

## Design and Realization of Array Signal Processor VLSI Architecture for Phased Array System

D. Govind Rao<sup>1</sup>, T.Kishore Kumar<sup>2</sup>, N. S. Murthy<sup>2</sup>, A.Vengadarajan<sup>1</sup>

1. LRDE, DRDO, Ministry of Defence, Bangalore, India, email: dgrao@lrde.drdo.in,

2. National Institute of Technology, Warangal, India, email: nsm@nitw.ac.in

**ABSTRACT:** A method for implementing an array signal processor for phased array radars. The array signal processor can receive planar array antenna inputs and can process it. It is based on the application of Adaptive Digital beam formers using FPGAs. Adaptive filter algorithm used here is Inverse Q-R Decomposition based Recursive Least Squares (IQRD-RLS) [1] algorithm. Array signal processor based on FPGAs is suitable in the areas of Phased Array Radar receiver, where speed, accuracy and numerical stability are of utmost important. Using IQRD-RLS algorithm, optimal weights are calculated in much less time compared to conventional QRD-RLS algorithm. A customized multiple FPGA board comprising three Kintex-7 FPGAs is employed to implement array signal processor. The proposed architecture can form multiple beams from planar array antenna elements.

**Keywords:** Array Signal Processor, Adaptive Filter, Phased Array, Radar, Digital Beam Forming, IQRD-RLS, Kintex-7.

### I. INTRODUCTION

The problem of designing an array signal processor is an important part of general radar and communication systems. In the Radar operations, it is necessary to analyze the incoming data, process it, and then form the beams in order to track the targets. The proposed array signal processors can be used in phased array radar systems. The adaptive algorithm contained in the array signal processor gives the ability to the antenna array system to automatically sense the presence of the interference signals and to suppress precisely these interference signals while simultaneously enhancing the desired signal.

There are wide varieties of radar signal processor designed earlier. The single FPGA based processors are insufficient if there is more number of input signals to be processed simultaneously with the adaptive algorithms, since they lack resources. The advantage with the proposed array signal processor is that it is based on multiple FPGAs, in a customized FPGA board of three latest FPGAs, so that it can take inputs from sixteen channels simultaneously, and process it with an adaptive algorithm that acts fast. The number of inputs can be increased further. This type of distributed signal processing can take the advantage of parallel processing and hence high speed operation also.

Since varieties of algorithms available for adaptive filtering, the following aspects must be taken into account: Structure of the Filter, Convergence rate, Misadjustment, Tracking and Computational aspects. The most suitable for the phased array Application will be the Inverse Q-R Decomposition based Recursive Least Square (IQRD-RLS) adaptive filter. The main advantage by the recursive least-squares algorithm based on Inverse Q-R decomposition is its possible implementation in pipelined manner. The proposed array signal processor contains this IQRD-RLS algorithms are based on Givens rotations [2]. The IQRD-RLS algorithms will use almost 50% of the FPGA resources. Adaptive filter finds application in Adaptive Digital Beam Forming [3] with the following advantages: Power required for beam steering is less; variations associated with temperature, time, and other environmental changes can be reduced and it is possible to minimize the side-lobe levels, cancelling interference as well as multiple beam forming without altering the physical structure of the antenna [4].

The array signal processor proposed can take the inputs from the planar antenna elements and will form the beams adaptively, tracking the changes in the angle of arrival of the input signals. It takes inputs from sixteen antenna elements and form the beam, in a parallel manner such that one set of eight elements are processed in one FPGA[5],[6] and the other eight elements in a second FPGA, and ultimately we will be getting beams from 16 elements. The number of input channels can be further increased easily, to make single or

multiple beams, is the advantage of the array signal processor from conventional beam former architectures. Operation Modes of the digital beam former[7] can be controlled with the codes written on programmable devices, in this case Xilinx Kintex-7 [5],[6] FPGA, more specifically.

## II. DIGITAL DATA PROCESSOR

In the FPGA based array signal processor, all the operations are performed digitally [8]-[9]. The array signal processor is designed to work with IEEE-754 single precision floating point format for higher precision. The RF analog signal received at each antenna element is converted to digital form using high speed ADCs. The digital data samples are further processed with the array signal processor, calculating weights and making beams adaptively. A main lobe with nulls and side lobes is produced. The advantage with the proposed array signal processor in three FPGA architecture is that, we can make multiple beams in a much faster way than a single FPGA architecture. The restricting factors would be the sampling rate of ADCs, computational capacity and operating frequencies of the processor. Then, a high speed data communication path is needed for sending beams for further processing and plotting.

## III. ADAPTIVE FILTER ALGORITHMS AND OPTIMUM WEIGHT CALCULATION

For the adaptive Beam forming, adaptive weight calculation [10]-[11] is important. In radar applications, recursive least squares (RLS) and constrained recursive least squares (CRLS) algorithms is widely used compared to least mean squares (LMS) algorithm due to their fast convergence rate. Matrix inversion has to be performed for the RLS methods, which will be the most resource consuming and difficult in terms of hardware. So, QR-RLS is a better method to solve the problem. The QR decomposition is an elementary matrix operation, which will decompose a matrix into a triangular matrix and an orthogonal matrix. Matrix M is Q-R decomposed as  $M = Q \times R$ , where R is an upper triangular matrix and Q is an orthogonal matrix ( $Q^T \times Q = I$ ). QR-RLS algorithm used here is based on Givens Rotation [12] orthogonalization technique. The Givens Rotation based QR decomposition algorithm can be implemented in parallel and pipelined manner which is the best option for high-speed applications [13]. QR-RLS based weight calculation is done on Floating point arithmetic for infinite precision [14].

## IV. MATHEMATICAL MODEL OF IQRD-RLS

Algorithm based on the update of the inverse Cholesky factor [1]-[2], known as the Inverse QR [20] decomposition (IQRD-RLS) algorithm. Starting from the RLS solution  $w(k) = R^{-1}(k) p(k)$  where  $R(k) = X^T(k)X(k)$  is the  $(N+1) \times (N+1)$  input-data deterministic autocorrelation matrix,  $p(k) = X^T(k)d(k)$  is the  $(N+1) \times 1$  deterministic cross-correlation vector and  $w(k)$  is the Weight vector. We can show that

$$w(k) = w(k-1) + e(k)U^{-1}(k)U^{-T}(k)x(k) \quad (1)$$

where  $e(k) = d(k) - x^T(k)w(k-1)$  is the a priori error and the term multiplying this variable is known as the Kalman Gain. Since we know that  $Q\theta(k)$  is unitary, and we know that

$$g(k) = -\gamma(k)\lambda^{-1/2}U^{-T}(k-1)x(k) \quad \text{and} \quad a(k) = -\gamma^{-1}(k)g(k) = \lambda^{-1/2}U^{-T}(k-1)x(k) \quad (2)$$

Now, observe that  $\lambda^{-1/2}E(k)U^{-T}(k-1) = U^{-T}(k)$ , and suggests that  $U^{-T}(k-1)$  can be updated with the same matrix that updates  $U(k-1)$ .

For convenience,

$$u(k) = \lambda^{-1/2}U^{-1}(k-1)g(k) = -\lambda^{-1/2}\gamma(k)U^{-1}(k-1)a(k) = -\gamma^{-1}(k)U^{-1}(k)U^{-T}(k)x(k) \quad (3)$$

Finally,  $w(k)$  can be rewritten as  $w(k) = w(k-1) - e(k)\gamma(k)u(k)$  where the Kalman vector is now expressed as  $-\gamma(k)u(k)$ .

## V. HARDWARE IMPLEMENTATION

### Design of Array Signal Processor

For implementing the algorithm in real-time, it maps in a pipelined manner to the basic computation cells first. The cells runs in parallel, such that in each clock period all the cells are activated. The calculation can be decomposed in to two steps: calculating the sine and cosine of the rotation matrix and then applying the obtained rotation matrix to the input data[15]. Thus, the computational elements required for the array signal processor implementation of the IQRD-RLS[19] algorithm [16] introduced are the Angle cell, Balancing cell and the Weighing cell [14], and can be arranged as shown in Fig. 1.

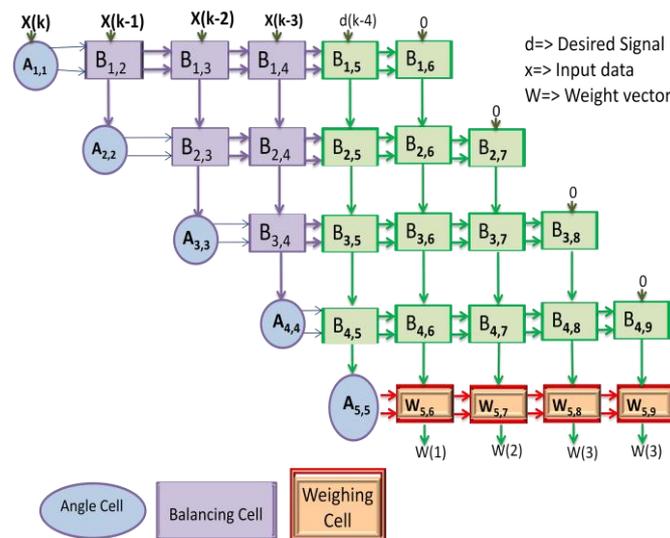


Fig. 1: Pipelined Architecture of IQRD-RLS dataflow for VLSI Implementation

**Angle cell:** Calculating the rotation angles based on the input signal.

**Balancing cell:** Performs the multiplication of the input signals, that is the output of the angle cells, with the array input signal.

**Weighing cell:** It calculating the weights from the outputs of the angle cells and the Balancing cells.

For calculating the weights, it gets the outputs from the balancing cells and performs multiplication with the Angle cell outputs of that row, then it weight the result with a weighing factor (K) and negates the results for getting the final Weight vector. The error signal [1]-[2] is calculated by the angle cell using the cosines and sines as below:

$$\epsilon(k) = \epsilon_{q1}(k) \prod_{i=0}^N \cos\theta_i(k) = \epsilon_{q1}(k)\gamma(k) \tag{4}$$

The balancing cell performs the balancing of the rest data by rotation between the data coming from angle cell with the internal element of the matrix U(l) and transfers the result to next cell. This processor also updates the elements of U(l) and transfers the cosine and sine values to the neighboring cell on the left. Assume that upper triangular matrix U(k) arranged below, the rows consisting of the new information data vector as in

$$Q(k)X(k) = \tilde{Q}(k) \begin{bmatrix} x(k) & x(k-1) & \dots & x(k-N) \\ 0 & 0 & \dots & 0 \\ \vdots & \vdots & \ddots & \vdots \\ 0 & \lambda^{\frac{1}{2}}U(k-1) & \dots & 0 \end{bmatrix} \tag{5}$$

and

$$Q_{\theta}(k) \begin{bmatrix} x^T(k) \\ \lambda^{\frac{1}{2}}U(k-1) \end{bmatrix} = Q'_{\theta N}(K)Q'_{\theta N-1}(K) \dots Q'_{\theta 1}(K) \begin{bmatrix} x'_1(k) \\ u'_1(k) \end{bmatrix} = \begin{bmatrix} 0 \\ U(K) \end{bmatrix} \tag{6}$$

Following the same pattern, we can arrange the basic cells in order to compute the rotations of the IQRD-RLS algorithm as shown in “Fig3”, with the input signal x(k) entering the array serially.

$$d(k) = \begin{bmatrix} \epsilon_{q1}(k) \\ d_{q2}(k) \end{bmatrix} = Q_{\theta}(k) \begin{bmatrix} d(k) \\ \lambda^{\frac{1}{2}}d_{q2}(k-1) \end{bmatrix} \tag{7}$$

which are also initialized with zero and updated in each clock cycle. The column on the left-hand side of the array performs the rotation and stores the rotated values of the desired signal vector which are essential to compute the error signal.

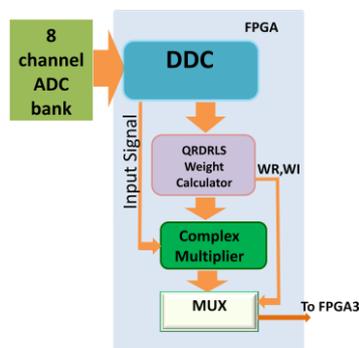
The last row in the pipelined architecture contains the weighing cells are intended for calculating the final weights from the Angle Cell and Balancing Cell outputs[16].

In order to allow the pipelining, the outputs of each cell are computed at the present clock period and made available to the neighboring cells in the following clock period. Each row of cells in the array implements

a basic Givens rotation between one row of  $\lambda U(k-1)$  and a vector related to the new incoming data  $x(k)$ . The top row of the systolic array performs the zeroing of the last element of the most recent incoming  $x(k)$  vector. The result of the rotation is then passed to the second row of the array. This second row performs the zeroing of the second-to-last element in the rotated input signal. The zeroing processing continues in the following rows by eliminating the remaining elements of the intermediate vectors  $x^i(k)$

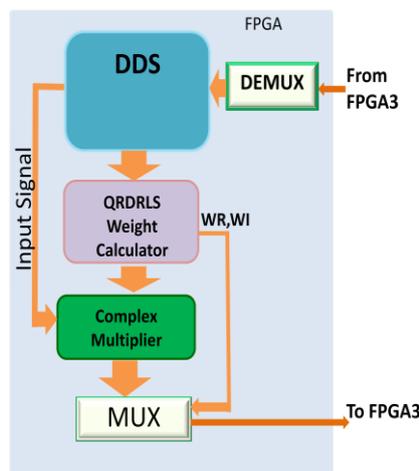
**Implementation of Array Signal Processor**

The pipelined architecture of the IQRD-RLS algorithm shown in Fig. 3 is contained in FPGA1 and FPGA2 in the Array signal processor architecture. The Function of the algorithm is to adaptively calculate the weight according to the input signal. Then these adaptively calculated weights [18] will be multiplied with the input in phase and quadrature signals with a complex multiplier. Then the complex multiplier outputs will be summed together to get the final beam. The architecture of the Array Signal Processor is divided into three different FPGA's for the purpose of the modular design and parallel processing approach. First and second FPGAs are doing the same work, taking input from the ADC's, frequency translation of signal using DDCs, weight calculating adaptively, followed by multiplying the input signal with the weights. These multiplier outputs are fed to the third FPGA which performs the summation of these signals from both FPGA1 and FPGA2, and produce the beam.



**Fig. 2:** VLSI Architecture of FPGA1 and FPGA2 with ADC and DDC

The Block diagram shown in Fig. 2 and Fig. 3 are the architectures of FPGA1 and FPGA2, and that in Fig. 6 is of FPGA3. The Fig. 2 is the architecture with ADC, DDC, IQRD-RLS weight calculator etc, and Fig. 5 is the architecture with a DDS inside the FPGAs. In the former architecture of the array signal processor, it take input from the ADC, and then DDC will make two signals (I & Q). In the latter, a DDS is designed inside the FPGAs so that the DDC module can be avoided, in turn, resulting the reception of signals from the ADC modules are not needed, which is best method for the testing environment. We can change the Frequency, Phase, and Amplitude of the DDS output signals through commands.



**Fig. 3:** VLSI Architecture of FPGA1 and FPGA2 without ADC and DDC

The array signal processor will communicate with the external world via the RS232 interface, which is in the FPGA3. The user commands to the signal processor as well as calculated weights and beam information

from the signal processor is taken through this serial interface. FPGA3 will do the final summing of the complex multiplier outputs from FPGA1 and FPGA2 and form the Array Factor and beam, as shown in Fig. 6.

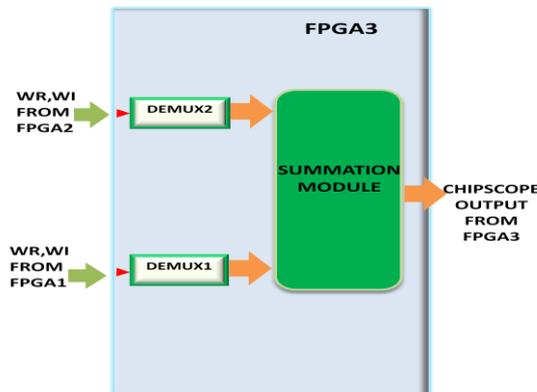


Fig. 4: VLSI Architecture of FPGA3

This implemented architecture is for sixteen element array, but can be extended to any number of antenna arrays. The complete test setup of the array signal processor using the hardware emulator is shown in Fig. 7. The DDC module block diagram is shown in Fig. 6.

The digital data is sampled at 50 MHz clock frequency and then passed through a digital Mixer consisting of a 50 MHz Numerically Controlled Oscillator (NCO), a multiplier (14x14 bit), suitable low pass decimation and compensating filters (CIC and CFIR filters) of 5 MHz bandwidth for filtering the unwanted signals which are outside the band and a decimator of rate 10 for bringing down the sampling rate to 5 MS/s. From the DDC we get two signals which are In phase (I) and Quadrature (Q) to the input signal. By multiplying the input data, with the locally generated quadrature cosine and sinusoidal signals, we are getting two frequency translated signals as mentioned earlier in the base band region. The developed array signal processor can accept inputs from sixteen antenna elements simultaneously, and can form the beam. The three FPGA architecture helps the array signal processor to increase the number of inputs as well as parallel processing. For making multiple beams, the array signal processor can be preferred over the conventional single FPGA architectures since it can calculate multiple beams faster because of parallel processing. The hardware emulator block diagram is shown in Fig. 7.

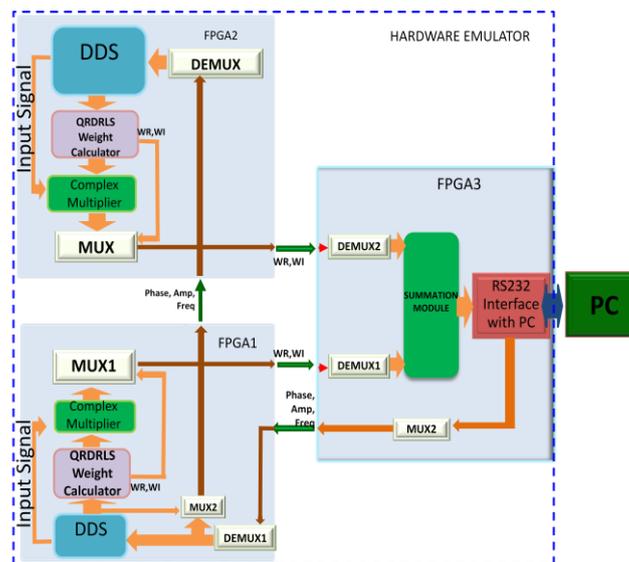
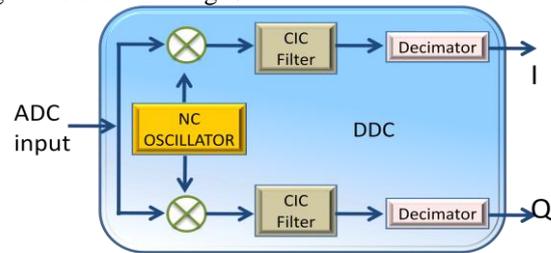


Fig. 5: Complete test setup including Hardware Emulator

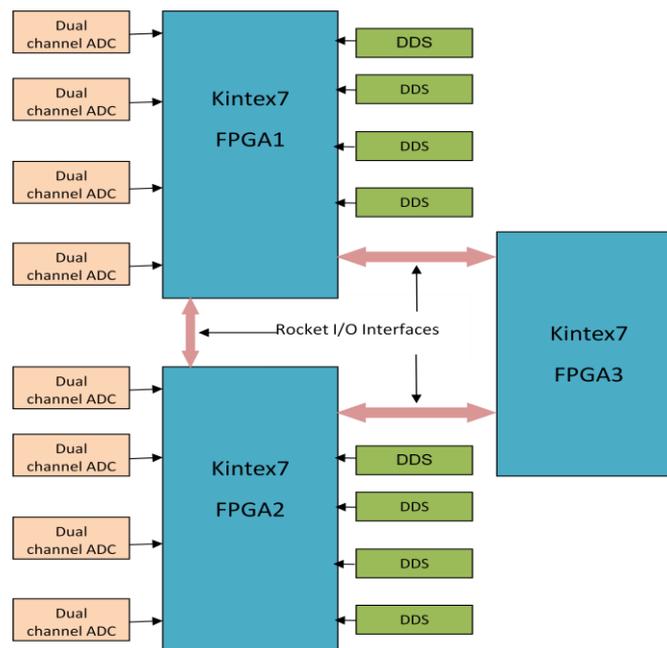
The DDC module block diagram is shown in Fig. 6.



**Fig. 6:** Digital Down Converter (DDC) Block Diagram

The Digital data is sampled at 50 MHz clock frequency and then passed through a digital Mixer consisting of a 50 MHz Numerically Controlled Oscillator (NCO), a multiplier (14x14 bit), suitable low pass decimation and compensating filters (CIC and CFIR filters) of 5 MHz bandwidth for filtering the unwanted signals which are outside the band and a decimator of rate 10 for bringing down the sampling rate to 5 MS/s. From the DDC we get two signals which are In phase (I) and Quadrature (Q) to the input signal. By multiplying the input data, with the locally generated quadrature cosine and sinusoidal signals, we are getting two frequency translated signals as mentioned earlier in the base band region.

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**Fig. 7:** Architecture of Prototype Hardware

The important features of the hardware emulator are below:

- FPGA- Kintex-7 XC7K410T- 1FBG900 [2]
- Clock: Onboard Oscillators: 100 MHz and 200 MHz
- Memory: One Flash 64M x 16 Memory – S29GL512S from SPANSION
- Optical interface @ 6.5 Gb/s
- Dual channel 14 bit ADC, ADS4249.

The prototype of the hardware is in Fig. 8.

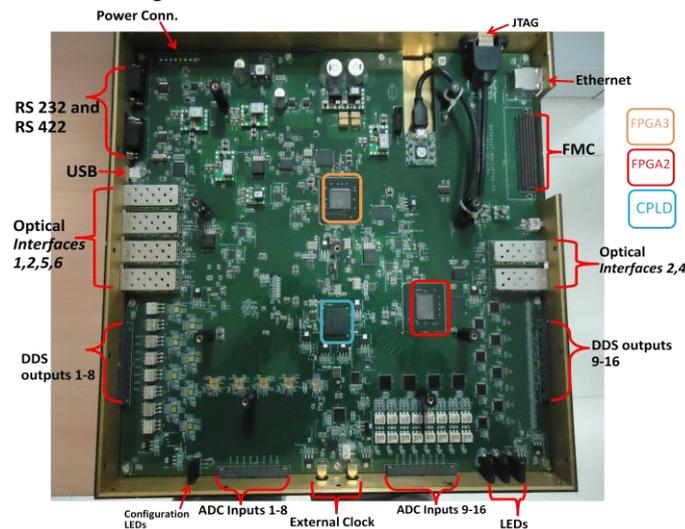


Fig. 8: Prototype Hardware (Kintex-7)

The tools used for the implementation of Array Signal Processor are as given:

- Matlab 2010a for algorithm modeling and simulation purpose.
- Xilinx ISE 13.4 for functional simulation and implementation.

**VI. SIMULATION AND IMPLEMENTATION RESULTS OF ADAPTIVE BEAM FORMER**

The FPGA based Array signal processor is designed on VHDL. The VLSI architecture modeling has been carried out and simulated also. The functional verification has been done to validate the correctness of the results. Initially the 8 element linear array and 16 element linear array configuration is simulated. Then a planar array of 16 element configuration has been implemented. The implementation of ADBF for 1 beam is done on Kintex-7 series FPGA and later 2 beams also. Due to the requirement of more resources on hardware, for multiple beams, some reuses of the modules are necessary and hence time required for calculating the multiple weights will be large. The resource utilization for this implementation is given in Table 1 and 2.

Logic Utilization	Used	Available	Utilization
Number of Slice Registers	119840	508400	23%
Number of Slice LUTs	144126	254200	56%
Number of fully used LUT-FF pairs	82594	181372	45%
Number of bonded IOBs	73	500	14%
Number of Block RAM/FIFO	8	795	1%
Number of BUFG/BUFGCTRLs	5	32	15%
Number of DSP48E1s	643	1540	41%

TABLE 1: Resource utilization of FPGA-I and FPGA II for implementation of Inverse QRD RLS for Adaptive beam former of 8 elements

Logic Utilization	Used	Available	Utilization
Number of Slice Registers	119840	508400	23%
Number of Slice LUTs	144126	254200	56%
Number of fully used LUT-FF pairs	82594	181372	45%
Number of bonded IOBs	73	500	14%
Number of Block RAM/FIFO	8	795	1%
Number of BUFG/BUFGCTRLs	5	32	15%
Number of DSP48E1s	643	1540	41%

TABLE 2: Resource utilization of FPGA-III for implementation of Adaptive beam former for sixteen elements

- Result of the implementation of adaptive beam forming using Inverse QRD-RLS algorithm for 16 element planar array is shown in Fig. 9. Input azimuth angle of arrival for desired signal =10 deg, elevation angle of arrival for desired signal =35 deg, and produces a single beam.

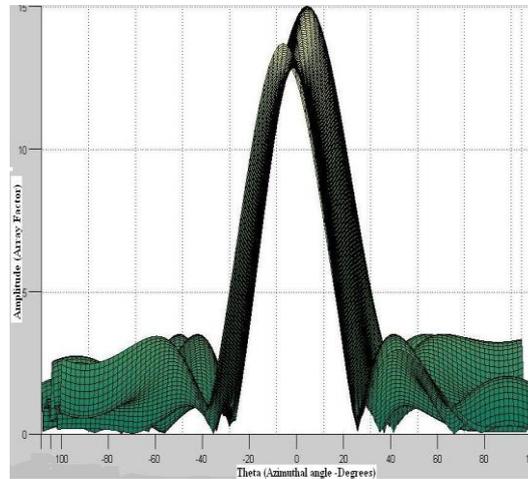


Fig. 9: Planar array of sixteen elements single beam

- Result of the implementation of adaptive beam forming using Inverse QRD-RLS algorithm for 16 element planar array is shown in Fig. 10. Input azimuth angle of arrival for desired signal = 15 deg, Input elevation angle of arrival for desired signal =45 deg, and 2 beams are produced.

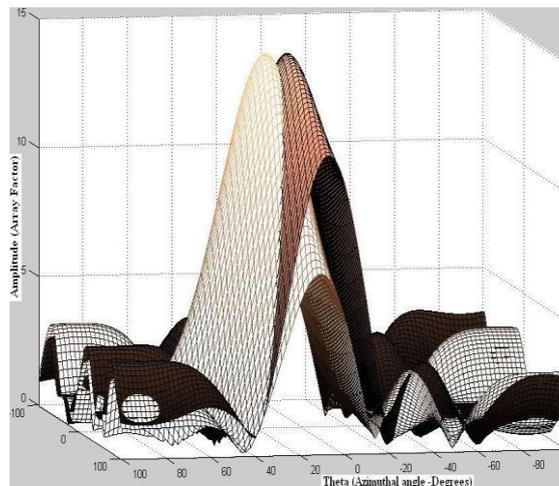


Fig. 10: Planar array of sixteen elements two adaptive beams

- Chipscope result image of weights calculated by FPGA1 is shown in Fig. 11

Waveform - DEV0 MyDevice0 (XC7K410T) UNIT:0 MyILA0 (ILA)		523	524	525	526	527	528	529	530
Bus:Signal	X 0								
..._op_ar1	0 0								
/w1	CF23 CF23				CF23D70A				
/w2	DC06 DC06				DC061E95				
/w3	531C 531C				531C7B71				
/w4	4490 4490				4490E6B7				
/w1	5B21 5B21				5B217453				
/w2	DC92 DC92				DC929C2D				
/w3	5388 5388				5388A382				
/w4	C42E C42E				C42EC5D7				

Fig. 11: FPGA1 ChipScope Result of weights calculated

## VII. CONCLUSION

We have implemented an FPGA based array signal processor for phased array radars, capable of processing 16-element planar inputs. The weights used for Adaptive beamforming are calculated adaptively using the IQRD-RLS algorithm. Xilinx's Kintex-7 FPGAs are used to implement the array signal processor. It is done on a customized multi FPGA board containing 3 Kintex-7 FPGAs and the design used here is modular. The weights are calculated and beam is formed from the 16 element array inputs. Beams can be plotted in MATLAB with exporting calculated data to the PC.

The array signal processor can be easily modified to increase the number of inputs further. The array signal processor developed can run at a frequency of 50 MHz. Depending on the availability of the resources and the input sample rate, we can increase speed of operation as well as can be extended for planar arrays of larger array dimensions. FPGA based signal processing finds huge applications in modern radars making the system immune to the limitations that the analog methods face. The array signal processor enables the radars to track the changes in continuously varying environment and makes the ADBF system robust and efficient

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