduces ΔV_{LPF} since the asymptotic attenuation slope of the LPF is $O_{LPF} \cdot 20 \frac{dB}{dec}$ at filtered frequencies. However, both solutions also increase $t_{Settling}$.

The LPF implemented is a Bessel filter, as it has the shortest settling time ($t_{Settling}$) compared to other types of filters [52]. A lower $t_{Settling}$ reduces the conversion time, enhancing the ADC sampling frequency ($f_{Sampling}$). The Bessel LPF is implemented on a Sallen–Key topology. A Sallen–Key stage implements an $O_{LPF} = 2$ filter by a pair of resistors and capacitors attached to an operational amplifier (OA). Several stages are cascaded to increment the O_{LPF} .

It is required to find a trade-off solution for the LPF. The $t_{Settling}$ is noticeably much more affected by decreasing f_C than by increasing the O_{LPF} . A solution based exclusively on incrementing O_{LPF} requires more stages and circuit complexity, but a solution solely based on reducing the f_C significantly affects the $t_{Settling}$.

3.1.3. LPF Parameters for a N-Bit SAR-Based ADC

It is presented a methodology to obtain the optimal LPF parameters (f_C and O_{LPF}) from the ADC resolution (N) and the required $f_{Sampling}$. The magnitude plot of the frequency response of a Bessel LPF, according to f_C and O_{LPF} , is depicted in Figure 4a, and Figure 4b shows the normalized step response of the Bessel LPF for different O_{LPF} .

Firstly, Equation (11) computes the maximum filtered PWM output from Equation (10) in logarithmic scale. The filtered PWM output at f_{PWM} is shown in Equation (12).

$$A_{LPF(max)} = 20 \cdot \log_{10} \left(\frac{\pi}{4} \cdot \frac{1}{2^N} \right) \tag{11}$$

$$A_{LPF}(f_{PWM}) = -20 \cdot O_{LPF} \cdot \log_{10} \left(\frac{f_{PWM}}{f_C} \right) = -20 \cdot O_{LPF} \cdot \log_{10} \left(\frac{f_{clk}}{2^N \cdot f_C} \right) \tag{12}$$

Since the condition $A_{LPF}(f_{PWM}) < A_{LPF(max)}$ must be accomplished, Equation (13) is obtained from the previous two equations, which is applied to calculate a set of maximum cutoff frequency f_C and filter order O_{LPF} pairs that satisfy the required attenuation. The selected solution is the pair with the minimal O_{LPF} that accomplishes the required $f_{Sampling}$, as expressed in Equation (14)

$$-O_{LPF} \cdot \log_{10} \left(\frac{f_{clk}}{2^N \cdot f_C} \right) < \log_{10} \left(\frac{\pi}{4} \cdot \frac{1}{2^N} \right) \tag{13}$$

$$\left(\frac{f_C}{f_{clk}} \right) < \frac{O_{LPF} \sqrt{\pi}}{2 \left(\frac{2+N+O_{LPF}}{O_{LPF}} \right)}$$

$$\frac{1}{N \cdot t_{Settling}} \geq f_{Sampling} \tag{14}$$

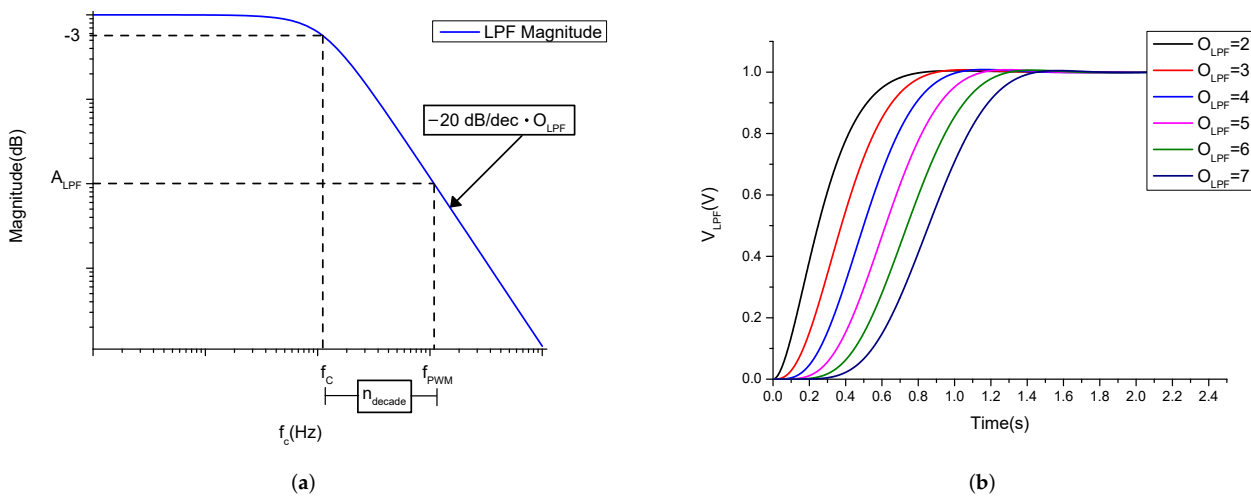


Figure 4. (a) Magnitude plot of the LPF frequency response. (b) Normalized step response at $f_C = 1$ Hz for different O_{LPF} .

3.1.4. FPGA Implementation of an 8-Bit SAR-Based ADC

The FPGA used to test the implementation of a 8-bit SAR-based ADC is from the Zynq-7000 Xilinx family. The clock frequency is $f_{clk} = 100$ MHz and the input/output blocks are configured to $\Delta V_O = 3.3$ V. However, the settings can be changed for others FPGAs following the same methodology. The design parameters are shown in Table 1.

Table 1. Starting parameters for the 8-bit ADC.

Parameters	Value
N	8
f_{CLK}	100 MHz
f_{PWM}	390.625 kHz

Following the steps described in Section 3.1.3, it is calculated the set of solutions (f_C, O_{LPF} pairs) shown in Table 2. The adopted solution should be based on the required $f_{Sampling}$ for the specific application. The $t_{Settling}$ is obtained from normalized LPF step response presented on Figure 4b for the different O_{LPF} .

Table 2. $O_{LPF}, f_C, t_{Settling}$ LPF solutions and $N \cdot t_{Settling}$ and $f_{Sampling}$ ADC performance.

O_{LPF}	LPF f_C	LPF $t_{Settling}$	ADC $t_{Conversion}$	ADC $f_{Sampling}$
2	21.64 kHz	30.04 μ s	240.3 μ s	4.16 kSPS
3	56.76 kHz	17.62 μ s	140.9 μ s	7.10 kSPS
4	91.93 kHz	13.60 μ s	108.8 μ s	9.19 kSPS
5	122.8 kHz	12.38 μ s	99.04 μ s	10.1 kSPS
6	148.9 kHz	11.75 μ s	94.02 μ s	10.6 kSPS

Figure 5 shows graphically the allowed solutions of f_C vs. $f_{Sampling}$ for different O_{LPF} in the 8-bit ADC. Valid solutions are below the maximum cutoff frequency, and solutions that do not accomplish the required attenuation are represented by dashed lines. The values of $f_{Sampling}$ and f_C are normalized at $f_{clk} = 100$ MHz, and, consequently, they can be easily calculated for other clock frequencies by simple multiplication with a ratio R , represented in Equation (15).

$$R = \frac{f_{clk}}{100 \text{ MHz}} \tag{15}$$

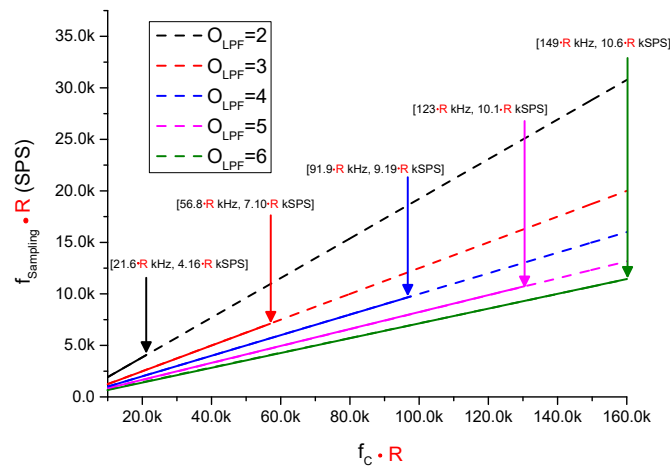


Figure 5. f_C vs. $f_{Sampling}$ for different O_{LPF} for the 8-bit ADC, normalized at $f_{clk} = 100$ MHz.

The same procedure is used for an N-bit ADC. Figure 6a,b shows f_C vs. $f_{Sampling}$ for 10-bit and 12-bit ADCs, respectively.

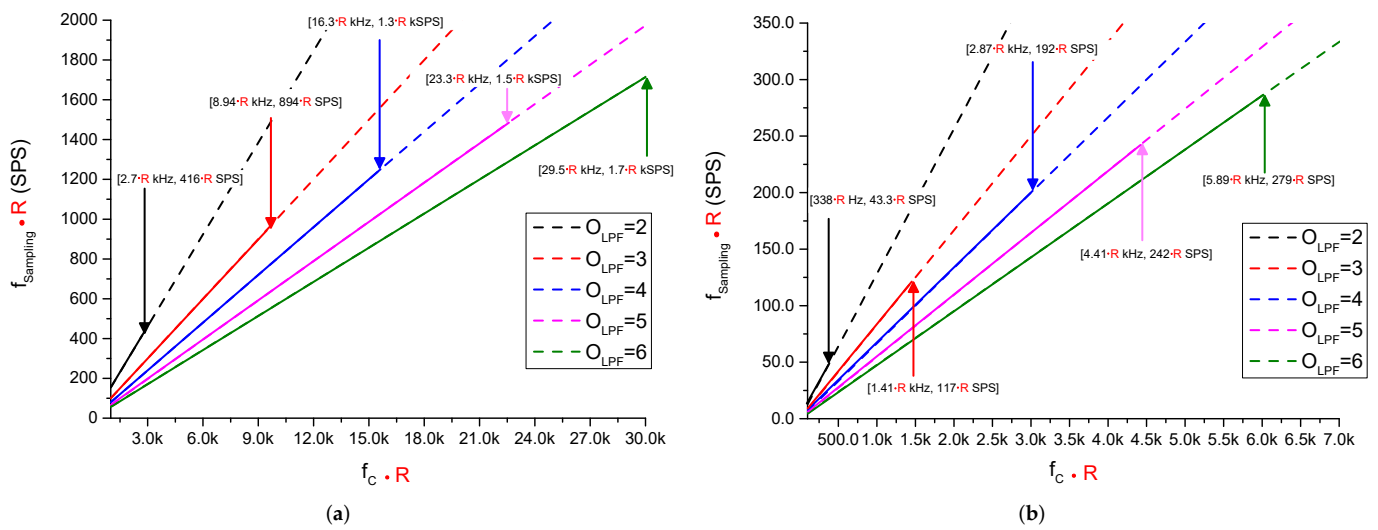


Figure 6. (a) f_C vs. $f_{Sampling}$ for a 10-bit ADC, normalized at $f_{clk} = 100$ MHz. (b) f_C vs. $f_{Sampling}$ for a 12-bit ADC, normalized at $f_{clk} = 100$ MHz.

Figure 7 shows the simulation model of the ADC in Simulink-Matlab and the System Generator tool from Xilinx. The Xilinx tool permits simulation of VHDL hardware descriptions attached to circuit models, such as the Bessel LPF, the analog adder and comparator. Firstly, we provide to the VHDL modules the number of bits (N) and $t_{Settling}$ of the LPF, in order to configure the ADC controller which drives the SAR. N also configures the SAR and PWM modules. The controller periodically starts ADC conversions, initializing the SAR register. It also asserts the bit-acquisition signal when the $t_{Settling}$ is achieved, in order to store the comparator result into the current converted bit of the SAR and storing a '1' to the next conversion bit. The SAR stores the d value, which drives the PWM input in order to generate the corresponding modulated output. The filter gets V_{LPF} from the modulated signal, in order to obtain its averaged value, which is shifted $-\frac{1}{2}LSB$ by the analog adder to V_{DAC} and compared with the V_{in} . After N iterations, the SAR stores the last d value into a register to provide the conversion result (B).

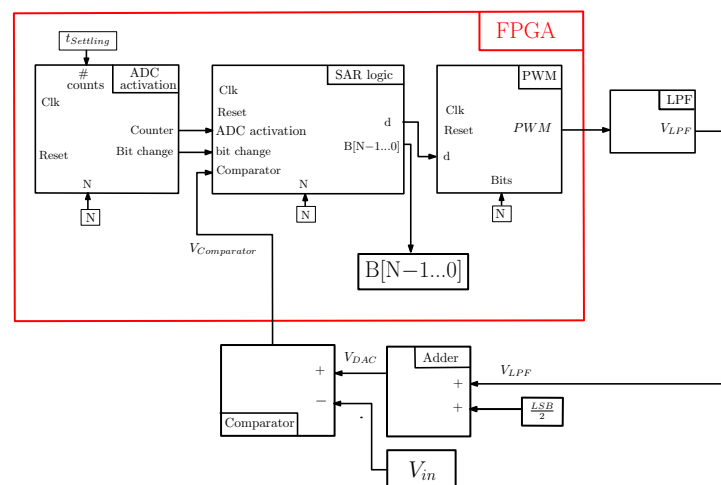


Figure 7. Simulation model.

Figure 8 shows the simulation result for the 8-bit ADC with $O_{LPF} = 6$ and $f_C = 148.9$ kHz solution. We can observe that V_{DAC} starts at the voltage $\frac{\Delta V_0}{2}$ and converges to V_{in} after eight iterations, each one requiring $t_{Settling} = 1.75 \mu s$. The signals driven by the ADC controller start the conversion and the bit change in the SAR.

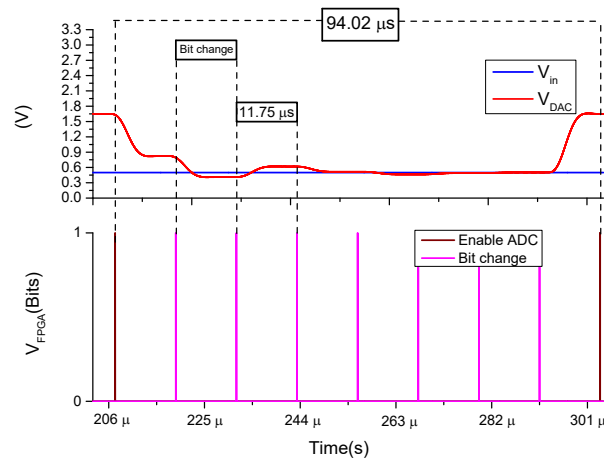


Figure 8. Simulation result of V_{DAC} from a fixed V_{in} .

Finally, Figure 9 depicts the conversion results, where we can see B is correctly obtained for a voltage ramp V_{in} from V_{OL} to V_{OH} .

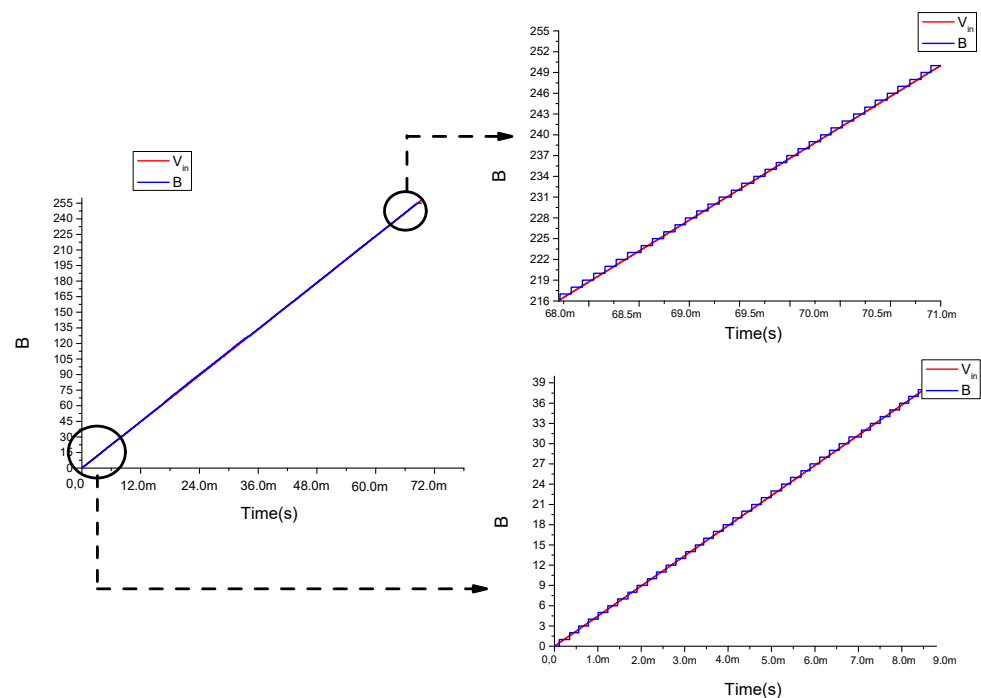


Figure 9. Simulation of conversion result B from voltage ramp V_{in} .

4. Double Data Rate PWM

Double data rate (DDR) registers in FPGAs are usually devoted as interface logic attached to external synchronous dynamic RAM memories, although other applications can use them to double the acquisition rate [53]. Therefore, DDR registers are not available in the internal FPGA fabric, but they are located in the input/output blocks [54,55]. An output DDR (ODDR) flip-flop can drive its output with two different logic levels during a single clock cycle, one logic level synchronized at the rising edge of the clock and the other level at the falling edge.

An easy way to increment $f_{Sampling}$ is by using the input clock as the least significant bit of the internal PWM counter and driving the FSM output through an ODDR flip-flop, doubling the f_{PWM} from Equation (6), as expressed in Equation (16).

$$f_{PWM-DDR} = \frac{2 \cdot f_{clk}}{2^N} = 2 \cdot f_{PWM} \tag{16}$$

Therefore, Equation (13) is modified since f_{PWM} in Equation (10) is increased due to the DDR register, obtaining the solutions f_C and O_{LPF} of the filter in the Equation (17)

$$\left(\frac{f_C}{f_{clk}}\right) < \frac{2 \cdot O_{LPF} \sqrt{\pi}}{2^{\left(\frac{2+N+N \cdot O_{LPF}}{O_{LPF}}\right)}} \tag{17}$$

Figure 10 depicts the architecture of the N-bit DDR PWM. The least significant bit of the counter is the clock signal, and the rest of the counter bits are stored in a register. The FSM generates the two logic levels for the ODDR flip-flop, which synchronously drives the output levels at the rising and falling edges of the clock.

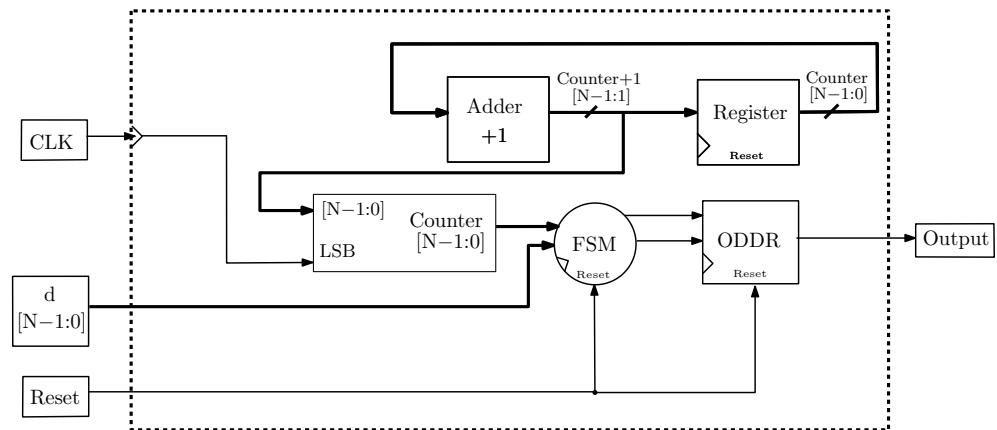


Figure 10. Schematic of the N-bit DDR PWM.

Figure 11 shows allowed solutions of $f_{Sampling}$ and f_C for different O_{LPF} in the 8-bit ADC with a $f_{CLK} = 100$ MHz and DDR-PWM, using Equation (17). Figure 12a,b presents the $f_{Sampling}$ and f_C solutions for 10-bit and 12-bit ADCs, respectively, with a $f_{CLK} = 100$ MHz and DDR-PWM.

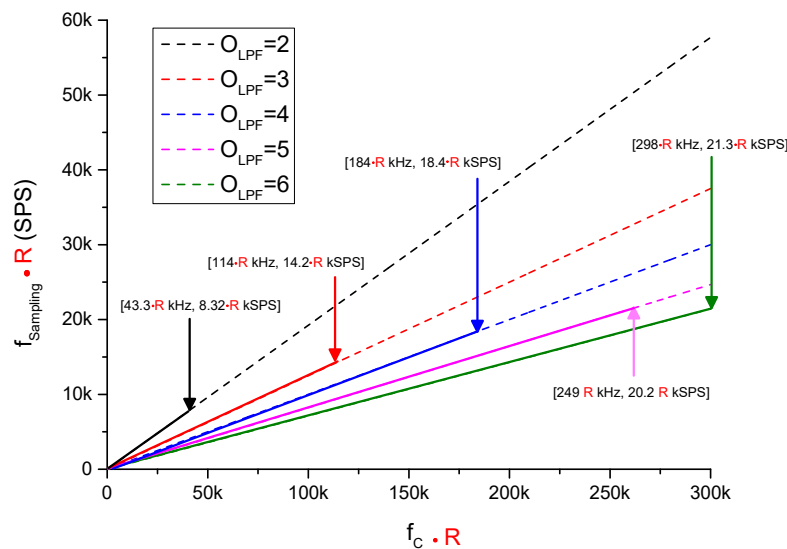


Figure 11. f_C vs. $f_{Sampling}$ for different O_{LPF} for the 8-bit ADC, normalized at $f_{clk} = 100$ MHz with DDR-PWM.

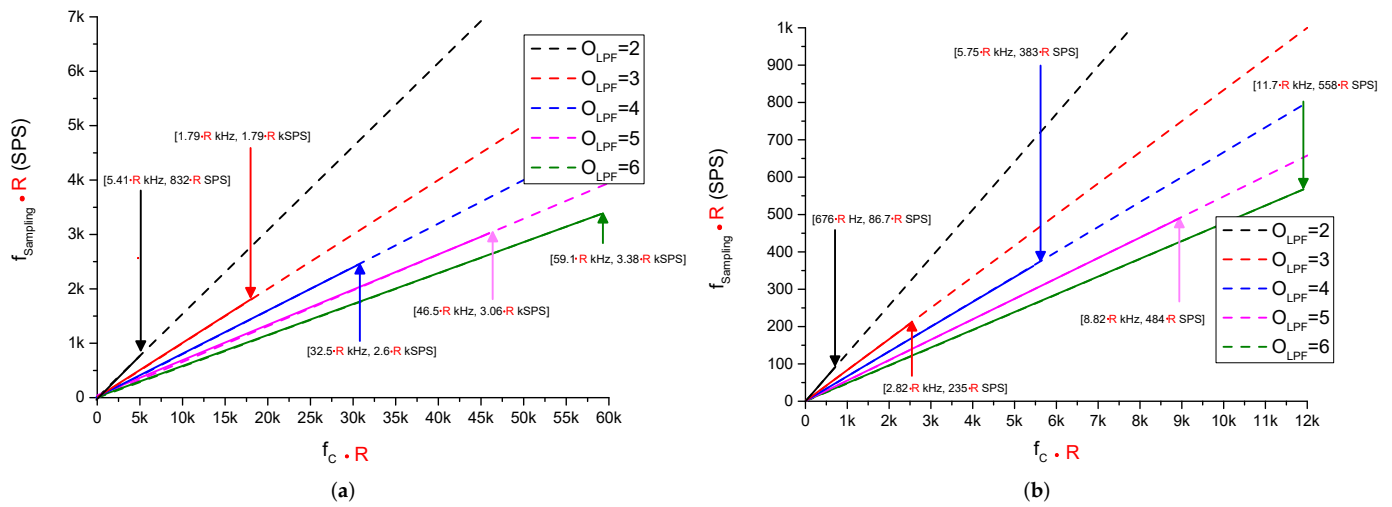


Figure 12. (a) f_C vs. $f_{Sampling}$ for a 10-bit ADC, normalized at $f_{clk} = 100$ MHz with DDR-PWM. (b) f_C vs. $f_{Sampling}$ for a 12-bit ADC, normalized at $f_{clk} = 100$ MHz with DDR-PWM.

5. Paralleled PWM

Another way to enhance the ADC $f_{Sampling}$ is by a parallel design of the PWM to increase the f_{PWM} . The N-bit PWM is divided into a set of K PWM blocks, each one built around an $\frac{N}{K}$ -bit counter. Therefore, the divided PWM reduces greatly the number of required counts, increasing the f_{PWM} from Equation (6) by a factor $2^{\frac{N}{K}(K-1)}$, as expressed in Equation (18).

$$f_{PWM-K} = \frac{f_{clk}}{2^{\frac{N}{K}}} = \frac{f_{clk}}{2^{N \cdot \frac{K-1}{K}}} = 2^{\frac{N}{K}(K-1)} \cdot f_{PWM} \tag{18}$$

Due to the change on the PWM frequency, the Equation (13) has to be modified, as shown on Equation (19).

$$-O_{LPF} \cdot \log_{10} \left(\frac{f_{clk}}{2^{\frac{N}{K}} \cdot f_C} \right) < \log_{10} \left(\frac{\pi}{4} \cdot \frac{1}{2^N} \right) \tag{19}$$

$$\left(\frac{f_C}{f_{clk}} \right) < \frac{O_{LPF} \sqrt{\pi}}{2^{\left(\frac{N}{K} + \frac{N+2}{O_{LPF}} \right)}}$$

Figure 13 shows the paralleled N-bit PWM architecture. The input d of the PWM is divided as shown in Figure 14, where each subset d_j ($0 \leq j \leq K - 1$) is composed of a $\frac{N}{K}$ -bit slice of the input. Each d_j drives a $\frac{N}{K}$ -bit PWM, generating the V_{DAC} from the weighted addition of the filtered PWM outputs.

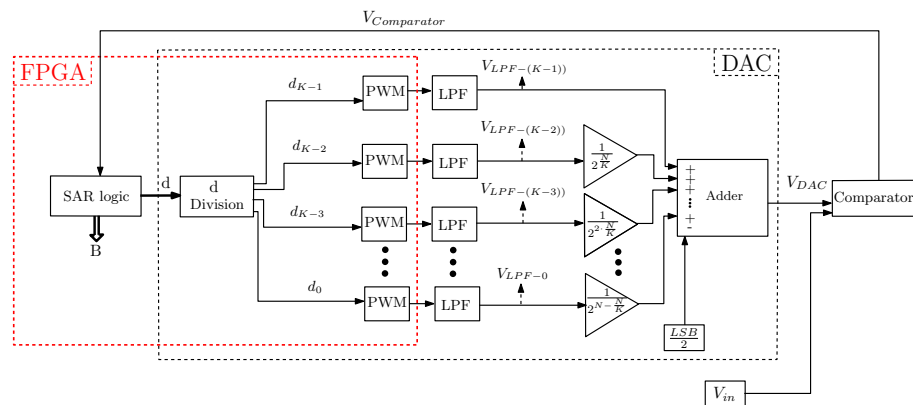


Figure 13. Schematic of the paralleled PWM.

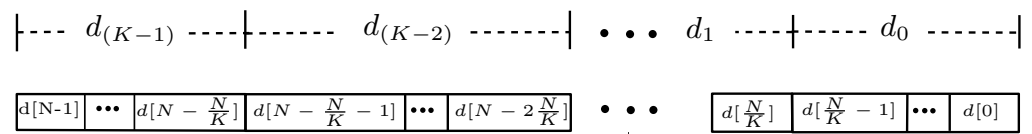


Figure 14. Division of d into a set of K d_j slices.

In order to calculate the V_{DAC} expression, firstly, Equation (20) is obtained by replacing d to the set of d_j from Equations (2)–(4). Then, Equation (7) is applied to compute the continuous coefficient of the filtered voltage for each $\frac{N}{K}$ -bit PWM, as expressed in Equation (21). Finally, Equation (22) is obtained from the previous two equations, which provides the weights of the addition of the V_{DAC} .

$$V_{DAC} = \Delta V_O \left(\frac{d_{K-1} \cdot 2^{\frac{N}{K} \cdot (K-1)}}{2^N} + \dots + \frac{d_1 \cdot 2^{\frac{N}{K}}}{2^N} + \frac{d_0}{2^N} \right) - \frac{LSB}{2}$$

$$V_{DAC} = \Delta V_O \cdot \sum_{j=0}^{K-1} \left(\frac{2^{\frac{N}{K} \cdot j}}{2^N} \cdot d_j \right) - \frac{LSB}{2}$$

$$V_{LPF-j} = A_{0-j} = \Delta V_O \cdot \frac{d_j}{2^{\frac{N}{K}}} \tag{21}$$

$$V_{DAC} = \sum_{j=0}^{K-1} \left(\frac{2^{\frac{N}{K} \cdot (j+1)}}{2^N} \cdot V_{LPF-j} \right) - \frac{LSB}{2} = \sum_{j=0}^{K-1} \left(\frac{2^{\frac{N}{K} \cdot j}}{2^{N-\frac{N}{K}}} \cdot V_{LPF-j} \right) - \frac{LSB}{2}$$

$$V_{DAC} = V_{LPF-(K-1)} + \sum_{j=0}^{K-2} \left(\frac{2^{\frac{N}{K} \cdot j}}{2^{N-\frac{N}{K}}} \cdot V_{LPF-j} \right) - \frac{LSB}{2}$$

Although the previous description applies when the division $\frac{N}{K}$ is an integer value, it can be extended if it is not the case. If we denote P_1 and P_2 as integers that accomplish $N = P_2 + P_1 \cdot (K - 1)$, the distributed PWM is composed of one P_2 -bit PWM and a set of $K - 1$ P_1 -bit PWMs. Assuming the P_2 -bit PWM is associated to the $(K - 1)$ LPF, the weighted addition is expressed in Equation (23).

$$V_{DAC} = \frac{2^{P_2+P_1 \cdot (K-1)}}{2^N} \cdot V_{LPF-(K-1)} + \sum_{j=0}^{K-2} \left(\frac{2^{P_1 \cdot (j+1)}}{2^N} \cdot V_{LPF-j} \right) - \frac{LSB}{2}$$

$$V_{DAC} = V_{LPF-(K-1)} + \sum_{j=0}^{K-2} \left(\frac{2^{P_1 \cdot (j+1)}}{2^N} \cdot V_{LPF-j} \right) - \frac{LSB}{2}$$

As K is increased, the f_C can be increased to improve $f_{Sampling}$, and the O_{LPF} can be decreased to reduce the filter complexity, at the expense of augmenting the numbers of devoted FPGA pins and filters. The extreme case is when $K = N$, which provides a set of K 1-bit PWMs as the solution. Nevertheless, PWMs and LPFs are unnecessary in such a case since the weighted addition of the d_j is equivalent to a R-2R ladder DAC [56,57].

Figure 15 shows the allowed solutions of $f_{Sampling}$ and f_C for different O_{LPF} in the 8-bit ADC with $f_{CLK} = 100$ MHz and the paralleled PWM with $K = 2$. Figure 16a,b shows $f_{Sampling}$ solutions for a 10-bit and 12-bit ADCs, respectively, with the same parameters.

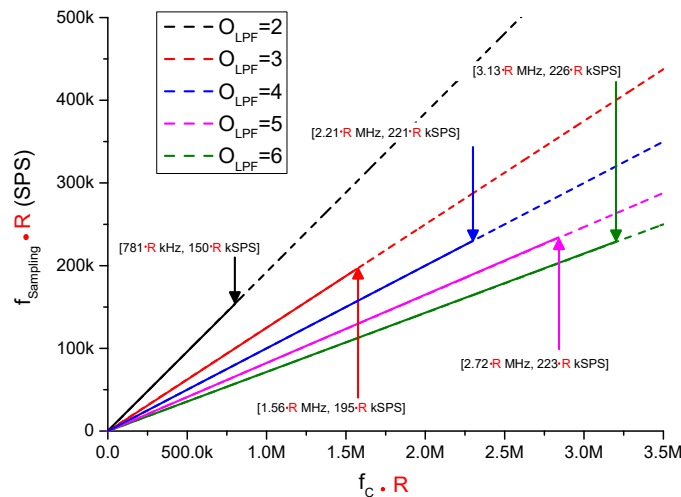


Figure 15. f_c vs. $f_{Sampling}$ for different O_{LPF} for the 8-bit ADC with 2 4-bit PWM, normalized at $f_{clk} = 100$ MHz.

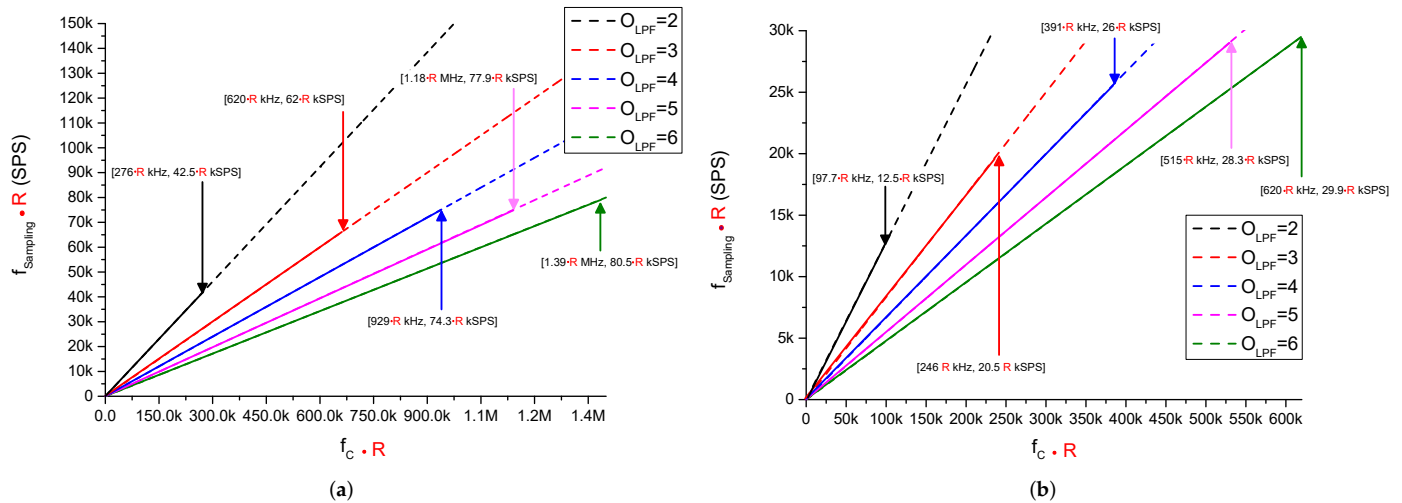


Figure 16. (a) f_c vs. $f_{Sampling}$ for a 10-bit ADC with 2 5-bit PWM, normalized at $f_{clk} = 100$ MHz (b) f_c vs. $f_{Sampling}$ for a 12-bit ADC with 2 6-bit PWM, normalized at $f_{clk} = 100$ MHz.

6. Distributed Duty-Cycle PWM

This section presents an optimization focused on enhancing the f_c , instead of incrementing the f_{PWM} directly as in the previous optimizations.

The filtered PWM signal must ensure $\Delta V_{LPF} < LSB$ at f_{PWM} to properly perform the comparison against V_{in} . The ΔV_{LPF} is dominated by the first harmonic of the PWM output, which is maximal when $D = \frac{1}{2}$. Reducing ΔV_{LPF} can be used to increase the f_c , which reduces the $t_{Settling}$ for higher performance. Additionally, the filter order can also be reduced to provide a simpler implementation. Distributed duty cycle PWMs were implemented in applications as NPC inverters or adjustable losses distribution [58–60], but not in ADCs to reduce ΔV_{LPF} .

For the sake of clarity, Figure 17a shows a 3-bit PWM at $D = \frac{1}{2}$; consequently, during one half of the period count, the signal is asserted. In order to significantly reduce ΔV_{LPF} , the distributed duty cycle PWM (DDC-PWM) minimizes the elapsed time between assertions, distributing the assertions during the 2^N counts that compose a PWM period, as shown in Figure 17b–d. for $D = \frac{2}{8}, \frac{3}{8}$ and $\frac{4}{8}$, respectively.

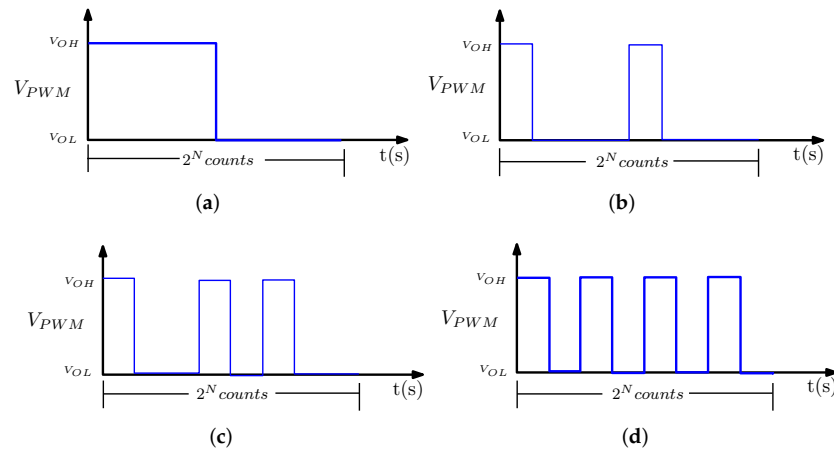


Figure 17. (a) 3-bit PWM at $D = \frac{4}{8}$. (b) 3-bit DDC-PWM at $D = \frac{2}{8}$. (c) 3-bit DDC-PWM at $D = \frac{3}{8}$. (d) 3-bit DDC-PWM at $D = \frac{4}{8}$.

Figure 18 shows the filtered output V_{LPF} comparison for a 3-bit PWM and DDC-PWM at $D = \frac{3}{8}$ using the same LPF parameters, showing that the ΔV_{LPF} is significantly reduced.

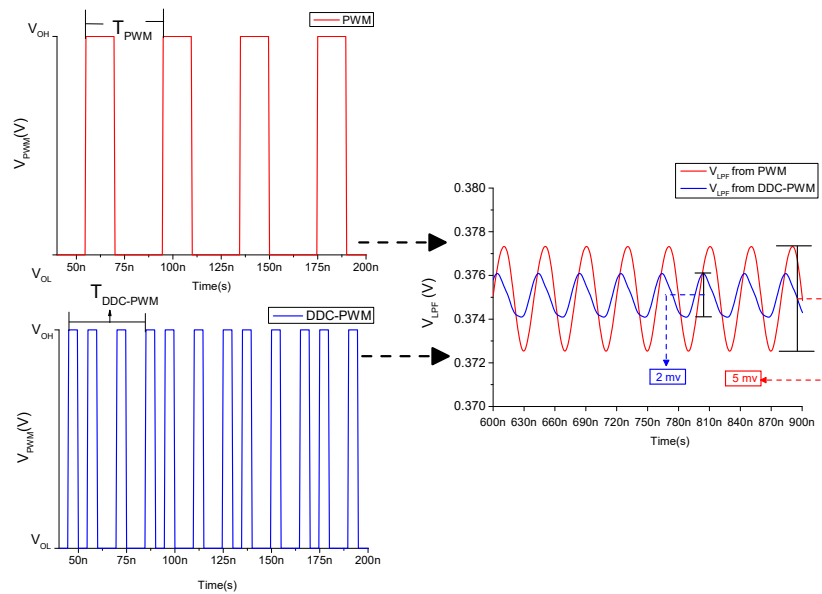


Figure 18. V_{LPF} on a 3-bit PWM and 3-bit DDC-PWM, $V_{OH} = 1\text{ V}$, $D = \frac{3}{8}$, $O_{LPF} = 3$ and $f_c = 4\text{ MHz}$.

The distribution of the duty cycle increments the modulation frequency of the PWM ($f_{PWM-DDC}$) [61–64]. The maximum $f_{PWM-DDC}$ is achieved at the condition $D = \frac{1}{2}$, which minimizes the first harmonic A_1 of the PWM output. Contrary to the PWM, which achieves the maximum A_1 at $D = \frac{1}{2}$, the maximum A_1 in the DDC-PWM is achieved at the condition $D = \frac{1}{2^N}$ or $D = \frac{2^N - 1}{2^N}$, which equals the $f_{PWM-DDC}$ to f_{PWM} [65–67]. Applying Equation (7) at the condition $D = \frac{1}{2^N}$, the maximum first harmonic for DDC-PWM is obtained in Equation (24).

$$A_{1-DDC}(max) = 2 \cdot \frac{\Delta V_O}{\pi} \cdot \sin\left(\frac{\pi}{2^N}\right) \tag{24}$$

The upper limit for the LPF gain is obtained at $f_{PWM-DDC} = f_{PWM}$ in Equation (25).

$$A_{LPF}(f_{PWM-DDC}) < \frac{\pi}{4} \cdot \frac{1}{2^N} \cdot \frac{1}{\sin\left(\frac{\pi}{2^N}\right)} \tag{25}$$

Following the same steps that were applied in Equation (13), filter solutions for the DDC-PWM are calculated in Equation (26)

$$-O_{LPF} \cdot \log_{10}\left(\frac{f_{clk}}{2^N \cdot f_C}\right) < \log_{10}\left(\frac{\pi}{4} \cdot \frac{1}{2^N} \cdot \frac{1}{\sin\left(\frac{\pi}{2^N}\right)}\right)$$

$$\left(\frac{f_C}{f_{clk}}\right) < \frac{O_{LPF} \sqrt{\pi}}{\left[2^{\left(\frac{2+N+O_{LPF}}{O_{LPF}}\right)}\right] \cdot \left[\left(\sin\left(\frac{\pi}{2^N}\right)\right)^{\left(\frac{1}{O_{LPF}}\right)}\right]}$$
(26)

The previous equation can be simplified by applying the small-angle approximation for sine, $\sin \Theta \approx \Theta$, leading to Equation (27)

$$\left(\frac{f_C}{f_{clk}}\right) < \frac{1}{2^{\left(\frac{2+N+O_{LPF}}{O_{LPF}}\right)}}$$
(27)

Figures 19 and 20a,b show allowed solutions of $f_{Sampling}$ and f_C for different O_{LPF} in the 8-bit, 10-bit and 12-bit ADCs, respectively, with a $f_{CLK} = 100$ MHz and using DDC-PWM.

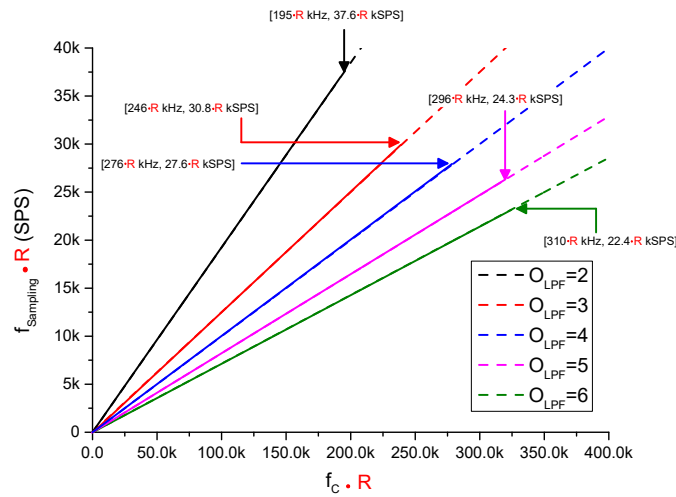


Figure 19. f_C vs. $f_{Sampling}$ for different O_{LPF} for the 8-bit ADC with DDC-PWM, normalized at $f_{clk} = 100$ MHz.

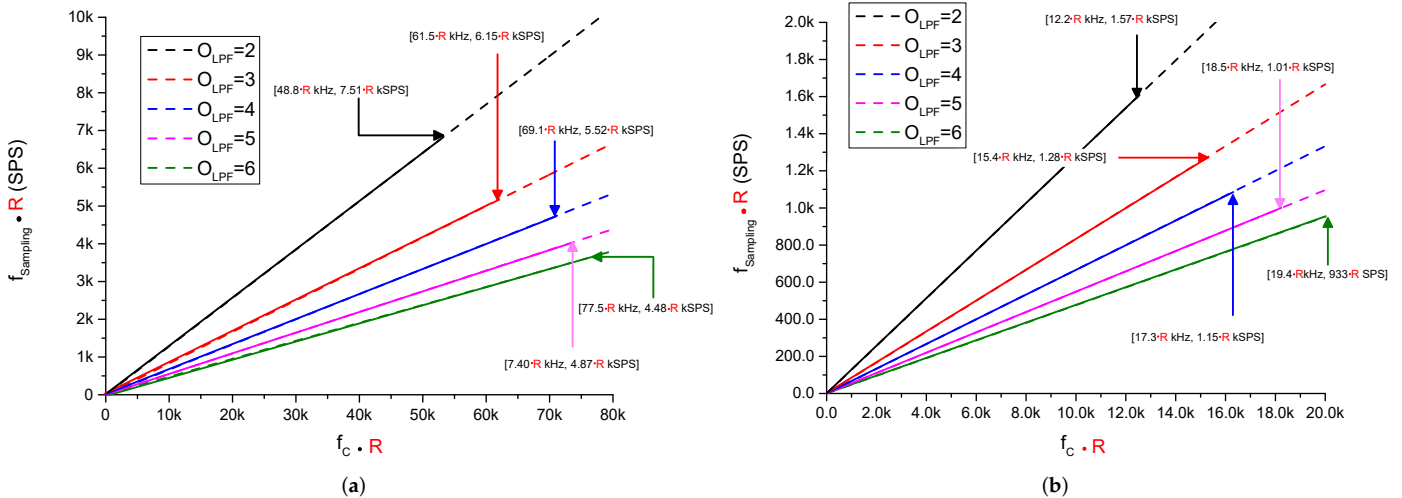


Figure 20. (a) f_C vs. $f_{Sampling}$ for a 10-bit ADC, normalized at $f_{clk} = 100$ MHz with DDC-PWM (b) f_C vs. $f_{Sampling}$ for a 12-bit ADC, normalized at $f_{clk} = 100$ MHz with DDC-PWM.

7. Paralleled DDR and DDC PWM

The distributed duty cycle PWM can be combined with the paralleled PWM and the DDR optimizations to enhance the $f_{Sampling}$ of the ADC. Combining the DDC-PWM and the DDR-PWM, Equation (27) is modified to Equation (28), which can be used to increment the $f_{Sampling}$ by augmenting the f_C of the filter.

$$\left(\frac{f_C}{f_{clk}}\right) < \frac{2}{2^{\left(\frac{2+N \cdot O_{LPF}}{O_{LPF}}\right)}} \tag{28}$$

The DDC-PWM can also be combined with the paralleled PWM. The DDC-PWM can be implemented from a set of $K \frac{N}{K}$ -bit DDC-PWMs to improve the $f_{Sampling}$. The maximum magnitude of the first harmonic is reduced from Equation (24), to Equation (29). This will lead to the improvement of the LPF solutions, according to Equation (30).

$$A_{1-DDC(max)} = 2 \cdot \frac{\Delta V_O}{\pi} \sin\left(\frac{\pi}{2\left(\frac{N}{K}\right)}\right) \tag{29}$$

$$\left(\frac{f_C}{f_{clk}}\right) < \frac{2^{\left(\frac{N}{O_{LPF} \cdot K}\right)}}{2^{\left(\frac{N+2}{O_{LPF}} + \frac{N}{K}\right)}} \tag{30}$$

Finally, Equation (31) shows the LPF solutions when there are applied the three optimizations together.

$$\left(\frac{f_C}{f_{clk}}\right) < \frac{2^{\left(1+O_{LPF} \cdot K\right)}}{2^{\left(\frac{N+2}{O_{LPF}} + \frac{N}{K}\right)}} \tag{31}$$

The resolution and sampling frequency of a set of ADCs on a target FPGA can be individually adjusted to the requirements of the application. This work introduces different optimization on the PWM module to improve the ADC $f_{Sampling}$ performance. The adoption of DDR and DDC-PWM enhances the ADC $f_{Sampling}$ performance without requiring additional external components for the filters and the analog comparator; however, they devote more internal FPGA resources. On the contrary, the paralleled PWM increases the number of external components and devoted FPGA pins, but the $f_{Sampling}$ performance can be greatly enhanced.

Figure 21a,b shows the LPF solutions for a fixed filter order $O_{LPF} = 3$ and $O_{LPF} = 5$, respectively, in order to implement an 8-bit ADC. The best $f_{Sampling}$ is obtained by combining DDR-PWM and DDC-PWM on a non-paralleled ($K = 1$) or paralleled PWM ($K = 2, K = 4$).

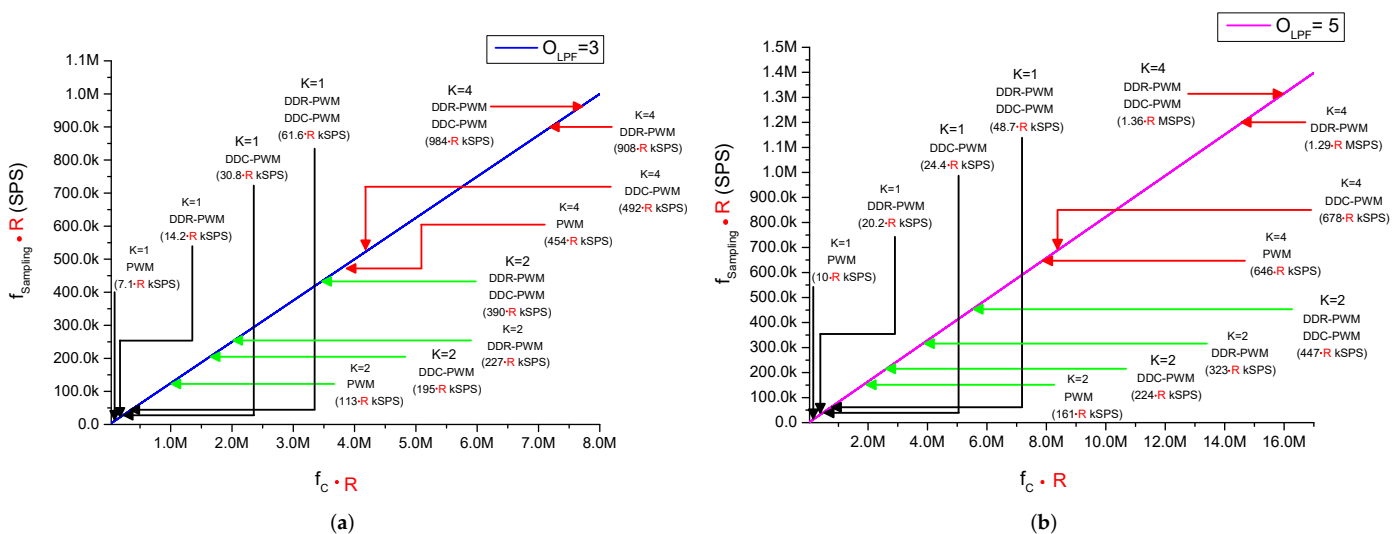


Figure 21. (a) f_C vs. $f_{Sampling}$ normalized for a fixed filter order $O_{LPF} = 3$ on the 8-bit ADC (b) f_C vs. $f_{Sampling}$ normalized for a fixed filter order $O_{LPF} = 5$ on the 8-bit ADC.

The paralleled PWM greatly enhances the ADC $f_{Sampling}$ since it leads to obtain the LPF solution with lower f_C . However, the set of $K \frac{N}{K}$ -bit PWMs requires $K + 1$ FPGA input/output and additional external components for the filters. Table 3 shows the maximum $f_{Sampling}$ achieved and the external components required to implement the 8-bit ADC for different K , using fixed order ($O_{LPF} = 3$ and $O_{LPF} = 5$) filters. Implementing the $O_{LPF} = 3$ filter requires an operational amplifier (OA), three resistors (R) and capacitors (C) attached to a FPGA output, while for implementing the $O_{LPF} = 5$ filter, two operational amplifiers, five resistors and capacitors are required. The additional OA and $K + 3$ resistors are devoted to implement the analog adder for the filtered outputs shift voltage $-\frac{1}{2}LSB$, and the comparator attached to a FPGA input.

Table 3. Comparison for the implementation of an 8-bit ADC with different O_{LPF} using paralleled DDC-PWM and DDR-PWM.

ADC = 8-Bit	DDC-PWM DDR-PWM					
	1	2	4	1	2	4
K	1	2	4	1	2	4
O_{LPF}	1	3	4	5	5	4
$f_{Sampling}$ (kSPS)	61.5	390.6	984.3	48.3	447	1360
FPGA in/out	2	3	4	2	3	4
R	7	11	15	9	15	21
C	3	6	9	5	10	15
OA	2	3	4	3	5	7
Comparators		1			1	

8. Experimental Results

The experimental results shown were obtained implementing 8-bit ADCs on a FPGA from Xilinx Zynq-7000 driven by $f_{clk} = 100$ MHz, with the output ports configured to obtain $\Delta V_O = 3.3$ V ($LSB = 12.89$ mV). The FPGA development board is a ZedBoard Zynq 7000, which is connected with an external prototyping board containing the filters and the analog adder and comparator. The LPFs and adder are implemented using LT1364 operation amplifiers from Analog Devices [68] due to their bandwidth and slew-rate features. To implement the comparator, we used the MAX9691EPA from Analog Devices due to the fast response.

8.1. Implementation of an 8-Bit ADC with 8-Bit DDR-PWM and LPF

The desired $f_{Sampling}$ is 8 kSPS; f_C is obtained from Equation (17) with minimum O_{LPF} . The parameter values of Figure 22 show solutions of $f_{Sampling}$ and f_C with different O_{LPF} . Figure 22 is the same as Figure 11, but with the addition of the 8 kSPS restriction. The ADC obtained is shown in Table 4. Figure 23a,b V_{DAC} represents the experimental results for $V_{in} = 1.2$ V and $V_{in} = 2.2$ V, respectively.

Table 4. Implementation characteristics of an 8-bit ADC with DDR-PWM.

ADC = 8-bit	DDR-PWM
f_C	43.3 kHz
$t_{Settling}$	15 μ s
O_{LPF}	2
$f_{Sampling}$	8 kSPS
FPGA in/out	2
R	6
C	2
OA	2
Comparators	1

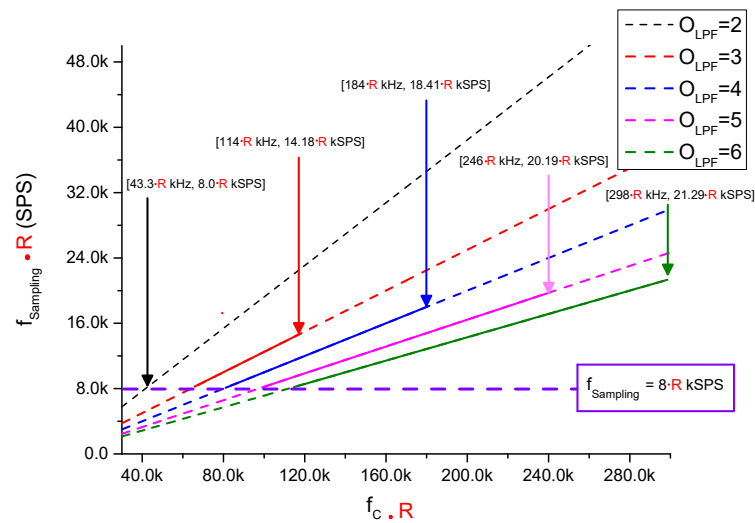


Figure 22. f_c vs. $f_{Sampling}$ for a LPF with different O_{LPF} implementing an $f_{Sampling} = 8$ kSPS 8-bit ADC with DDR-PWM and $f_{clk} = 100$ MHz.

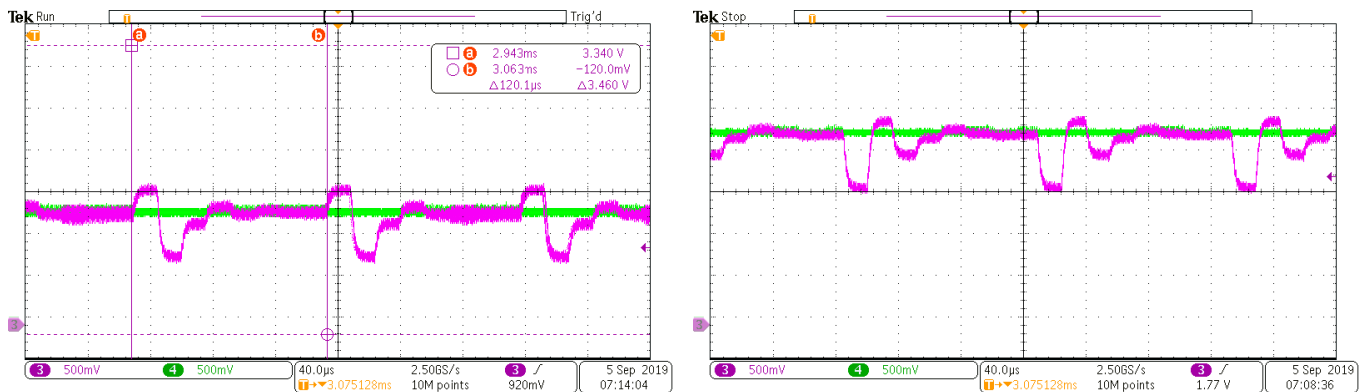


Figure 23. (a) V_{DAC} from a fixed $V_{in} = 1.2$ V (b) V_{DAC} from a fixed $V_{in} = 2.2$ V.

To observe B for every possible V_{in} value, a triangular signal of V_{in} is generated from 0 V to 3.3 V, with a slope that increments LSB during the conversion time ($N \cdot t_{Settling}$). Figure 24 shows how B (green signal) follows the triangular signal V_{in} (blue signal). To show B , an extra DAC is implemented, and once the conversion is done, B is sent to the DAC.

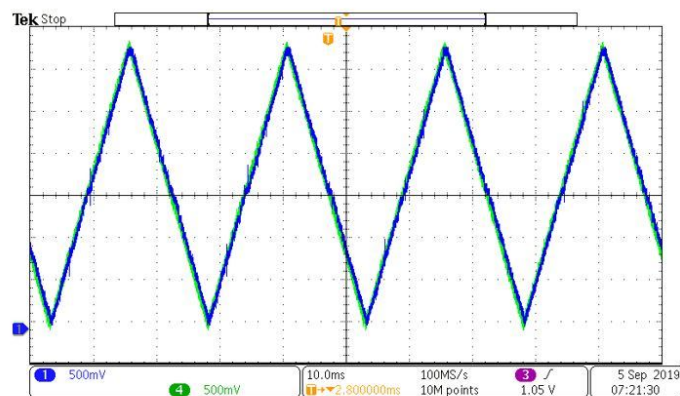


Figure 24. B from voltage ramp V_{in} with an 8-bit ADC, implemented with a DDR-PWM.

8.2. Implementation of an 8-Bit ADC with 2 4-Bit DDC-PWM and DDR-PWM

An 8-bit ADC with $f_{Sampling} = 230$ kSPS is design using DDC-PWM and DDR-PWM. The implementation schematic is presented in Figure 25 where an extra DAC is added to show the obtained B . To implement the filters and the adder, the LT1364 from Analog Devices is used. Figure 26 shows the solutions of $f_{Sampling}$ and f_C with different O_{LPF} . For designing the LPF, the minimum O_{LPF} that accomplishes the desired $f_{Sampling}$ is chosen. $O_{LPF} = 2$ is chosen to implement the LPF, and the parameters values are shown in Table 5.

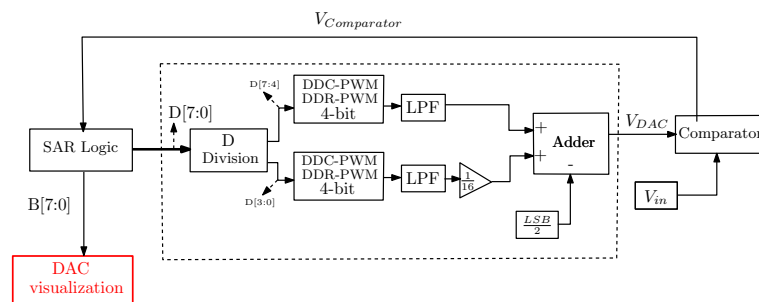


Figure 25. Implementation schematic of an 8-bit ADC with 2 4-bit DDC-PWM and DDR-PWM.

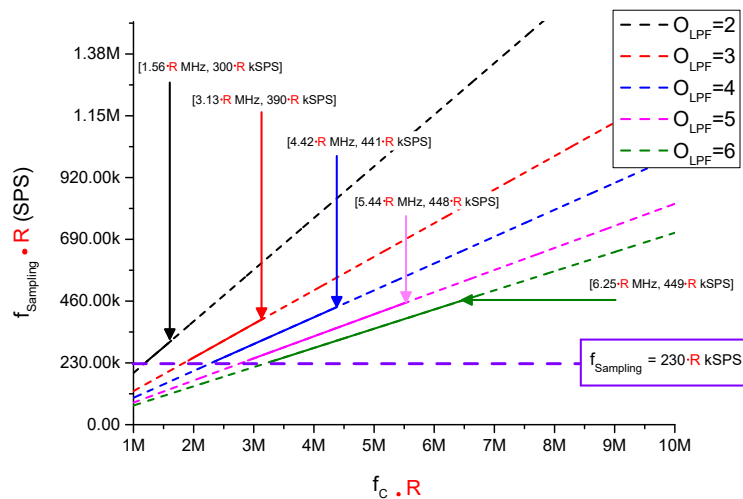


Figure 26. f_C vs. $f_{Sampling}$ for a LPF with different O_{LPF} implementing an $f_{Sampling} = 230$ kSPS 8-bit ADC with 2 4-bit DDC-PWM and DDR-PWM with $f_{clk} = 100$ MHz.

Figure 27 shows V_{DAC} behavior for a fixed V_{in} of 2.2 V. Figure 28 shows B (pink signal) for every possible V_{in} (blue signal).

Table 5. Implementation parameters of an 8-bit ADC with 2 4-bit DDC-PWM and DDR-PWM.

ADC = 8-Bit	DDR-PWM DDC-PWM
f_C	1.20 MHz
$t_{Settling}$	543 ns
O_{LPF}	2
$f_{Sampling}$	230 kSPS
FPGA in/out	3
R	9
C	4
OA	3
Comparators	1

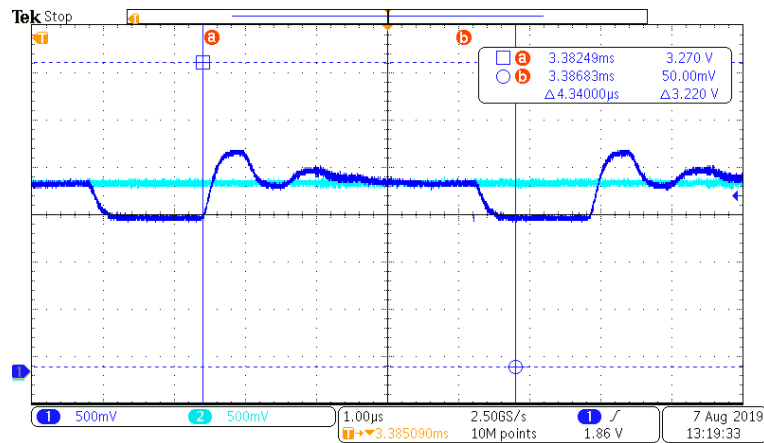


Figure 27. V_{DAC} from a fixed $V_{in} = 2.2$ V, DAC with 2 4-bit DDC-PWM and DDR-PWM.

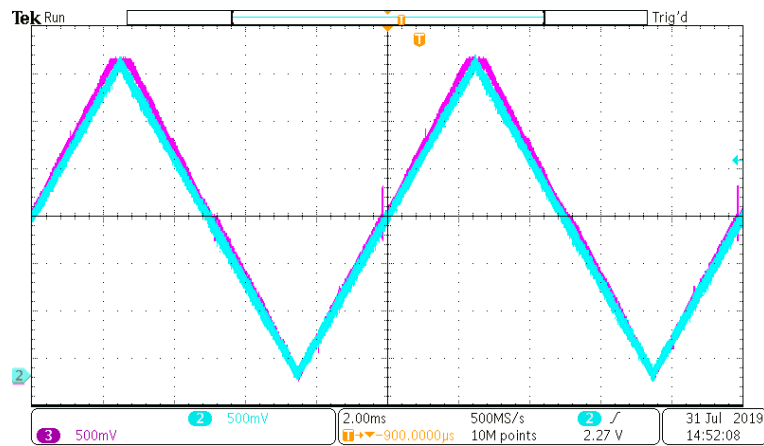


Figure 28. B from voltage ramp V_{in} with an 8-bit ADC, implemented with 2 4-bit DDC-PWM and DDR-PWM.

8.3. Implementation of an 8-Bit ADC with 4 2-Bit DDC-PWM and DDR-PWM

An 8-bit ADC with $f_{Sampling} = 500$ kSPS is designed, using 4 2-bit DDC-PWM and DDR-PWM. The schematic is presented in Figure 29 with the extra DAC for visualization. The LPF and adder will be implemented with the OA LT1364. Figure 30 shows the LPF parameters found out, where it can be seen that with an $O_{LPF} = 2$ the desired $f_{Sampling}$ can be accomplished. Figure 31a,b shows V_{DAC} for a fixed V_{in} of 1 V and 1.8 V, respectively. Figure 32 shows B (pink signal) for every possible value of V_{in} (blue signal).

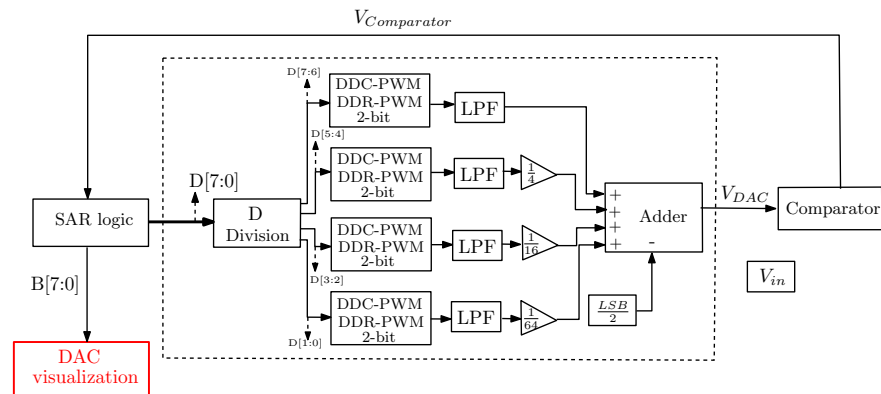


Figure 29. Implementation schematic of an 8-bit ADC with 4 2-bit DDC-PWM and DDR-PWM.

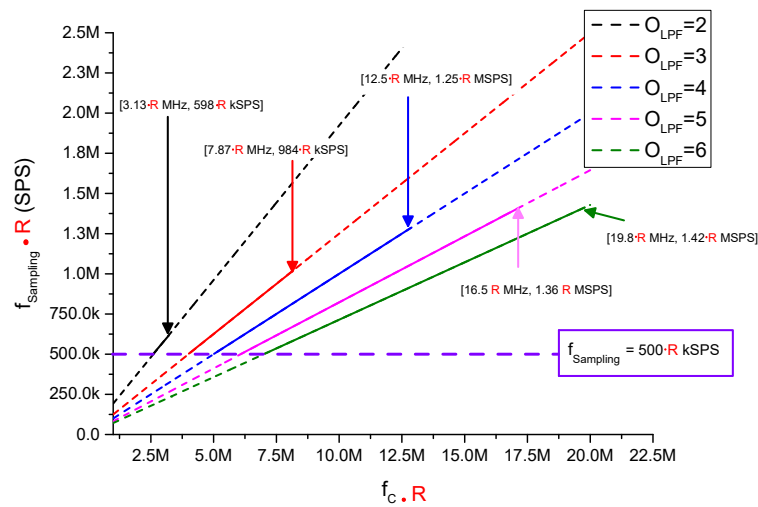


Figure 30. f_c vs. $f_{Sampling}$ for a LPF with different O_{LPF} implementing an $f_{Sampling} = 500$ kSPS 8-bit ADC with 2 4-bit DDR-PWM and DDC-PWM with $f_{clk} = 100$ MHz.

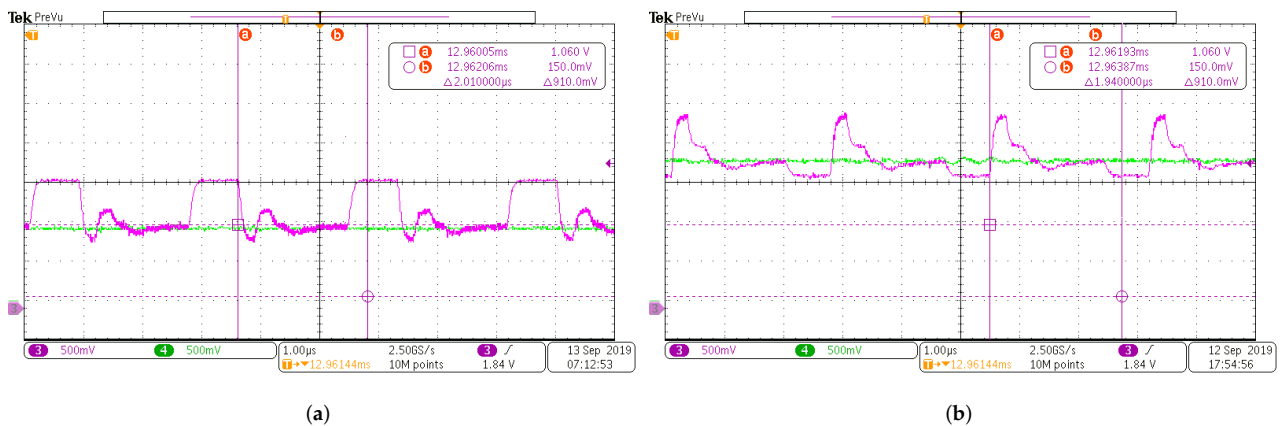


Figure 31. (a) V_{DAC} from a fixed $V_{in} = 1$ V, DAC with 4 2-bit DDC-PWM and DDR-PWM (b) V_{DAC} from a fixed $V_{in} = 1.8$ V, DAC with 4 2-bit DDC-PWM and DDR-PWM.

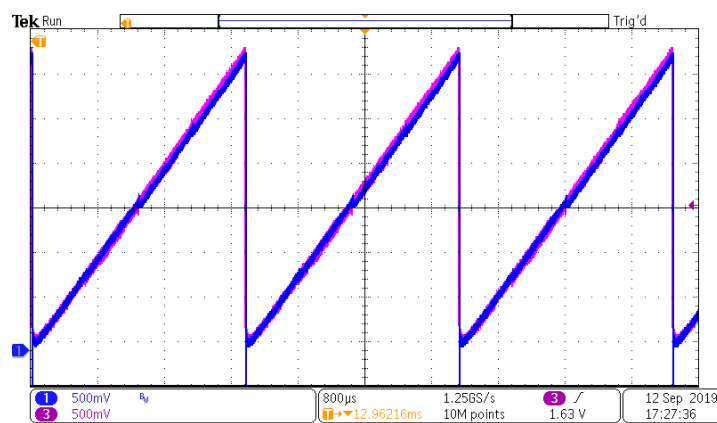


Figure 32. B from voltage ramp V_{in} with an 8-bit ADC, implemented with 4 2-bit DDC-PWM and DDR-PWM.

9. Discussion

One of the main characteristic from the ADCs are DNL (differential non-linearity) and INL (integral non-linearity). The DNL is defined as the difference in the step width between the ADC response and the ideal response, while the INL is defined as the deviation of the ADC response curve from the ideal response curve, both are expressed in LSB. To calculate

and simulate the DNL from the proposed ADC, the 8-bit ADC with $K = 4$ DDC-PWM and DDR-PWM presented in the last section is selected. The DNL is showed on Figure 33a, while Figure 33b shows the INL. Figure 33c,d show DNL and INL, respectively, for an 8-bit ADC with $K = 2$ DDC-PWM and DDR-PWM.

Another characteristic from the ADCs is the ENOB. The ENOB is an indicator of the ADC resolution at a specific input frequency and $f_{Sampling}$. To calculate ENOB, Equation (32) is used, where SNDR is the signal-to-noise-and-distortion ratio as showed in [38].

$$ENOB = \frac{SNDR - 1.76}{6.02} \quad (32)$$

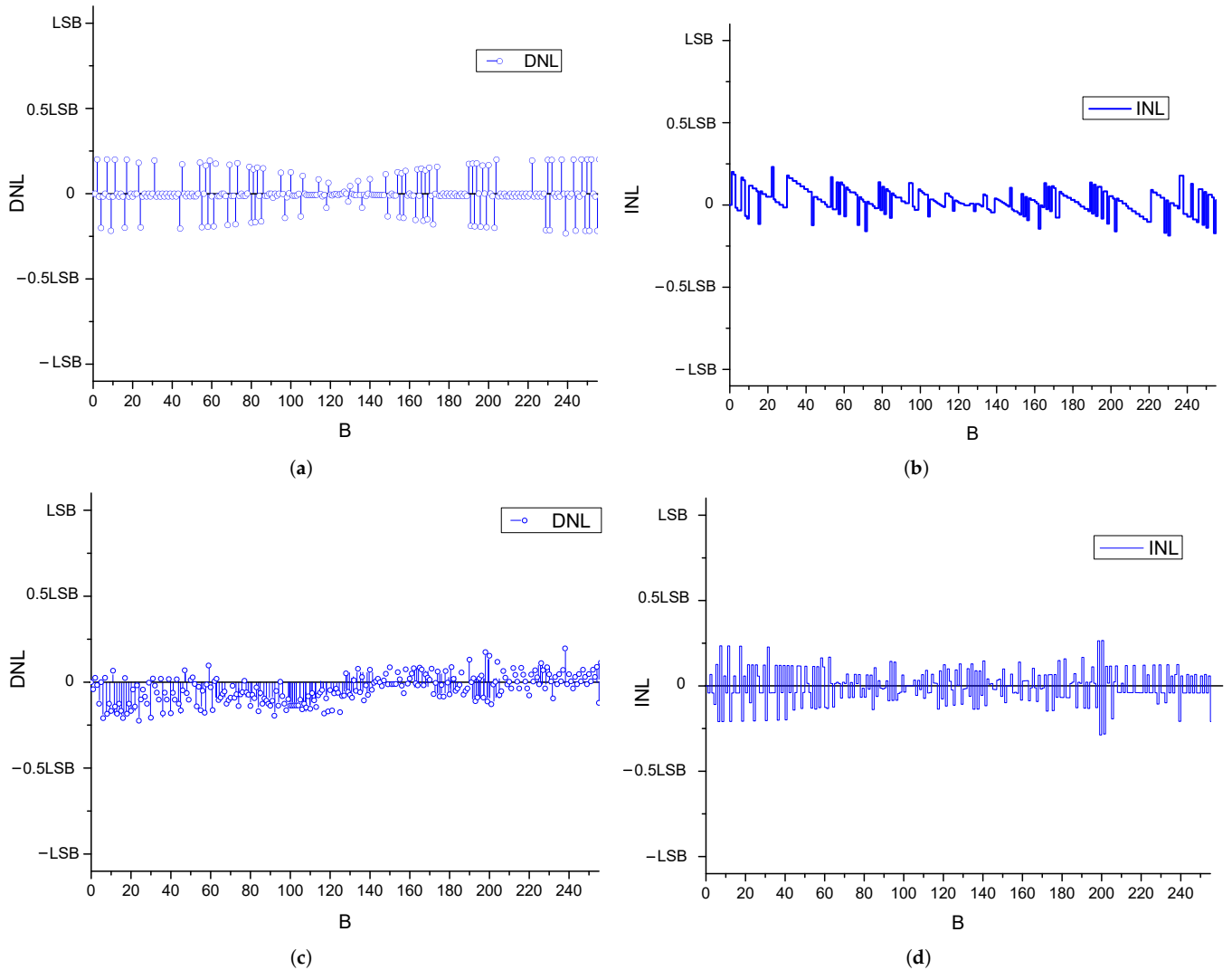


Figure 33. (a) 8-bit ADC with $K = 4$ DDC-PWM and DDR-PWM DNL (b) 8-bit ADC with $K = 4$ DDC-PWM and DDR-PWMM INL (c) 8-bit ADC with $K = 2$ DDC-PWM and DDR-PWM DNL (d) 8-bit ADC with $K = 2$ DDC-PWM and DDR-PWMM INL.

Table 6 compares related ADC from the state of the art with the proposed ADC, using $K = 4$ and $K = 2$. The proposed ADCs, even though they do not provide high sampling rates, are parametrizable and portable to any FPGA family. Table 7 compares the proposed ADC with different resolution (10-bit and 12-bit) with different implementation method ($K = \frac{N}{2}$ and $K = 2$).

Table 6. Comparison of ADCs from the state of the art and the proposed 8-bit ADC ($K = 4, K = 2$).

	[35]	[38]	[40]	$K = 2$ DDC-PWM DDR-PWM	$K = 4$ DDC-PWM DDR-PWM
Resolution (bit)	7.2	9.3	8	8	8
Signal input	Voltage	Voltage	Voltage/current	Voltage	Voltage
Voltage range (V)	0–2.5	0.15–0.45	0–1	0–3.3	0–3.3
Architecture	TDC	TDC	SAR	SAR	SAR
$f_{Sampling}$	200 MSPS	600 MSPS	2.8 kSPS/2.2 kSPS	450 kSPS	1.42 MSPS
DNL (LSB)	−0.9 to 1.4	±0.9	±1.9	±0.28	±0.28
INL (LSB)	−1.1 to 1.6	−1.1 to 0.9	±1.5	±0.25	±0.32
ENOB	6	7	6.4/6.7	7.22	7.24
Calibration	Yes	Yes	No	No	No
Easily portable	No	No	No	Yes	Yes
External elements	No	No	No	Yes	Yes
Technology	FPGA Spartan 6 families	FPGA UltraScale+	CMOS standard cells	FPGA	FPGA

Table 7. Comparison of the proposed 10-bit ADC ($K = 5, K = 2$) and 12-bit ADC ($K = 6, K = 2$).

	$K = 2$ DDC-PWM DDR-PWM	$K = 5$ DDC-PWM DDR-PWM	$K = 2$ DDC-PWM DDR-PWM	$K = 6$ DDC-PWM DDR-PWM
Resolution (bit)	10	10	12	12
Signal input	Voltage	Voltage	Voltage	Voltage
Voltage range (V)	0–3.3	0–3.3	0–3.3	0–3.3
Architecture	SAR	SAR	SAR	SAR
$f_{Sampling}$	160 kSPS	910 kSPS	59.7 kSPS	602 kSPS
DNL (LSB)	±0.42	±0.46	±0.51	±0.49
INL (LSB)	±0.38	±0.4	±0.47	±0.46
ENOB	9.3	9.3	11.1	11.1
Calibration	No	No	No	No
Easily portable	Yes	Yes	Yes	Yes
External elements	Yes	Yes	Yes	Yes
Technology	FPGA	FPGA	FPGA	FPGA

10. Conclusions

This work presents an scalable and parametrizable implementation architecture for analog-to-digital converters (ADC) on field programmable gate array (FPGA) devices. The ADC is based on successive approximation register (SAR) and the digital-to-analog converter (DAC) is implemented by a pulse-width modulator (PWM) and a low-pass filter (LPF). The paper presents a systematic method to implement the ADC based on the FPGA parameters as clock frequency (f_{clk}) and FPGA output voltage (ΔV_O) and allowing the designer to find different solutions to implement the LPF that meet the ADC resolution bits (N) and sampling frequency ($f_{sampling}$) required for the application.

Three improvements of the PWM are also presented in order to enhance the sampling frequency ($f_{sampling}$) of ADCs. The double data rate (DDR) PWM focuses on increasing the PWM frequency (f_{PWM}) to enhance the $f_{sampling}$. The paralleled PWM also focuses on increasing the f_{PWM} by using a set of $K \frac{N}{K}$ -bit PWMs. The $f_{sampling}$ is significantly increased in the paralleled PWM, but it requires a set of K filters and the weighted addition of their filtered outputs, which increases the number of external components. The last presented improvement is the distributed duty cycle (DDC) PWM, which focus on decreasing the required cutoff frequency (f_C) of the filter by the distribution of the PWM assertions of the duty cycle to reduce the peak-to-peak filtered output.

The paralleled PWM greatly improves f_{Sampling} at the expense of increasing the complexity of the external circuits. On the contrary, the DDR and DDC PWM improvements do not require additional external circuits, but the f_{Sampling} gain is more limited. Depending on the number of ADCs and the application requirements, the three PWM improvements can be combined to increase the ADC performance. The methodology and PWM improvements were tested on a Zynq-7000 device, but they can be directly applied to different FPGA families.

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Abbreviations

The following abbreviations are used in this manuscript:

N	Bit resolution
ADC	Analog-to-digital converter
DAC	Digital-to-analog converter
LPF	Low pass filter
PWM	Pulse width modulation
f_c	Cutoff frequency
O_{LPF}	LPF order
SAR	successive approximation register
f_{Sampling}	ADC sampling frequency.
f_{CLK}	Clock frequency
LSB	Less significant bit
DDR	Double data rate
DDC	Distributed duty cycle

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