

Transients in Reconfigurable DSP Systems

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Abstract – Nowadays in solving measurement and control problems the processing of input data is performed typically by model-based digital signal processing (DSP) systems which contain a representation of our knowledge about the nature and the actual circumstances of the problem in hand. If the nature and/or the actual circumstances change the corresponding model should also be changed. Similarly, if due to measurements the amount of knowledge about the problem increases, an improved model can be suggested which provides better performance. As a consequence the real-time adaptation or reconfiguration of the DSP system to be applied can hardly be avoided. In this paper the transients caused by these adaptations/reconfigurations is investigated. It is shown that in the case of feedback systems, i.e. for infinite impulse response (IIR) filters these transients are strongly structure-dependent and that the so-called orthogonal filter structures provide good performance also in this respect.

Keywords – Reconfigurable Systems, Transients, Structure Dependency, IIR Filters, Adaptive Filters.

I. INTRODUCTION

The study of reconfigurable DSP systems is a very important area of research related mainly to larger scale, distributed intelligence monitoring and control systems. To use reconfiguration techniques in monitoring and control system has real meaning if drastic changes may occur in the physical system. Changes due to faults evolving into system degradation are typical examples. In such cases our supervisory computer program should observe the changes and turn to another operation mode or program. With other words the models applied within the computer program are also to be changed to correctly represent the physical system. Model changes can be performed using different techniques [1]. For conventional system models the typical solution is the adaptation or direct change of the coefficients and/or the (signal processing) structure. These changes, however, can cause large transients, since there is a real difference between the stationary behavior of the system before and after the change.

In this paper the reconfiguration transients of DSP systems are investigated. For simplicity we will consider systems with changing coefficients but fixed structure. It is

shown that reconfiguration transients depend significantly on the DSP structure applied. This structure dependency is strongly related to the energy distribution within the processing structure, therefore the famous orthogonal structures provide good performance [2], [3]. The behavior of the widely used direct structure (see e.g. [4]) is rather poor. The reconfiguration can be performed also in multiple steps. In parameter adaptive systems this strategy corresponds to gradually changing the coefficients. However, it is not trivial yet how to determine the best sequence of the coefficient modification steps.

The results of the above considerations can contribute also to the design problem of narrow-band measuring channels, especially to that of the null indicators. It is well known that narrow-band filters have long transients which increase the overall response time of our measuring system. To reduce this effect at the beginning of the measurement we can start with a wider bandwidth and switch to the narrow-band somewhat later as the energy storage devices of the filters are almost settled. If we apply reconfigurable filter structures having good transient properties this strategy may result in remarkable improvements. In this paper the one-step reconfiguration issues will be illustrated by this idea, i.e. by the step responses of different lowpass filters with one-step bandwidth reduction.

II. STRUCTURE DEPENDENCY OF RECONFIGURATION TRANSIENTS

The step response of time-invariant linear systems is well understood and seems to be an appropriate tool to characterize the transient properties of the system. It is also well known that if the internal energy of the system is originally zero, then the step response can be calculated from the corresponding differential or difference equation. If we change the coefficients of an already operating discrete IIR filter then the output can be considered as the sum of different components. One is the response to the input, while the others are the responses due to the initial conditions, i.e. due to the stored values, which equal state variable values if the state variable formulation is

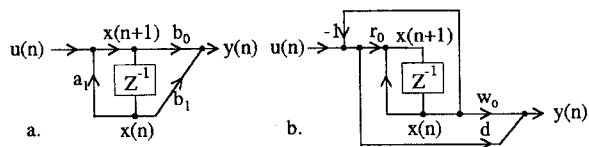


Fig. 1. a: first-order direct structure, b: first-order resonator-based structure

applied. These initial conditions generate an impulse response from the output of the storage device to the output of the filter.

Obviously in the majority of the practical cases these state variable values are not known, however, their possible range can be estimated. It is well known from the literature of digital signal processing (see e.g. [4]) that the different structures have quite different internal dynamic range. As an example consider the first-order direct structure on Figure 1.a. The state variable description has the form of

$$\begin{aligned} x(n+1) &= a_1 x(n) + u(n) \\ y(n) &= b_0 x(n+1) + b_1 x(n) \end{aligned} \quad (1)$$

If $u(n) = 1$ for $\forall n$, then $x(n) \rightarrow 1/(1-b_1)$ as $n \rightarrow \infty$ and can be a large value if a_1 (< 1) is close to 1. Note that the dynamic range of this filter is parameter dependent. As a counterexample consider the structure on Figure 1.b. Its state variable description is

$$\begin{aligned} x(n+1) &= x(n) + r_0[u(n) - x(n)] \\ y(n) &= w_0 x(n) + d[u(n) - x(n)] \end{aligned} \quad (2)$$

If $u(n) = 1$ for $\forall n$, then $x(n) \rightarrow 1$ as $n \rightarrow \infty$, i.e. it is independent of the coefficients. These two examples show that the nature and the value of the state variables can differ considerably. Obviously the coefficients of the two structures are different and the reconfiguration will change them differently, therefore a complete characterization without knowing the input samples is not possible. Under mild restrictions, however, a very interesting link can be established to structures having minimum round-off noise [5], [6]. These structures can be characterized by relatively uniform energy distribution among the state variables. As a consequence the output sensitivity to the rounding errors is relatively low, which results in lower (or minimum) output noise level. In this model the samples of the noise sources for modeling roundoff errors are directly added to the actual state variable values. Concerning transients we have a similar situation, since the initial conditions behave like additive impulses to the actual state variables. In both cases the additionally introduced energy must be transferred from the state variables

to the output, but in the second the time-domain behavior is more emphasized. The relatively uniform energy distribution requirement is met by the so-called orthogonal structures (see [3], [4]), which have very good internal dynamic range, low roundoff noise, and can reduce zero-input limit cycles. From the above reasoning it turns out that they are good candidates also for implementing reconfigurable IIR filters.

It is important to note that in the theory and practice of adaptive filters this structure dependence is not recognized properly. In the majority of the adaptation schemes the correction terms are based on direct measurements of the output signal. Correction of the parameters can be considered as reconfiguration producing transients at the output. Since these transients may disturb the overall performance of the adaptive filter, it is reasonable to apply DSP structures with low reconfiguration transients. Another consequence can be the reconsideration whether adaptation at lower rate results in better performance.

As simple illustrations the step responses of the direct and the resonator-based structures are presented on Figures 2-7 for different filter order. The idea behind is the assumption that the steady state behavior can be reached faster if as the first step a filter with wider bandwidth is operated, and this is followed by a one-step bandwidth reduction. The coefficients for the first-order filters are given in Table I.

| direct | | resonator-based | |
|-------------|---------|-----------------|--------|
| $b_{0,old}$ | 0.1367 | $r_{0,old}$ | 0.2735 |
| $b_{1,old}$ | 0.1367 | $w_{0,old}$ | 1.0000 |
| $a_{1,old}$ | -0.7265 | d_{old} | 0.1367 |
| $b_{0,new}$ | 0.0155 | $r_{0,new}$ | 0.0309 |
| $b_{1,new}$ | 0.0155 | $w_{0,new}$ | 1.0000 |
| $a_{1,new}$ | -0.9691 | d_{new} | 0.0155 |

TABLE I
COEFFICIENTS FOR THE FIRST-ORDER FILTERS

III. MULTIPLE STEP RECONFIGURATION

The multiple step strategy and the structure dependency are illustrated by a simple pole migration example. The reason of this illustration is the observation that gradually changing the coefficients in certain filter structures may move the poles temporarily out of the unit circle. Even if for time-invariant systems it is hard to give the proper interpretation of the poles, unstable temporary pole positions can indicate the danger of larger transients. An 8th order Butterworth lowpass filter was reconfigured from cut-off frequency of $0.1 f_s$ to $0.01 f_s$. On Figures 8-11 the

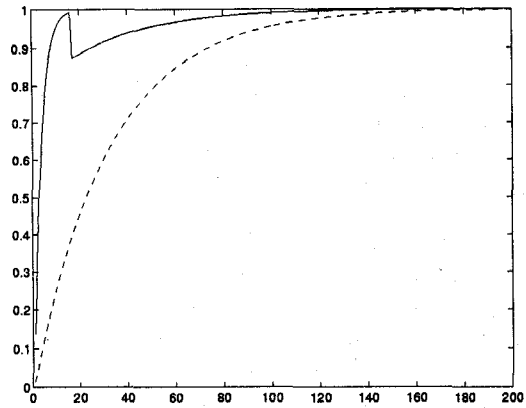


Fig. 2. solid line: Step response of a first-order direct structure reconfigured at step 16. dashed line: Step response of the narrow-band first-order direct structure.

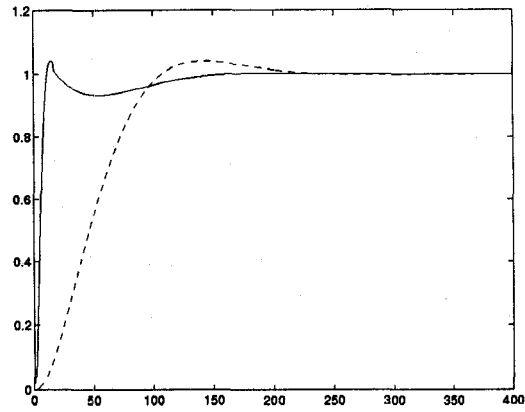


Fig. 5. solid line: Step response of a second-order resonator-based structure reconfigured at step 16. dashed line: Step response of the narrow-band second-order resonator-based structure.

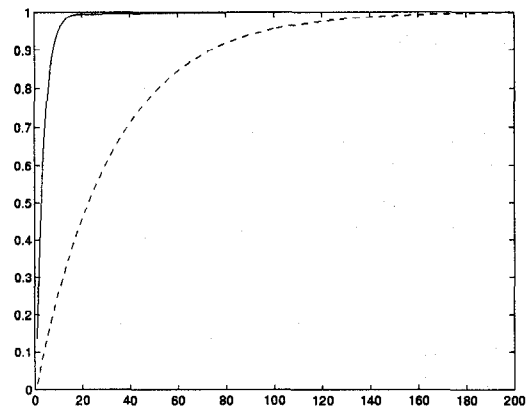


Fig. 3. solid line: Step response of a first-order resonator-based structure reconfigured at step 16. dashed line: Step response of the narrow-band first-order resonator-based structure.

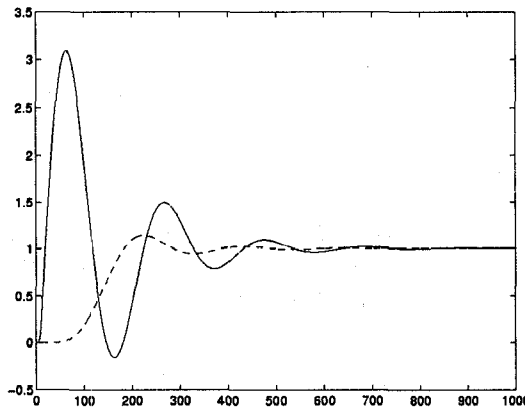


Fig. 6. solid line: Step response of a sixth-order direct structure reconfigured at step 16. dashed line: Step response of the narrow-band sixth-order direct structure.

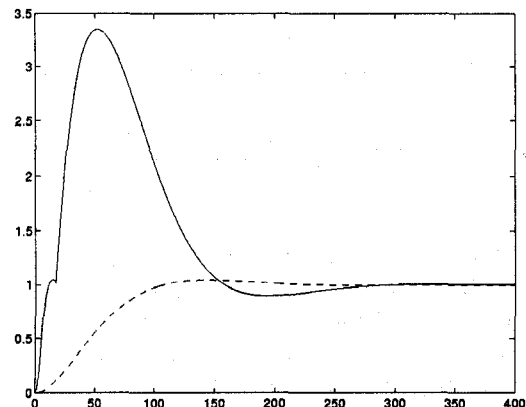


Fig. 4. solid line: Step response of a second-order direct structure reconfigured at step 16. dashed line: Step response of the narrow-band second-order direct structure.

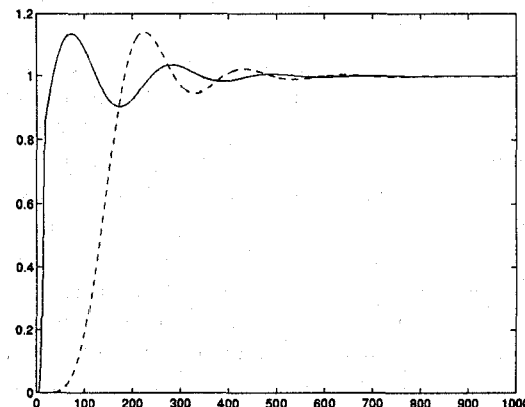


Fig. 7. solid line: Step response of a sixth-order resonator-based structure reconfigured at step 16. dashed line: Step response of the narrow-band sixth-order resonator-based structure.

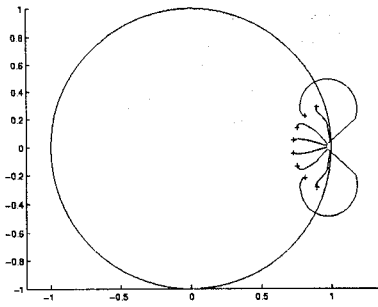


Fig. 8. Pole migration of an eighth-order direct filter

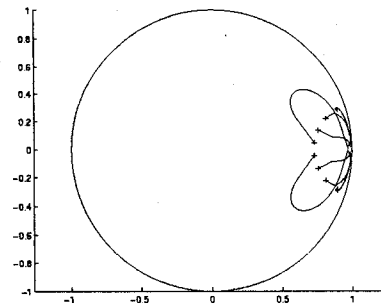


Fig. 10. Pole migration of an eighth-order normalized Lattice filter

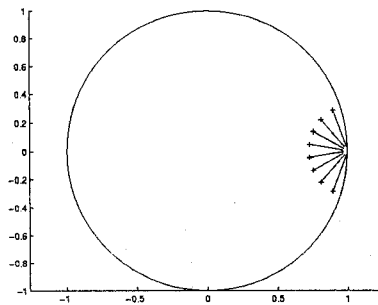


Fig. 9. Pole migration of an eighth-order parallel filter

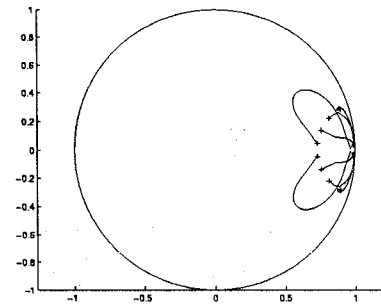


Fig. 11. Pole migration of an eighth-order resonator-based filter

pole-migration of four different structures (the direct [4], the normalized lattice [2], the resonator-based [3] and the parallel [4]) are recorded. The filter coefficients were linearly interpolated in 100 steps. The direct structure temporarily has lost stability because his poles have migrated out of the unit circle. The parallel structure seems to provide rather good behavior.

A better behavior can be achieved if instead of linearly interpolating the coefficients, the filter design is performed in every step, and in the multiple step reconfiguration the corresponding coefficients are applied. In this case the "interpolation" is on the level of the filter specification, i.e. we gradually change the bandwidth and/or the center frequency. The optimal strategy of this "interpolation" is still an open question, however, the results related to measurements with sweep generators may help.

IV. CONCLUDING REMARKS

In this paper we have reported our investigations related to reconfiguration transients. This topic has real importance if our models representing our knowledge about the reality are to be changed during operation. As it is illustrated, these transients are structure-dependent, therefore in the design of adaptive/reconfigurable systems structures with better transient behavior should be preferred. The large transients of drastic changes can be

reduced by multiple step adaptation/reconfiguration, as well. The optimal strategy is still an open question. In our investigations up till now we have considered only parameter changes. The next step is the evaluation and design of reconfiguration if the order and/or the structure of the filter is also to be changed.

V. ACKNOWLEDGMENT

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